REPORT DOCUMENTATION PAGE

1. AGENCY USE ONLY (Leave blank)  2. REPORT DATE  3. REPORT TYPE AND DATES COVERED

4. TITLE AND SUBTITLE
Active Control of Radiated Sound with Integrated Piezoelectric Composite Structures

5. FUNDING NUMBERS

6. AUTHOR(S)
C. R. Fuller and L. E. Cross

7. PERFORMING ORGANIZATION NAME(S) AND ADDRESS(ES)
Vibration and Acoustics Laboratories, Department of Mechanical Engineering
VPI & SU, Blacksburg, VA 24061

8. PERFORMING ORGANIZATION REPORT NUMBER

9. SPONSORING/MONITORING AGENCY NAME(S) AND ADDRESS(ES)
Dr. Kam Ng
Office of Naval Research
Code 334
800 North Quincy Street
Arlington, VA 22217

10. SPONSORING/MONITORING AGENCY REPORT NUMBER

11. SUPPLEMENTARY NOTES

12a. DISTRIBUTION/AVAILABILITY STATEMENT  12b. DISTRIBUTION CODE

13. ABSTRACT (Maximum 200 words)
See Following Pages

14. SUBJECT TERMS

15. NUMBER OF PAGES

16. PRICE CODE

17. SECURITY CLASSIFICATION OF REPORT  18. SECURITY CLASSIFICATION OF THIS PAGE  19. SECURITY CLASSIFICATION OF ABSTRACT  20. LIMITATION OF ABSTRACT
**GENERAL INSTRUCTIONS FOR COMPLETING SF 298**

The Report Documentation Page (RDP) is used in announcing and cataloging reports. It is important that this information be consistent with the rest of the report, particularly the cover and title page. Instructions for filling in each block of the form follow. It is important to **stay within the lines** to meet optical scanning requirements.

| Block 1. **Agency Use Only** (Leave blank). |
| Block 2. **Report Date**. Full publication date including day, month, and year, if available (e.g. 1 Jan 88). Must cite at least the year. |
| Block 3. **Type of Report and Dates Covered**. State whether report is interim, final, etc. If applicable, enter inclusive report dates (e.g. 10 Jun 87 - 30 Jun 88). |
| Block 4. **Title and Subtitle**. A title is taken from the part of the report that provides the most meaningful and complete information. When a report is prepared in more than one volume, repeat the primary title, add volume number, and include subtitle for the specific volume. On classified documents enter the title classification in parentheses. |
| Block 5. **Funding Numbers**. To include contract and grant numbers; may include program element number(s), project number(s), task number(s), and work unit number(s). Use the following labels: |
| C - Contract | PR - Project |
| G - Grant | TA - Task |
| PE - Program | WU - Work Unit |
| Element Accession No. |
| Block 6. **Author(s)**. Name(s) of person(s) responsible for writing the report, performing the research, or credited with the content of the report. If editor or compiler, this should follow the name(s). |
| Block 7. **Performing Organization Name(s) and Address(es)**. Self-explanatory. |
| Block 8. **Performing Organization Report Number**. Enter the unique alphanumeric report number(s) assigned by the organization performing the report. |
| Block 9. **Sponsoring/Monitoring Agency Name(s) and Address(es)**. Self-explanatory. |
| Block 10. **Sponsoring/Monitoring Agency Report Number**. (If known) |
| Block 11. **Supplementary Notes**. Enter information not included elsewhere such as: Prepared in cooperation with...; Trans. of...; To be published in... When a report is revised, include a statement whether the new report supersedes or supplements the older report. |
| Block 12a. **Distribution/Availability Statement**. Denotes public availability or limitations. Cite any availability to the public. Enter additional limitations or special markings in all capitals (e.g. NOFORN, REL, ITAR). |
| DOD - See DoDD 5230.24, "Distribution Statements on Technical Documents." |
| DOE - See authorities. |
| NTIS - Leave blank. |
| Block 12b. **Distribution Code**. |
| DOD - Leave blank. |
| DOE - Enter DOE distribution categories from the Standard Distribution for Unclassified Scientific and Technical Reports. |
| NASA - Leave blank. |
| NTIS - Leave blank. |
| Block 13. **Abstract**. Include a brief (Maximum 200 words) factual summary of the most significant information contained in the report. |
| Block 14. **Subject Terms**. Keywords or phrases identifying major subjects in the report. |
| Block 15. **Number of Pages**. Enter the total number of pages. |
| Block 16. **Price Code**. Enter appropriate price code (NTIS only). |
| Block 20. **Limitation of Abstract**. This block must be completed to assign a limitation to the abstract. Enter either UL (unlimited) or SAR (same as report). An entry in this block is necessary if the abstract is to be limited. If blank, the abstract is assumed to be unlimited.
ABSTRACT

The work investigated in this URI project is concerned with the active control of radiated sound using advanced structural systems with fully integrated actuators and sensors driven by realistic forms of disturbances. The overall goal was to demonstrate active control of sound radiated from vibrating structures with a fully integrated, practical active material including multiple actuators and sensors. This ultimately implies addressing the requirements for realistic active structures with integrated actuators and sensors as well as developing new sensing, control theory and design approaches so that the active material systems can be correctly and efficiently implemented.

This report summarizes three years of research work to achieve these goals. The work was evenly split between two main groups. The Materials Research Laboratory (MRL) at Penn State University addressed the development and construction of a suitable actuator system. The Vibration and Acoustics Laboratories (VAL) at Virginia Tech focussed upon developing new approaches for radiation control, system component development and integration and integrated system testing and demonstration. The work essentially followed a general theme of continual component development, system integration and testing through various phases tightly coordinated between MRL and VAL.

The two core technologies of the project were a new air piezoceramic actuator system conceived and developed by MRL and a new radiation control approach based upon a continuous active skin conceived and developed by VAL.

The report describes work by MRL-PSU on developing an constructing a new actuator called PANEL based upon using double amplification obtained from a system of orthogonal bimorph piezoelectric elements covered with an acoustic diaphragm configured in a flextransional type manner. The resultant PANEL source after many iterations of analysis, development, construction and testing was found to provide amplification ratios of around 250:1 and generate diaphragm vibration levels of the order of 500 microns (on resonance) and 200 microns (off resonance) over a frequency range of 0 to 1500Hz. The corresponding sound pressure levels generated by the PANEL source at 1m ranged from 80dB at 200Hz to 90dB above 400Hz. These performance levels were considered high enough to enable the PANEL source to be applied to a number of practical noise problems such as interior noise in aircraft and cars as well as electrical transformer noise and jet engine inlet noise. For applications below 200Hz, where the performance of the PANEL falls off, a new pseudo-shear multi-layer actuator utilizing folded multi-layer piezoelectric elements was developed and tested. The new pseudo-shear actuator was found to have significantly enhanced very low frequency performance below 200Hz.

The report also describes the new active noise control approach based upon implementing an active-skin which completely covers the structure conceived by VAL-
VPI. In the VAL part of the project, multiple PANEL actuators were integrated into a continuous skin system with independently controllable sections. A new structural acoustic sensing approach which enable the integration of sensors directly into the skin and yet allowed estimation of far-field sound radiation was developed and implemented in the active skin system. New control approaches and system optimization and design approaches were developed and used to efficiently configure and control the skin system. New active-passive approaches, which take advantage of the system natural dynamics to lower control authority requirements and increase robustness, were investigated. Finally the component technologies were integrated into an active skin approach designed to control broadband sound radiation from a test panel. The active skin system with integrated sensors was tested and found to provide total attenuation of the plate radiated sound power of 7.3dB over a bandwidth from 200 to 600Hz. This bandwidth encompasses multiple plate mode resonances. In order to handle the very low frequency region below 200Hz a new distributed active vibration absorber was implemented and tested. The work has demonstrated the high potential of an active skin with integrated piezoelectric amplifier elements and structural sensors for controlling structurally radiated sound with realistic loads.

The report is divided into two main parts. The first part summarizes work at MRL-PSU on developing the new actuator systems. The second part describes the system integration and testing performed at VAL-VPI. Throughout the report, reference is made to a set of published papers which describe the project work in detail. These papers are provided for convenience of the reader in Appendices at the end of the report.
APPENDICES


35. “Recent Advances in the Active Control of Structurally Radiated Sound,” C. R. Fuller. Proc. 5th International Congress on Sound and Vibration, University of Adelaide, Australia (December 1997).


APPENDIX 1
Piezoelectric air transducer for active noise control

Baoxin Xu, Qiming Zhang, V. D. Kugel and L. E. Cross

Intercollege Materials Research Laboratory, Pennsylvania State University, University Park, PA 16802

ABSTRACT

A new type of piezoelectric air transducer has been developed for active noise control and other air acoustics applications. The transducer is based on the composite panel structure of a bimorph-based double amplifier, that is, two parallel bimorphs or bimorph arrays with a curved cover plate as an active face attached to the top of the bimorphs. The electro-mechanical and electro-acoustic properties of the double amplifier structure and the transducer are investigated in this paper. The displacement of the cover plate of the double amplifier structure can reach millimeter scale with a relatively low driving voltage, which is more than ten times larger than the tip displacement of bimorphs. The sound pressure level (SPL) of the transducer can be larger than 90dB (near field) in the frequency range from 50 to 1000Hz and be larger than 80dB (far field) from 200Hz to 1000Hz, with the largest value more than 130dB (near field). Because of its light weight and panel structure, it has the potential to be used in active noise control.

Keywords: piezoelectric transducer, bimorphs, double amplifier structure, air acoustics, active noise control

1. INTRODUCTION

The electro-mechanical transducers based on piezoelectric materials have been widely used for many decades. However, most of these transducers are used in the areas of ultrasonics and underwater acoustics. It is still difficult to apply piezoelectric materials or devices in air acoustics, especially at low frequency range. On the other hand, because of their unique properties, such as high electro-mechanical coupling efficiency, the ability of both sensing and transmitting, low loss, light weight and panel structure, it is highly desirable to use piezoelectric transducers at low frequency range and active noise control.

There are two main problems to be solved to apply piezoelectric transducers in air acoustics and active noise control. One is that the displacement must be much larger than that of the common ceramic actuators because the radiated acoustic energy is proportional to the square of the displacement amplitude. Another one is that the acoustic impedance should be matching to air to ensure an effective acoustic flow from the transducer to the medium.

Among all the piezoelectric actuators, bimorph type actuators generate the largest displacement (tip displacement). However, the displacement is still not large enough to be used in air acoustics at low frequency range. In this paper, a new type of piezoelectric transducer based on the idea of bimorph-based double amplifier structure is presented. The displacement generated by the double amplifier structure can be more than ten times larger than the tip displacement of bimorphs. The acoustic impedance matching between the transducer and air can be obtained by carefully choosing the cover plate (active face) materials and its geometric configurations. As a result of these new designs, the transducer is promising to be used in active noise control as a sound transmitter.

2. DESIGN PRINCIPLES

The basic configuration of the bimorph-based double amplifier structure is shown in Fig. 1. The structure mainly consists of two parallel bimorphs with a curved (triangle shape) cover plate as an active face attached to the top of the bimorphs. Higher displacement is achieved by changing the tip displacement of bimorphs to a flexural motion of the cover plate. That is, when the bimorphs vibrate horizontally, the cover plate vibrates vertically with larger amplitude. Fig. 1(b) is the schematic of the displacement of the cover plate and bimorph actuators. From the geometric consideration, the vertical displacement of the middle point on the cover plate can be described as:

\[
Y = \frac{2}{3} Y_0 \left(1 - \frac{Y_0}{a} \right)
\]
Therefore,
\[ \xi = \frac{L \cdot \Delta}{2g} \]  \hspace{1cm} (3)
and,
\[ \alpha = \frac{\xi}{\Delta} = \frac{L}{2g} \]  \hspace{1cm} (4)

This is why the displacement of the cover plate is much larger than the tip displacement of bimorphs.

Shown in Fig. 2 is the transducer constructed from the bimorph-based double amplifier structure. Instead of two parallel bimorphs, two parallel bimorph arrays are used here and the cover plate is fixed on the top of the bimorph arrays so that larger radiation area is obtained. Loudspeaker paper is chosen as the cover plate because of its excellent mechanical and acoustic properties and lightweight. The other two lateral faces were also sealed by loudspeaker paper so that the transducer can work effectively as a monopole source.

![Diagram of the Double Amplifier Transducer](image)

**Fig. 1. Configuration of the Double Amplifier Transducer**

3. **SAMPLE PREPARATION AND MEASUREMENT TECHNIQUE**

Bimorph actuators can be operated in series (the poling directions of the two piezoelectric plates are opposite) or in parallel (the poling directions of the two piezoelectric plates are consistent) configurations. All the bimorphs used in this paper are in parallel structure. They are made from a kind of PZT 5H type piezoelectric material and the dimensions are 20.0 mm X 7.0 X 1.0 mm. The loudspeaker paper is from Nu-Way Company with the thickness of 0.56 mm. The experiments were conducted both on the double amplifier structures and the transducer. For the double amplifier structure experiment, the bimorphs are mechanically clamped at one end, and the cover plate is fixed on the top of bimorphs by super glue. The displacement of middle point and tip point are measured by using MTT 2000 Potentiometric Sensor under quasistatic and dynamic conditions. Fig. 3 is the schematic drawing of the measurement setup. The resonant frequency of the structure was determined by an HP-4194 Impedance Analyzer. For the transducer study, the bimorph arrays were fixed in the slots on the plastic substrate by super glue. The displacement at the central point (point A in Fig. 2) of the cover plate and at the intersection point of the central line and edge of the cover plate (point B in Fig. 2) was measured. The sound pressure level of the transducer was also studied in an anechoic chamber by using B & K 4135 type condenser microphone. In measurement, the
transducer is baffled in a large rigid plane and the measurement points are on the axis which goes through the central point and perpendicular to the substrate as shown in Fig. 4. The distances between the measurement points and the central point are 5.0mm and 1.0m respectively. The former situation is called near field and the latter is far field.

Fig. 3. Schematic Drawing of the Displacement Measurement Setup

Fig. 4. Schematic Drawing of Sound Pressure Level Measurement Setup

4 RESULTS AND DISCUSSIONS

4.1 Displacement and resonant frequency of the double amplifier structures

The dependence of the middle point displacement and amplification factor on the height of the cover plate was evaluated under quasistatic condition (driving frequency = 1Hz) for the double amplifier structures with L = 50.0mm and 60.0mm (see Fig. 1) respectively, and the results were presented in Fig. 5. As the height of cover plate g decreases, the middle point displacement increases first, and then decreases. The largest displacement was obtained when the distance between bimorphs was 50mm and the height of cover plate was 1.0mm. The largest displacement can reach 450µm (0.450mm) with a driving voltage of 100V (peak-peak value). Fig. 5 also reveals that the amplification factor always increases as the height of the cover plate is reduced. The amplification factor can reach more than ten easily. When the driving voltage is 100V (peak-peak value), the tip displacement of the bimorphs without loading is 37.63µm, which is called free tip displacement. Hence the middle point displacement can be more than ten times larger than the free tip displacement of the bimorphs. Fig. 6 shows the
displacement of the middle point as a function of the driving voltage. The displacement of the middle point increases almost linearly with the driving voltage. It is expected that the displacement of the middle point can reach about 1.5mm when the driving voltage is raised to 300V (peak-peak value) or 110V (rms value).

Equations (1) to (4) show that the middle point displacement should always increase as the cover plate height is reduced, so does the amplification factor. However, when the height of the cover plate decreases, the effective loading of the cover plate will increase, and a larger force is needed to push the cover plate. The relationship between the generated force and tip displacement of bimorph can be described as:

\[ F = \frac{Y_0 s^2 w}{4h^3} \left( \frac{3d_{31} h^2 E}{2s} - \Delta \right) \]  

(5)

where \( Y_0 \) is Young's modules of ceramic material;
\( d_{31} \) is piezoelectric constant of ceramic material;
\( w \) is width of bimorph;
\( E \) is applied electric field.

When a larger force is needed, the tip displacement of bimorph will reduce. Therefore, when the height of cover plate decreases, there are two competing factors which influence the middle point displacement of the cover plate. One is that the middle point displacement will increase according to the geometric relationship, so does the amplification factor. Another one is that the tip displacement of bimorph will decrease, which will reduce the middle point displacement. Therefore, there is an optimum point where the maximum value of the middle point displacement is obtained. However, the amplification factor will always increase when the cover plate height decreases as long as the geometric relationship is valid. The similar situation will happen when the distance between the bimorphs is changed.
Fig. 6. Displacement of the Middle Point as a Function of Driving Voltage (f=1 Hz, L=50mm, g=1.0mm)

Fig. 7. Displacement of the Double Amplifier Structure (f=1 Hz, Vp=100V)
Besides these two factors, there is an additional factor should be considered is the buckling of the cover plate. If the buckling of the cover plate happens, the geometric relationship (equation (1)) will not be hold, and the middle point displacement of cover plate will become smaller.

The tip displacement of bimorphs, the measured displacement of the middle point, and the calculated displacement of the middle point according to equation (1) are presented in Fig. 7. The tip displacement decreases rapidly when g is below 1.0mm for L = 50.0mm and 1.5mm for L = 60.0mm, which results in a decrease in the middle point displacement. When g is larger than 1.0mm for L = 50mm and 1.5mm for L = 60mm, there is no buckling of the cover plate since the calculated middle point displacement is nearly the same as the measured value. Below these values buckling of the cover plate occurs.

The dependence of the resonant frequency on the height of the cover plate is shown in Fig. 8 for L = 50.0mm and 60.0mm. For this amplifier structure, the cover plate can be considered as an effective mass loading attached on the top of the bimorph beams, which reduces the resonant frequency of the bimorph beams. As the height of the cover plate decreases and the distance between bimorphs increases, the effective loading of the cover plate increases, resulting in a decrease of the resonant frequency. This is consistent with the experimental results.

![Graph showing Resonant Frequency vs. Height of the Cover Plate](image)

**Fig. 3: Resonant Frequency of the Double Amplifier Structure**

The relationship between the displacement and frequency is shown in Fig. 9 for L = 50.0mm, g = 1.0mm. From the comparison of the tip displacement of bimorphs, the calculated displacement of middle point, and the measured displacement of middle point, it can be concluded that the cover plate can keep its shape from buckling during vibration in the frequency range up to about two times of the resonant frequency, that is, the transducer can work effectively in the frequency range up to the two times of the resonant frequency.

### 4.2 Displacement and sound pressure level of the double amplifier transducer

Based on the experimental results and analyses of the double amplifier structures, the bimorph-based double amplifier transducer was constructed. The dimensions of the transducer are:
- the distance between the bimorphs L = 50.0mm;
- the height of the cover plate g = 1.5mm;
- the width of the bimorph array W = 60.0mm;
- the height of bimorphs h = 18.0mm.

Therefore, this transducer can be expected to work effectively up to 1000 Hz, which is the frequency range desirable for the active noise control.
Fig. 9. Dependence of the Displacement on Frequency ($L=50\text{mm}, g = 1.0\text{mm}, V_{p-p} = 100\text{V}$)

Fig. 10. Frequency Dependence of the Displacement of the Transducer ($V_{p-p} = 100\text{V}$)
The dependence of the displacement of the point A and point B of the cover plate (see Fig. 2) on frequency is shown in Fig. 10. Due to clamping effect of the two lateral faces which are made of loudspeaker paper and are inactive, and due to the loading effect of the air in the closed chamber of the transducer, the displacement of point A is some smaller than that of the middle point displacement of its double amplifier structure with the same L and g values, and the displacement of point B is smaller than that of point A, but the displacement of the cover plate is still much larger than the tip displacement of bimorphs. The height of the peak in the resonant frequency also becomes weaker and smooth due to these clamping and loading effects.

The frequency dependence of the sound pressure level (SPL) was presented in Fig. 11. The SPL of the near field can be larger than 90dB in the frequency range from 50 to 1000Hz, and the largest value can reach more than 130dB. For the far field, the SPL can reach more than 80dB when the frequency is higher than 200Hz, which is suitable for active noise control. Larger SPL value can be easily achieved by increasing the width of the bimorph array. Another approach to get larger SPL value is to construct the transducer with multi-double amplifier structure, as shown in Fig. 12.

5 CONCLUSIONS

Bimorph-based double amplifier structure and the transducer based on this structure have been constructed and studied in this paper. It was demonstrated that with relatively low voltage, the displacement of the active face can reach millimeter scale, which is more than ten times larger than the tip displacement of bimorphs. The sound pressure level of the piezoelectric transducer can be larger than 90dB (near field) between 50 to 1000Hz and 80dB (far field) between 200 to 1000Hz. And the largest value can reach 130dB (near field). Because of its light weight and panel structure, it is potential to use this new piezoelectric air transducer in active noise control.

As the first result of the bimorph-based double amplifier transducer, further improvements in transducer performance are expected using improved materials and design. Instead of bimorph, some other ceramic actuators such as unimorph or monomorph can also be used to double amplifier structure and transducer design.

6 ACKNOWLEDGMENT

This work was supported by the Office of Naval Research under the contract No. N00014-94-1-1140.

7 REFERENCES


4. K. Uchino, Piezoelectric Electroceramic Actuator, Chapter 4, Morikita Publishers, Tokyo, Japan, 1986.


7. NW 3200A Loudspeaker Paper, Nu-Way Speaker Products, Axtel, Illinois.


9. Cartridge Type 413S 1/4" Condenser Microphone, Bruel & Kjaer Instruments, Marlborough, Massachusetts.


Fig. 11. Sound Pressure Level of the Transducer
Fig. 12. Configuration of the Transducer Constructed from Multi-Double Amplifier Structure (the arrows represent the moving direction)
APPENDIX 2
Bimorph-based piezoelectric air acoustic transducer: model

V.D. Kugel *, Baomin Xu, Q.M. Zhang, L.E. Cross

Materials Research Laboratory, The Pennsylvania State University, University Park, PA 16802, USA

Received 23 October 1996; received in revised form 6 February 1998; accepted 16 February 1998

Abstract

A new type of bimorph-based piezoelectric air transducer with the working frequency range of 200-1000 Hz has recently been developed [B. Xu, Q. Zhang, V.D. Kugel, L.E. Cross. Piezoelectric air transducer for active noise control. Proc. SPIE, 2717 (1996) 388-398]. In the present work, basic acoustic characteristics of this device and its piezoelectric elements are analyzed. To model the vibration spectrum of the transducer, a one-dimensional approach is developed where inertia, elastic and damping forces are included. Analytical equations describing mechanical vibrations and electrical impedance of piezoelectric bimorph cantilevers under external forces are derived. In order to describe various losses in the transducer, complex piezoelectric, dielectric, and elastic constants are used. Results of the modeling are in good accord with experimental data. The suggested model can be used for device optimization. © 1998 Elsevier Science S.A. All rights reserved.

Keywords: Piezoelectricity; Piezoelectric transducers; Bimorph; Air acoustics

1. Introduction

Piezoelectric ceramics offer many advantages as sensors and actuators. Relatively high coefficient of electromechanical coupling makes it possible to use the material in ultrasonic devices and tweeters. However, in low frequency applications such as air acoustics, a larger displacement is often required, which is far beyond the range reachable by ceramics materials. In order to overcome this problem, a new type of bimorph-based piezoelectric transducer has recently been suggested [1]. The main peculiarity of this device is that through a double amplification scheme a large displacement can be generated which makes it suitable for acoustic applications at low frequency such as from 200 to 1000 Hz. It has been demonstrated that the device can produce a sound pressure level of 90-100 dB which makes this device very attractive for active noise control and a flat panel acoustic source [1].

The purpose of this study was to develop a model capable of characterizing the vibration behavior and structure-performance relationship of this transducer. The paper is structured as follows. In Section 2, an approach that will be used to describe this bimorph-based piezoelectric air transducer is discussed. In Section 3, analytical equations, describing mechanical vibrations and the electrical impedance of piezoelectric bimorph cantilevers, are derived. In Section 4, calculation of various forces acting in the transducer is carried out. Results of the device modeling and comparison with experimental data are presented in Section 5. Finally, in Section 6 a summary and conclusions are given.

2. Model

The piezoelectric transducer to be analyzed consists of two arrays of piezoelectric bimorph cantilevers bridged by a curved diaphragm (Fig. 1): inactive lateral faces are covered with non-radiating plates. The diaphragm serves for acoustic matching: it amplifies the tip displacement of the cantilevers and it increases the sound-emitting area. The amplification factor $K_{amp}$ is defined as the ratio of the diaphragm apex $\eta_d$ to the tip displacement of the bimorph cantilever $\eta(l)$, where $l$ is the cantilever length (Fig. 1). A typical amplification factor lies in the range of 5 to 20. In a current design, the bimorphs are made from soft piezoelectric ceramics PZT5H (Morgan Marcor) and a loudspeaker paper is used as a triangular-shaped diaphragm. Experimental results demonstrated that, if the amplification factor is less than 10 and the length of the diaphragm is less than 50 mm, the diaphragm can be considered as a rigid plate without buckling at frequencies below 800-1000 Hz. Therefore, for calculating vibrations of the diaphragm the quantities to be evaluated are: the displacement of the moving end of the

Fig. 1. Schematic view of bimorph-based piezoelectric air transducer with double amplification.

Diaphragm

Piezoelectric Bimorphs

Base Plate

Fig. 2. Diagram of the forces included in one-dimensional model. P denotes the vector of spontaneous polarization in the piezoelectric bimorph cantilever.

that is created by the motion of the mass of the diaphragm. elastic forces \( F_e \) that are caused by stiffness of the air enclosed in the cabinet and the stiffness of hinges. The last force is the air damping force, \( F_{\text{damp}} \), caused by the acoustic radiation. The force diagram is shown in Fig. 2 where \( R_1 \) and \( R_2 \) are the reaction forces and \( F_{\text{push}} \) is the force generated by the bimorph cantilever, which drives the diaphragm. Introducing \( F_{\text{push}} \), makes it possible to separately analyze vibrations of the bimorph cantilever and the motion of the diaphragm. In the model, it is assumed that the hinges for the diaphragm fastening generate an elastic force only.

3. Analysis of piezoelectric bimorph cantilever

To define bending vibrations of the piezoelectric bimorph cantilever, the equations describing vibrations of the neutral surface \( \eta \) of the vibrating beam are used [2]. In the case of the piezoelectric beam, the neutral surface is defined in such a way that any extension strain, even if allowed to exist there, does not contribute a bending moment [2]. For uniform symmetric bimorphs (in which the thicknesses of two piezoelectric plates are equal), the neutral surface coincides with the plane of symmetry, which is perpendicular to the \( Z \) axis (Fig. 2). The equations for bending of uniform symmetric piezoelectric bimorph beams with rectangular cross-section can be written as (Fig. 2):

\[
\begin{align*}
S_y &= -\varepsilon_{33} \frac{\partial \sigma_y}{\partial y}, \quad \text{Mechanical strain of the neutral plane describing bending} \ [2]. \\
S_z &= -\varepsilon_{33} T_1 - \varepsilon_{33} \sigma_z, \quad -\eta < z \leq \eta \quad \text{Piezoelectric constitutive equation,} \\
\varepsilon_y &= -\varepsilon_3 \frac{U}{h}, \quad 0 < z \leq \eta \quad \text{Electric field across bimorph with parallel connection,} \\
\varepsilon_z &= \varepsilon_3 \frac{U}{h}, \quad -\eta \leq z \leq 0.
\end{align*}
\]

where \( S_y \) is the mechanical strain along the \( Y \) axis, \( \varepsilon_{33} \) is the mechanical compliance, \( T_1 \) is the mechanical stress, \( \varepsilon_3 \) is the piezoelectric coefficient, \( E \) is the electric field across the sample, \( U \) is the applied voltage, and \( 2h \) is the bimorph thickness. A coordinate system \( XYZ \) is chosen such that this plane passes through \( z = 0 \) and the bimorph cantilever is clamped at \( y = 0 \) (Fig. 2). The translation force \( F_z \) that causes the displacement of the neutral plane in the \( Z \) direction (Fig. 2) is [2]

\[
F_z = \frac{\partial F_z}{\partial y} dy.
\]

where \( F_z \) is the internal shear force along the \( Z \) axis [2]:

\[
F_z = -\frac{\partial M}{\partial y}
\]

where \( M \) is the internal bending moment along the \( X \) axis:

\[
M = -\int \int z T_2 dx dz.
\]
By using Eqs. (2) and (3) the law of linear momentum conservation, i.e.,

\[ F_t = \rho \left( 2hw \right) \frac{\partial^2 \eta}{\partial t^2} \]  

where \( \rho \) is the density of the bimorph, \( w \) is the width of the bimorph beam, and \( 2hw \) is the area of the cross-section of the bimorph, can be written as

\[ \frac{\partial^2 M}{\partial y^2} = \rho \left( 2hw \right) \frac{\partial^2 \eta}{\partial t^2} \]  

(6)

Substitution of \( T_2 \) from set (1) into Eq. (4) yields

\[ M = \frac{2wh^3}{3\varepsilon_2} \frac{\partial^2 \eta}{\partial y^2} - \frac{Uwhd_{33}}{\varepsilon_2} = M_{ex} = 0 \]  

(7)

Thus, one can see that in the case of the symmetrical bimorph beam the external electric tension generates a pure bending moment only. Substituting this expression into Eq. (6) gives

\[ \frac{\partial^2 \eta}{\partial y^2} - \frac{3\varepsilon_2}{2h} \frac{\partial^2 \eta}{\partial t^2} = 0 \]  

(8)

To find a solution of Eq. (8) the following boundary conditions should be used [3]:

\[ \begin{cases} 
\eta = 0; & \eta_y = 0; \\
\eta_y = \phi(y); & \eta = \phi'(y) = 0; \\
\eta = \phi(y) + \phi'(y) = 0; & \eta_y = \phi''(y) = 0.
\end{cases} \]  

(9)

where \( M_{ex} \) is the \( r \) component of the external angular momentum applied at \( y = l \) and \( F_{ex} \) is the \( z \) component of the external force acting at the vibrating end of the cantilever. In our derivations, the effect of forces acting along \( Y \) and \( X \) axes was neglected. Before proceeding further with calculations of force \( F_{ex} \) for the transducer under consideration, a general expression for vibrations of the piezoelectric bimorph cantilever under the applied voltage and external forces will be derived. If the harmonic electric voltage \( U_{me} e^{j\omega t} \) and external forces \( F_{ex} = F_{ex} e^{j\omega t} \) applied to the cantilever are linear functions of the displacement, the displacement of the cantilever can be written as \( \eta = \eta_m e^{j\omega t} \) where \( \omega \) is the angular frequency. Therefore, Eq. (8) can be written as

\[ \frac{\partial^2 \eta_m}{\partial y^2} - \lambda^2 \eta_m = 0 \]  

(10)

where

\[ \lambda^2 = \frac{3\varepsilon_2^2 \rho}{2h^2} \]  

(11)

The solution of Eq. (10) with boundary conditions given by Eq. (9) is

\[ \eta = \text{const.} \frac{3}{2} \frac{dU_m}{k^2} \left( \cos(\lambda t) + \cos(\lambda t) \right) - \frac{3}{2} \frac{dF_{ex}}{k^2} \left( \sin(\lambda t) + \sin(\lambda t) \right) \]  

(12)

The resonant frequency \( \nu_m \), which is determined from zero value of the denominator in Eq. (12), coincides with the resonant frequency of a free piezoelectric cantilever:

\[ \frac{\sinh(\lambda t) \cos(\lambda t)}{\lambda^2 t^2 - \sin(\lambda t) \cos(\lambda t)} = -1 \]  

(13)

In the low-frequency limit \( (\lambda t \rightarrow 0) \) Eq. (12) may be rewritten as

\[ \eta_m = \text{const.} \frac{3}{2} \frac{dU_m}{k^2} \left( \frac{\sinh(\lambda t) - \sin(\lambda t)}{(2\lambda t)^2} \right) \]  

(14)

where \( K \) is a complex coefficient. As will be shown later, this coefficient is directly related to the mass (inertia forces), spring constant (elastic forces), and damping coefficient (damping forces) associated with the applied external force.

Substitution of the force amplitude from Eq. (15) into Eq. (1) produces at \( y = l \)

\[ \eta_m = \frac{1}{2} \frac{dU_m}{2h^2} \]  

(15)

By using Eqs. (12), (15) and (16), a general expression for \( \eta_m \) is written as

\[ \eta_m = \frac{1}{2} \frac{dU_m}{2h^2} \]  

(16)
Eq. (17) reveals that the external force of the type (Eq. (15)) changes the resonant characteristics of the vibrating cantilever. If the piezoelectric material has no losses and the external force does not tend to damp vibrations of the bimorph cantilever, i.e., \( K \) is real, the resonant frequencies may be calculated from the following equation

\[
\lambda^2 \left( 1 + \cosh(\lambda l) \cos(\lambda l) \right) - \frac{3 \epsilon_{33} K}{2 \omega h^2} \left( \cosh(\lambda l) \sin(\lambda l) \right)
\]

\[
- \sinh(\lambda l) \cos(\lambda l) = 0
\]

(18)

Apparently, the amplitude of the tip displacement is equal to zero when \( \sin(\lambda l) = 0 \) and the value of the corresponding frequency does not depend on external loading. It can also be shown that the lowest frequency of the zero vibrations is about 2.8 times higher than the fundamental frequency of bending vibrations of unloaded cantilever \( \lambda = 0 \) (Eq. (9)) is

\[
\frac{\epsilon_{33} T}{\epsilon_{33} S} \lambda (1 + \cosh(\lambda l) \cos(\lambda l)) \sinh(\lambda l) = 0
\]

The resonant frequencies may be calculated from the following equation

\[
\omega = \sqrt{\frac{\epsilon_{33} T}{\epsilon_{33} S} \lambda (1 + \cosh(\lambda l) \cos(\lambda l)) \sinh(\lambda l) = 0}
\]

(18)

Using Eqs. (1), (17) and (25) one obtains

\[
Y = j \omega \left( \frac{\epsilon_{33} T}{\epsilon_{33} S} \right) \left( \lambda (1 + \cosh(\lambda l) \cos(\lambda l)) \sinh(\lambda l) = 0 \right)
\]

(25)

At frequencies far below the fundamental bending resonance and without applied external force \( K = 0 \), the expression in parentheses in Eq. (26) reduces to \( \lambda \left( 1 + (1/2)(d_{33}^2/\epsilon_{33} S^2) \right) \). Therefore, for PK1550 ceramics, as follows from Table 2, the corresponding impedance should increase by \( \sim 5\% \) as compared to that for the bulk material with the same dimensions. Instead of that, the experimental value for the bimorph structure was by \( \sim 8\% \) lower than that for the bulk material. The reason for this is that in the theoretical calculations an approximate electric field \( E \) (Eq. (1)) was used and the effect of bending along the X axis (Fig. 2) was not taken into account.

4. Calculation of vibrations of the piezoelectric cantilever loaded with the diaphragm

To evaluate vibrations of the piezoelectric cantilever loaded with diaphragm (Figs. 1 and 2) by means of Eq. (16), the external mechanical force \( F \) acting on the vibrating end of the cantilever along the Z axis should be determined (forces acting along the Y axis are neglected). This force may be calculated using the laws of angular and linear momentum conservation [6] for the diaphragm:

\[
r \times (R_1 - R_2 - F_{\text{push}} + F_m) + \int_{\text{diaphragm}} r \times F_m dV = 0
\]

\[
\int_{\text{diaphragm}} r \times F_m dS = \frac{d}{dr} \int_{\text{diaphragm}} (r \times p) v dV
\]

\[
R_1 - R_2 - F_{\text{push}} + F_m = \int_{\text{diaphragm}} F_m dV = \int_{\text{diaphragm}} F_m dS
\]

(27)

Electrical current through the bimorph can be written as

\[
l = \frac{dQ}{dr} = j \omega Q
\]

(24)
where \( v(r) \) is the diaphragm velocity. Assuming that the thickness of the diaphragm is small with respect to its length, volume integrals in Eq. (27) can be transformed to surface integrals. The air damping force (surface density) \( F_{am} \) can be written as (7):

\[
F_{am} = -k_c v
\]

(28)

where \( k_c \) is the damping coefficient depending on the shape of the transducer (it is directly related to the acoustic impedance of the transducer (7)). By substituting various quantities, the set of Eq. (27) in \( X'Y'Z' \) coordinate system can be transformed to

\[
F_{pum} = F_{ai} - R_1 + k_c wL_d^2 \sin \varphi \frac{d\varphi}{dr}
\]

\[
= -\rho \frac{wL_d^2}{2} \sin \varphi \frac{d^2\varphi}{dr^2} + \cos \left[ \frac{d\varphi}{dr} \right]^2 Z \text{ axis}
\]

\[
R_2 - P_m - k_c wL_d^2 \cos \varphi \frac{d\varphi}{dr}
\]

\[
= \rho \frac{wL_d^2}{2} \cos \varphi \frac{d^2\varphi}{dr^2} - \sin \left[ \frac{d\varphi}{dr} \right] \cos \varphi Y \text{ axis}
\]

\[
R_1 \sin \varphi - R_2 \cos \varphi = P_m \cos \varphi - k_c wL_d \frac{d\varphi}{dr}
\]

\[
= -\rho \frac{wL_d^2}{2} \cos \varphi \frac{d^2\varphi}{dr^2} + \frac{wL_d^2}{6} \frac{d\varphi}{dr} \quad X' \text{ axis}
\]

(29)

where \( P_m \) is the weight of one half of the diaphragm with the width \( w \), and \( \varphi \) is the angle between the diaphragm and \( Z' \) axis (Fig. 2). From the set of Eq. (29) one can solve for the magnitude of \( F_{pum} \) acting along the \( Z' \) axis:

\[
F_{pum} = \frac{F_m}{2} \csc \varphi + \frac{1}{3} k_c wL_d \frac{1}{\sin \varphi} \frac{d\varphi}{dr}
\]

\[
- \frac{1}{2} \rho \frac{wL_d^2}{2} \frac{1}{\sin \varphi} \frac{d^2\varphi}{dr^2} + F_n
\]

\[
F_{pum} = -k F_{pum}
\]

(30)

where \( k \) is the unit vector along the \( Z \) axis. The right-hand side of the first equation in Eq. (30) is the external force \( F_{ex} \) acting at the moving end of the bimorph cantilever parallel to the \( Z \) axis since, according to Newton’s third law, \( F_{ex} = -F_{pum} \) and the \( Z \) and \( Z' \) axes coincide. Clearly, the external force \( F_{ex} \) is a non-linear one in a general case since \( ctn \varphi \) and \( \varphi \) are non-linear functions of \( \varphi \) and, hence, displacement \( \gamma \). For an air acoustic transducer it is desirable to work in a linear regime when the amplitude of the diaphragm vibration is much smaller than the diaphragm height \( h_d \) (Fig. 1). In this case:

\[
\frac{d\varphi}{dr} = \frac{1}{\sin \varphi} = \frac{L_d}{h_d}
\]

(31)

\[
K_{amp} = \frac{L_d}{h_d}
\]

The displacement of the diaphragm apex \( \eta_d \) is

\[
\eta_d = K_{amp} \eta(l)
\]

(32)

We also assume that the elastic force is linear:

\[
F_{ai} = -k_{spr} \eta(l)
\]

(33)

where \( k_{spr} \) is the equivalent spring constant of the air enclosed in the cabinet and hinges (Fig. 1). Thus, as follows from Eqs. (30)−(33), \( F_{ex} \) can be expressed as:

\[
F_{ex} = K_{amp} \frac{F_m}{2} - \frac{1}{3} K_{amp} \left( k_{amp} wL_a \frac{d\eta}{dr} + \rho \frac{wL_d}{2} \frac{d^2\eta}{dr^2} \right)
\]


\[
\eta = -k_{spr} \eta(l)
\]

(34)

Since, for harmonic excitation, the tip displacement of the bimorph cantilever can be written as \( \eta_m e^{i\omega t} \), Eq. (34) can be rewritten as:

\[
F_{ex} = K_{amp} \frac{F_m}{2} - \eta_m(l) e^{i\omega l} \left( -\frac{1}{3} K_{amp} (k_{amp} wL_a \omega) + \rho \frac{wL_d}{2} \frac{d^2\eta}{dr^2} \right)
\]

(35)

The first term in Eq. (35) produces constant deflection only; therefore it can be neglected in calculations of the spectrum of vibrations. In comparing Eqs. (15) and (35), one can find coefficient \( K \) in the expression of the tip displacement (Eq. (16)):

\[
K = \left( \frac{1}{3} K_{amp} (k_{amp} wL_a \omega) - \rho \frac{wL_d}{2} \frac{d^2\eta}{dr^2} \right)
\]

(36)

Thus, vibrations of the bimorph cantilever and diaphragm can be calculated by means of Eqs. (11), (16), (32), and (36).

5. Results of modeling

The strategy in calculating the spectrum of the diaphragm vibrations was as follows. Initially, the electromechanical properties of the piezoelectric bimorphs were determined by experimentally studying characteristics of the ceramics and vibration spectrum of unloaded piezoelectric bimorph cantilevers. Then, these data were used to calculate the vibrations of the cantilevers loaded with the diaphragm. Since the equivalent spring constant \( k_{spr} \) of the air enclosed in the cabinet and hinges was unknown, its value was determined experimentally by measuring the diaphragm vibrations at low frequencies under the assumption that \( k_{spr} \) is frequency independent. As follows from Eqs. (19), (32) and (36), the
magnitude of low-frequency vibrations of the diaphragm apex is directly related to the spring constant:

$$\eta = \frac{3}{2} d_{32} C_m \left( \lambda \right)^2 \frac{1}{\hbar} \varepsilon_0 k_{s-def} \left( \frac{l}{h} \right)^2 K_{\text{air}} \left( \frac{l}{h} \right)^2$$  \hspace{1cm} (37)

To determine the damping coefficient $k_\text{air}$, it was assumed that the vibrating diaphragm behaves as a baffled piston [7]:

$$k_\text{air} = 2 \rho_\text{air} c_\text{air} \left( \frac{2 \pi \nu \nu_\text{air}}{2 c_\text{air}} - \frac{16 \nu \nu_\text{air}}{3 c_\text{air}} \right)$$  \hspace{1cm} (38)

where $\rho_\text{air}$ is the air density, $c_\text{air}$ is the sound velocity in the air, $\nu$ is the frequency, and $r_\text{eq}$ is the equivalent radius of the transducer diaphragm with the area $A = 2 L_w r_\text{eq} = \sqrt{A} / \pi$.

Thus, the procedure described above made it possible to determine all parameters essential for calculating the vibration soeca-am of the transducer. Characteristics of the diaphragm and piezoelectric bimorph cantilevers used in the transducer are given in Table 1 [1]. It is important to note that complex piezoelectric, dielectric, and elastic constants of the ceramics used in the bimorph cantilevers should be determined for piezoelectric losses.

Soft piezoelectric ceramics were used for the bimorph fabrication [1]. It is known that the electromagnetic properties of these piezoelectric bimorph cantilevers are highly dependent on the magnitude of the applied electric field [8]. Calculated electromagnetic coefficients of PZT ceramics using data of Refs. [8,9] for the electric field $E = 0.71$ kV/cm are shown in Table 2. Frequency dispersion of these coefficients was neglected in the device modeling since it was not significant in the frequency range of interest. The calculated vibration spectrum of the unloaded bimorph cantilever is shown in Fig. 3. The resonant frequency at 0.71 kV/cm was $\nu_r = 10.4$ kHz and the mechanical quality factor was $Q_m = 101$. At small electric fields ($E < 0.01$ kV/cm) the corresponding parameters $\nu_r = 1167$ and $Q_m = 55$ were the same as the experimental data [1]. It should be noted that the vibration amplitude at the resonance and, consequently, $Q_m$ depend on the mechanical losses. Using notations $d_{32} = |d_{32}| e^{-\gamma}$ and $s_{33}^2 = |s_{33}^2| e^{-\gamma}/h$, and Eq. (11), Eq. (16) under conditions $K = 0$ (the unloaded bimorph) and $\theta_\text{d}, \theta_\text{d} \rightarrow 0$ (low losses) transforms to:

$$\eta_m(l) = \frac{3}{2} d_{32} C_m \left( \frac{l}{2 h} \right)^2 \left( \sinh \alpha l - j \alpha l \frac{d_3}{4} (\cosh \alpha l \sin \alpha l - \sinh \alpha l \cos \alpha l) \right)$$

where $d_3$ is the phase of the piezoelectric coefficient $d_{32}$ and $\theta_\text{d}$ is the phase of the mechanical compliance $s_{33}^2$. At the fundamental frequency of bending vibrations ($\nu_r = 1.875$), Eq. (39) gives:

$$\eta_m(l) = 3 C_m \left( \frac{l}{2 h} \right)^2 \left| d_{32} \right|^2 \frac{0.89 \varepsilon_0}{\theta_\text{d}} e^{-\gamma} \left| s_{33}^2 \right|^2 \left( \cosh \alpha l - 1 \right)$$

Thus, the amplitude of the resonant displacement of the unloaded cantilever is inversely proportional to the phase of the mechanical compliance.

### Table 1

Diaphragm and bimorph cantilever characteristics

<table>
<thead>
<tr>
<th>Parameters of the loudspeaker paper diaphragm</th>
<th>Dimensions of the piezoelectric bimorph cantilevers</th>
</tr>
</thead>
<tbody>
<tr>
<td>$A_\text{d} (\text{mm})$</td>
<td>$L_\text{d} (\text{mm})$</td>
</tr>
<tr>
<td>2.3</td>
<td>22</td>
</tr>
</tbody>
</table>

### Table 2

Magnitude of electromechanical coefficients of PZT5H for $E = 0.71$ kV/cm

<table>
<thead>
<tr>
<th>$d_{32} / \varepsilon_0$</th>
<th>$d_{32} (\text{m/V})$</th>
<th>$S_{33}^2 (\text{mN})$</th>
</tr>
</thead>
<tbody>
<tr>
<td>5770 \cdot 0.99 \cdots 3770</td>
<td>333 \cdot 10^{-12}(0.996 \cdots 0.091)</td>
<td>50.3 \cdot 10^{-12}(0.996 \cdots 0.088)</td>
</tr>
</tbody>
</table>
The calculated vibration spectrum of the diaphragm apex is shown in Fig. 4. The magnitude of the equivalent spring constant (N/m) shown in Fig. 4 is for the width of the part of the diaphragm driven by one bimorph (w = 5 mm). Clearly, at the low loss limit the value of the resonance frequency derived from the model does not depend on the losses of the equivalent spring constant and it reproduces the experimental values \( v_r = 330 \) Hz. Nevertheless, only complex \( k_{spr} \) can fit the amplitude of the diaphragm vibrations at the resonance. These results indicate that the elastic force caused by hinges and the air enclosed in the cabinet has losses. Thus, the model developed is capable of describing basic characteristics of the vibration spectrum of the transducer.

As seen from Fig. 4, the width of the experimental peak is narrower than that of the theoretical ones. This may be caused by an increase in the amplification factor of the diaphragm (Eq. (32)) around the resonant frequency that was observed experimentally. Nonlinear dependence of electromechanical properties on the level of vibrations may be another reason for this discrepancy. The data of Fig. 4 show that the model does not describe the minimum in the diaphragm vibrations observed near 300 Hz. An experimental study of the frequency dependence of the amplification factor \( K_{amp} \) of the diaphragm shows that the diaphragm has its own bending resonance near this frequency, i.e. it does not keep its shape. Another factor that can affect the diaphragm vibrations at high frequencies is losses in the hinges.

The model developed makes it possible to analyze mechanical vibrations and sound pressure generated by the transducer for various dimensions of the diaphragm and piezoelectric elements. Vibration characteristics and on-axis sound pressure at the distance \( r = 1 \) m for different thicknesses \( 2h \) of cantilevers and \( E = 0.71 \) kV/cm are shown in Fig. 5. A complex value of the spring constant \( k_{spr} = 9500(0.916 - 0.4j) \) N/m was used in the modeling. The sound pressure \( \rho \) of the transducer was calculated assuming that the transducer behaves as a baffled piston [7]:

\[
\rho = \frac{2\pi r^2 A_{out} \bar{\eta}_d}{r}
\]

where \( \bar{\eta}_d \) is the average amplitude (rms) of the diaphragm vibrations. \( \bar{\eta}_d = 0.5 \eta_d \). The calculated value of the sound pressure for \( 2h = 1 \) mm is close to the experimental one [1]. It should be noted that the results of the model (Fig. 5) are valid for the given equivalent spring constant only. As seen from Fig. 5, the optimal value of the bimorph thickness lies between 1 and 1.5 mm. Clearly, the sound pressure developed by the transducer sharply decreases below 300 Hz. Data of Fig. 5 show that the resonant frequency of the transducer is highly dependent on the thickness of piezoelectric elements.

The model developed also makes it possible to analyze the change in the resonant frequency of a double-amplifying structure [1]. The structure consists of two piezoelectric bimorph cantilevers bridged by a triangular diaphragm (as in the air transducer but much narrower in the x direction and without non-radiating plates covering inactive sides). Assuming that the spring constant and air-damping coefficient in this structure are equal to zero, Eq. (18), using Eqs. (11) and (36), reduces to

\[
\lambda^3 \left[ 1 - \cosh(\alpha) \cos(\alpha) - \frac{1}{3} K_{amp} \frac{m^2}{m_0} \alpha (\cosh(\alpha) \sin(\alpha) - \sinh(\alpha) \cos(\alpha)) \right] = 0
\]

where \( \alpha = \lambda l, m_0 = L_{eff} \rho_0 \) is the mass of one half of the
diaphragm with the width \( w \) and \( m = 2\pi \omega \rho \) is the mass of the piezoelectric cantilever. As follows from Eq. (11), the value \( \alpha \) obtained from Eq. (42) is directly related to the resonant frequency of the double-amplifying structure.

\[
\nu_r = \frac{\alpha^2}{4\pi^2} \frac{2h}{l^2} \frac{1}{\sqrt{s_{zz}^e \rho}}
\]

(43)

Results of modeling and experimental data for the structure with \( 2h = 1 \text{ mm} \), \( w = 7 \text{ mm} \), \( l = 20 \text{ mm} \), \( m = 1.092 \text{ g} \), \( L_2 = 25 \text{ mm} \), and \( L_3 = 0.057 \text{ g} \) are given in Fig. 5. As seen from Fig. 6, Eq. (43) describes experimental data for the apex height above 1.5 mm. Below this height the thickness of the diaphragm, \( \eta = 0.59 \text{ mm} \), becomes comparable with the height of the diaphragm apex that can increase the contribution of elastic forces and bending moments in the hinges.

6. Conclusions

In this work, a model describing basic acoustic characteristics of the bimorph-based piezoelectric transducer [1] has been developed. A one-dimensional approach was suggested in which inertia, elastic and damping forces are included. As an integral part of this model, analytical equations describing the spectrum of mechanical vibrations and electrical impedance of piezoelectric bimorph cantilevers under external forces were derived. In the model, complex electromechanical parameters of the transducer were used to include losses in the transducer. Results of modeling and experimental data are in good agreement. The model can be used for optimization of the acoustic characteristics of the transducer.

7. Nomenclature

- \( h_d \): height of the diaphragm apex
- \( L_d \): length of the diaphragm
- \( w_d \): width of the diaphragm
- \( t_d \): thickness of the diaphragm
- \( \eta_d \): displacement of the diaphragm apex
- \( \hat{\eta} \): average displacement of the diaphragm
- \( \rho_d \): density of the diaphragm
- \( m_d \): mass of the part of diaphragm acting on piezoelectric bimorph cantilever
- \( 2h \): thickness of piezoelectric bimorph cantilever
- \( l \): length of the piezoelectric bimorph cantilever
- \( w \): width of piezoelectric bimorph cantilever
- \( \eta \): displacement of the neutral plane of the piezoelectric bimorph cantilever
- \( \eta_m \): amplitude of the displacement of the neutral plane of the piezoelectric bimorph cantilever
- \( \eta_m^r \): amplitude of the displacement of the neutral plane of the piezoelectric bimorph cantilever at resonance
- \( \rho \): density of the piezoelectric bimorph cantilever
- \( m \): mass of the piezoelectric bimorph cantilever
- \( s_{zz}^e \): \( y \) component of the mechanical compliance of the piezoelectric bimorph cantilever
- \( d_{12} \): piezoelectric coefficient along the \( Y \) axis
- \( \theta_d \): phase angle of the piezoelectric coefficient \( d_{12} \)
- \( \theta_\gamma \): phase angle of the mechanical compliance \( s_{zz}^e \)
- \( \lambda \): wave coefficient describing bending vibrations of the piezoelectric bimorph cantilever
- \( \omega \): angular frequency
- \( \nu \): frequency
- \( \nu_r \): the fundamental frequency of bending vibrations
- \( \varphi \): angle between the diaphragm and \( Z' \) axis
- \( j \): imaginary unit
- \( r \): distance between the sound source and the measurement point
- \( r_{eq} \): equivalent radius of the diaphragm
- \( c_{\text{air}} \): sound velocity in the air
- \( k_v \): air damping coefficient
- \( \rho_{\text{air}} \): density of the air
- \( p \): sound pressure
- \( \alpha \): coefficient determining the resonant frequency of the loaded piezoelectric bimorph cantilever
- \( k_{\text{spr}} \): equivalent spring constant
- \( \tau \): time
- \( x, y, z \): rectangular Cartesian coordinates
- \( k \): unit vector along the \( Z \) axis
- \( r \): radius-vector in the \( X'Y'Z' \) coordinate system
- \( v \): diaphragm velocity
\( XYZ, X'Y'Z' \) Cartesian coordinate systems

- Area of the diaphragm: \( A \)
- Electrical field along the \( Z \) axis: \( E_z \)
- Electric displacement along the \( Z \) axis: \( D_z \)
- Electrical current through the piezoelectric bimorph cantilever: \( I \)
- Amplification factor of the diaphragm: \( K_{amp} \)
- Total electric charge on surfaces of the piezoelectric bimorph cantilever: \( Q \)
- Mechanical strain along the \( Y \) axis: \( S_y \)
- Voltage across the piezoelectric bimorph cantilever: \( U \)
- Amplitude of the voltage across the piezoelectric bimorph cantilever: \( U_m \)
- Electrical admittance of the piezoelectric bimorph cantilever: \( Y \)
- Elastic force acting along the \( Z' \) axis: \( F_{el} \)
- External force acting along the \( Z \) axis: \( F_{ext} \)
- Amplitude of the external force acting along the \( Z \) axis: \( F_{ext} \)
- Inertia force directed along the \( Y' \) axis: \( F_i \)
- Internal shear force along the \( Z' \) axis: \( F_s \)
- Internal translation force along the \( Z \) axis: \( F_t \)
- Complex force coefficient: \( K \)
- Internal bending moment along the \( X \) axis: \( M \)
- External bending moment along the \( X \) axis: \( M_{ext} \)
- Vector elastic force: \( F_{ele} \)
- Vector inertia force (volume density): \( F_{in} \)
- Vector air-damping force (surface density): \( F_{sam} \)
- Vector force driving the diaphragm: \( F_{push} \)
- Vector reaction forces: \( R_1, R_2 \)
- Vector of spontaneous polarization: \( P \)

Acknowledgements

This work was supported by the Office of Naval Research under the contract N00014-94-1-1140.

References


Biographies

Valery D. Kugel received his MSc degree in EE from Riga Polytechnical Institute, Latvia in 1984, and PhD degree in EE (Ferroelectricity) from Tel Aviv University, Israel in 1995. He is currently a Postdoctoral fellow at the Materials Research Lab of the Pennsylvania State University. His research interests involve ferroelectricity, piezoelectric, pyroelectric, and dielectric materials and devices for sensor, actuator and transducer applications.

Bao Min Xu received his BSc degree in materials science from Tsinghua University, China in 1986, and PhD degree in ceramics from Shanghai Institute of Ceramics, Chinese Academy of Sciences in 1991. He is currently a Postdoctoral fellow at the Materials Research Lab of the Pennsylvania State University. His research interests involve piezoelectric materials and devices for sensor, actuator and transducer applications, dielectric materials and ceramic processing. Dr. Xu is a member of the American Ceramic Society.

Q.M. Zhang is currently an Associate Professor of Electrical Engineering at the Materials Research Laboratory and Department of Electrical Engineering of the Pennsylvania State University. His research interests include ferroelectric ceramics, polymers, and composites for actuator, sensor, and transducer applications; new design and modeling of piezocomposites; characterization of piezoelectric and electrostrictive effects in ferroelectrics; smart materials and structures; effect of defect structure on electromechanical properties of ferroelectrics; structural studies of interfaces in ferroelectrics. Dr. Zhang is a member of IEEE, American Ceramic Society, and Materials Research Society.

L. Eric Cross is the Evan Pugh Professor of Electrical Engineering at the Pennsylvania State University. His research interests are in dielectric and ferroelectric crystals: piezoelectric and electrostrictive ceramics: composites for sensor, actuator and transducer applications, and as components in "smart" materials and structures. He has co-authored more than 500 technical papers, and sections of six books. Dr. Cross is a member of the National Academy of Engineering, a Fellow of the American Institute of Physics, of the American Ceramic Society, and the American Optical Society, and a member of the Japanese Physical Society. He is chairman of the IEEE Committee of Ferroelectrics, US representative for ferroelectrics on IUPAP, and a member of the Defense Sciences Council of ARPA.
APPENDIX 3

Optimization of Bimorph Based Double Amplifier Actuator
under Quasistatic Situation

Baomin Xu, Q. M. Zhang, V. D. Kugel, Qingming Wang, L. E. Cross

Materials Research Laboratory, Pennsylvania State University, University Park, PA 16802

Abstract - Bimorph based double amplifier actuator is a new type of piezoelectric actuation structure which combines both bending and flextensional amplification concepts. As a result, the displacement of the actuator can be more than ten times larger than the tip displacement of bimorphs and can be used in air acoustic transducers as an actuation element. This paper studied the dependence of displacement on actuator parameters and optimum design issues for the cover plate (the flextensional part of the actuator) theoretically and experimentally.

I. INTRODUCTION

How to get larger displacements is always a main objective in the development of piezoelectric transducer and actuator devices. Since the direct extensional strain in most piezoelectric ceramic materials is at best a few tenths of one percent, the means of enhancing or amplifying the displacement is essential in many device designs[1].

Except for multi-layer type actuators, which enhance the displacement by a direct dimension effect, presently there are two ways to amplify the extensional strain of piezoelectric materials[2]. One is to make use of bending amplification mechanism, which leads to the development of bimorph type actuators. Another way is the utilization of flextensional amplification scheme, which leads to the development of flextensional transducers widely used in underwater acoustics, moonie and cymbal actuators[3].

Recently, we presented a new kind of piezoelectric actuation structure named bimorph based double amplifier[4], because it can be considered as the combination of bending-type actuators and flextensional elements, as shown in Fig. 1. As a result the displacement of the new actuator can be more than ten times larger than the tip displacement of bimorphs, and can be used in air acoustic transducers as an actuation element. Some theoretical analyses have been given earlier[5].

In this work, the displacement of the actuator is studied in detail, with emphasis on the optimum dimension design of the cover plate of the actuator.

II. PRINCIPLE AND THEORETICAL ANALYSIS

Fig. 1. Evolution of piezoelectric actuators with larger displacement

The basic configuration of bimorph based double amplifier is shown in Fig. 2. The structure mainly consists of two parallel-mounted bimorphs with a triangle shaped cover plate as an active diaphragm fixed on the top of the bimorphs. Higher displacement is achieved by converting the tip displacement of bimorphs to the motion of the cover plate. Since one of the objective of the actuator is to work in air acoustic transducers as an actuation element, loudspeaker...
paper is chosen as cover plate material in this work because of its excellent mechanical-acoustic property and light weight.

The relationship between displacement of the cover plate (middle point) and actuator parameters is explained below: as shown in Fig. 3, letters A, B, and C represent the joint parts of the cover plate, which are called as hinge regions. Generally, the deformation of the cover plate can be considered to concentrate in these hinge regions. When a displacement of the cover plate is generated, moments will be induced in these hinge regions due to the deformation, which will balance the moments produced by the bimorphs. Therefore, the motion of the cover plate is rotation-dominated and can be treated as two rigid beams connected by a torsional spring at the middle point. When a voltage V is applied to the bimorphs, a tip displacement Δ and forces F are generated, which in turn produces a displacement ξ at the middle point of the cover plate and induces a moment M in the torsional spring due to the displacement. From the geometric consideration, it can be found that:

$$\xi = a_1 - a_0 = \sqrt{a_2^2 + L \cdot \Delta - a_0}$$  \hspace{1cm} (1)

The moment balance equation is:

$$2F \cdot (a_0 + \xi) = M = k \cdot 2 \cdot (\theta_1 - \theta_2)$$

or

$$F = (a_0 + \xi) = k(\theta_1 - \theta_2) = k \cdot 2 \xi / L$$  \hspace{1cm} (2)

Since ξ is much smaller than L, k in equation (2) is the spring constant. It has been known that:

$$F = \frac{1}{4} \cdot \frac{w}{s_1} \cdot (\Delta_0 - \Delta) = c_9 (\Delta_0 - \Delta)$$  \hspace{1cm} (3)

where w is the width and s_1 is the elastic compliance of the piezoelectric material, Δ_0 is the tip displacement under free condition, and

$$\Delta_0 = 3d_3V(h/t)^2$$  \hspace{1cm} (4)

if parallel-type bimorph is used.

The equivalent torsional spring constant k can be determined in this way: Suppose the original length of the hinge region is b, and it is also the neutral line length of the deformation area. R_A, R_B and R_C are the elastic curvature radii at hinge regions A, B, and C respectively. From Fig. 3 we can get:

$$R_{a_1} = R_{c_1} = b / \tan \theta_1, \text{ and } R_{a_2} = b / (2 \cdot \tan \theta_1), \text{ i = 0, 1 }$$

Let M_A, M_B and M_C represent the induced moments at areas A, B and C due to the displacement, then the total induced moment is:

$$M = M_A + M_B + M_C$$

where E is the Young's modulus of the cover plate material and I is area moment of inertia of cross section of the hinge region. Compare to equation (2) we get the spring constant:

$$k = \frac{2E}{b}$$  \hspace{1cm} (5)

This means that k just depends on the properties of the cover plate material. Substituting equations (3) and (5) in equation (2) yields:

$$c_9 (\Delta_0 - \Delta) \cdot a_0 = (2E/b) \cdot (2 \xi / L)$$  \hspace{1cm} (6)

From equations (1), (6) and (4), the dependence of Δ and ξ on the actuator parameters and driving voltage can be obtained, so that optimization on actuator design can be conducted. However, the analysis is rather complicated because the relationship between Δ and ξ is nonlinear according to equation (1).

Considering a simple situation, that is, if L · Δ ≪ a_0^2, we get the linear approximations for equations (1) and (6):

$$\xi = L \cdot \Delta / (2c_9)$$  \hspace{1cm} (7)

$$c_9 (\Delta_0 - \Delta) \cdot a_0 = (2E/b) \cdot (2 \xi / L)$$  \hspace{1cm} (8)
Hence \( \Delta \) and \( \xi \) can be obtained as:

\[
\Delta = \frac{\Delta_a}{1+2EI/(a^2b)} \quad (9)
\]

\[
\xi = \frac{L \cdot \Delta_a}{2a_0 + 4EI/(a^2b)} \quad (10)
\]

Equation (10) indicates that \( \xi \) will increase linearly with \( L \), but there is an optimum value for \( a_0 \) where \( \xi \) is maximum. This optimum value is:

\[
2a_0 = 4EI/(a_0^2b), \quad \text{or} \quad a_0 = \sqrt{2EI/(c_0b)} \quad (11)
\]

However, as the initial height \( a_0 \) is very small, another possibility is that the deformation of the cover plate becomes larger and cannot be neglected, that is, flextensional motion of the cover plate will occur. Hence the rigid-beam model shown in Fig. 3 is not valid. From elastic theory\[8,9\], when a triangle shaped continuous beam is under the action of compressed axial force \( P \), the displacement of middle point is:

\[
\xi = a_0 \left( \frac{\tan(\pi/2)\sqrt{P/P_e}}{(\pi/2)\sqrt{P/P_e}} - 1 \right) \quad (12)
\]

where \( P_e = \frac{EI}{L^3} \) is Euler load.

Equation (12) shows that \( \xi \) will increase linearly with \( a_0 \). Actually, this is the situation similar for cymbal actuators where the displacement is linearly related to the cavity depth of endcaps and exponentially related to the cavity diameter because of the flextensional motion of endcaps\[10\].

Therefore, if flextensional motion of the cover plate occurs above the optimum \( a_0 \) value determined from the rotation-dominated situation, the real optimum cover plate height will be in the vicinity from rotation-dominated situation to the flextension-dominated situation.

**III. EXPERIMENTS**

The bimorphs used in this work are operated in parallel configuration and made from Matsuta 1203HD (PZT type 5H) material. Their dimensions are 20.0 \( \times \) 7.7 \( \times \) 1.5mm. The loudspeaker paper with the same width and 0.56mm thickness is fixed on the top of bimorphs by super glue. The displacement of middle point and tip point are measured by using MTS 2000 Photonic Sensor under different dimensions of cover plate. The applied voltage on bimorphs is fixed at 150V (p-p value) and frequency is 1Hz.

**IV. RESULTS AND DISCUSSIONS**

Fig. 4 is the dependence of middle point displacement and tip point displacement on the height of cover plate. The length of cover plate is fixed at 55.0mm in this experiment. It clearly shows that there is an optimum value for initial cover plate height (about 1.5mm) at which middle point displacement reaches the maximum. Defining the amplification factor as the ratio of middle point displacement to the free tip displacement of bimorphs. Fig. 4 shows that the amplification factor is more than 10 in the vicinity of optimum initial cover plate height.

In order to determine the motion situation of the cover plate, the middle point displacement is also calculated by using equations (1) and (7) with measured tip point displacement and compared to experimental values in Fig. 5. It can be seen that when the cover plate height is larger than 3.0mm, the linear approximation can be used. When the cover plate height is between 1.5mm and 3.0mm, the calculated value using equation (1) is consistent with experimental results. This means it is still rotation-dominated situation but nonlinear effect must be considered. When the cover plate height is smaller than 1.5mm, the calculated value still increases but the measured value decreases, which means...
Fig. 6 shows the dependence of middle point displacement on the length of cover plate. When the cover plate height is 3.0mm, the linear approximation can be used so that the middle point displacement will linearly increase with the length of cover plate. When the cover plate height is 0.5mm, flexextensional motion becomes obvious, and the middle point displacement increased with cover plate length more rapidly than linearly. This is consistent with our theoretical analysis.

V. CONCLUSIONS

The displacement of bimorph based double amplifier actuators can be more than ten times larger than the tip displacement of bimorphs. The displacement strongly depends on the initial height of cover plate. When the cover plate height decreases, the motion of cover plate is from rotation-dominated to flexextension-dominated. And the optimum cover plate height is in the transfer region of these two kinds of situations. The displacement increases with cover plate length, but the relationship in detail is quite different according to rotation-dominated or flexextension-dominated situation.

REFERENCES

CRESCENT: A novel piezoelectric bending actuator

Sanjay Chandran, V. D. Kugel, and L. E. Cross
Intercollege Materials Research Laboratory, The Pennsylvania State University, University Park, PA 16802

ABSTRACT

Piezoelectric actuators have significant potential for use in smart systems like vibration suppression and acoustic noise canceling devices. In this work, a novel piezoelectric bending actuator CRESCENT was developed. CRESCENT is a stress-biased ceramic-metal composite actuator. The technology involves the use of the difference in thermal contraction between the ceramic and the metal plates bonded together at a high temperature by a polymeric agent to produce a stress-biased curved structure. An extensive experimental investigation of this device in the cantilever configuration was carried out. The tip displacement, blocking force and electrical admittance and were chosen to characterize the performance of the actuator under quasistatic conditions. The device fabricated at optimum temperature exhibits large tip displacement and blocking force and possesses superior electromechanical characteristics to conventional unimorph actuators.

Keywords: piezoelectric actuators, unimorph, bimorph, CRESCENT

1. INTRODUCTION

Piezoelectric ceramic materials are being increasingly investigated for use as solid-state actuators in applications requiring large displacements (>10 μm) such as loudspeakers, pumps, vibration suppression and acoustic noise canceling devices. Since the direct extensional strain in piezoelectric ceramics is quite small, novel strain amplification mechanisms are being explored. The bending mechanism is an effective strain amplification technique and actuators utilizing this mechanism are widely used for the above mentioned applications. A classical example of such a device is the unimorph actuator which consists of the non-piezoelectric and electroded piezoelectric plates bonded together. A schematic view of the unimorph actuator with rectangular cross-section in the cantilever configuration is shown in Fig. 1a. The flexural displacement \( \eta \) under applied voltage \( U \) (Fig. 1b) is caused by the piezoelectric effect in the direction perpendicular to the polar Z axis (piezoelectric \( d_{31} \) coefficient).

![Diagram](image-url)

Fig. 1. (a) A schematic view of the unimorph actuator in cantilever configuration. \( P_s \) denotes the vector of the spontaneous polarization. (b) Flexural displacement \( \eta \) of the unimorph in the ZX plane under the applied voltage.
Recently, a novel technique of using pre-stress produced during the device fabrication was introduced. The piezoelectric actuator made using this technique was named RAINBOW (Reduced And Internally Biased Oxide Wafer). RAINBOW is made by subjecting the ceramic to a selective high temperature reduction with graphite in an oxidizing atmosphere resulting in a reduced electrically conductive layer and an unreduced piezoelectric layer. The resulting stress-biased monolithic unimorph has a dome or saddle-shaped structure because of difference in the thermal contraction between the reduced and unreduced layers of the ceramic plate. It was stated that this dome shape enables the actuator to generate significant axial displacement.

In this work, the authors describe a novel piezoelectric stress-biased bender called CRESCENT. This device is very promising considering that it generates fairly large displacement and blocking force at reasonable driving fields. In addition, ease of fabrication, low cost and surface mountable configuration are some of its other attractive features. Details regarding the fabrication of CRESCENT and its electromechanical characterization are given below.

2. FABRICATION OF CRESCENT

The technology of fabricating CRESCENT involves the use of the difference in thermal contraction between the piezoelectric ceramic and metal plates bonded together by a polymeric agent to produce a stress-biased curved structure. The electroded piezoelectric plate is cemented to a metal plate by a thin layer of high temperature curing epoxy by subjecting it to high temperature (200-400°C) for a fixed period of time (typically 30 minutes - 6 hours) so that the epoxy cures and hardens. Then the structure is rapidly cooled (typically air cooled) to room temperature to achieve the differential thermal contraction. This process produces a dome-shaped structure with significant internal stresses. After the device is fabricated it is poled. A schematic view of the CRESCENT is given in Fig. 2.

Fig. 2. Schematic view of CRESCENT.

3. EXPERIMENTAL PROCEDURE

In all experiments, soft piezoelectric ceramics PKL550 (Piezo Kinetics, Inc.) and stainless steel SS302 were used. This category of piezoelectric ceramics is analogous to soft piezoelectric ceramics PZT5H. Stainless steel SS302 was used because of its very high Young's modulus. Two epoxies with different curing temperatures were used: one of the epoxies was cured at 350 °C for 6 hours and the other was cured at 350 °C for 45 minutes.

Each piezoelectric plate was rectangular in cross-section and had the following dimensions: total length $L = 30$ mm, width $w = 11$ mm and thickness $t = 1.09$ mm (Fig. 1). The SS 302 plates had dimensions: total length $L = 30$ mm, width $w = 11$ mm and thickness $t = 0.37$ mm (Fig. 1). Since the Curie temperature (~200 °C) was lower than the device fabrication temperature, the actuators were poled after fabrication. The radius of the curvature of both the CRESCENT actuators before poling was about 0.4 m and after poling it increased to 0.8-0.9 m. For comparison, standard $d_{31}$ unimorph actuator of identical dimensions was fabricated. In case of unimorph, the piezoelectric ceramic plate was first poled along its thickness and then bonded to the SS302 plate using commercially available J-B Weld epoxy (J-B Weld Company) at
room temperature for 24 hours. Hence the unimorph has a greatly reduced level of internal stress compared to the CRESCENT actuators.

The piezoelectric $d_{33}$ coefficients of the CRESCENT and unimorph actuators after poling were measured using the piezoelectric $d_{33}$ meter ZJ-2 (Institute of Acoustics, Academia Sinica). To characterize the CRESCENT and unimorph actuators, their electromechanical response as a function of the applied electric field well below the fundamental frequency of bending vibrations (quasistatic conditions) was investigated. Under quasistatic conditions, the following parameters of these devices in the cantilever configuration (Fig. 1) were measured as a function of amplitude of the electric field: (i) displacement $\eta$ of the free end of the cantilever, (ii) blocking force $F_{bl}$ (when $\eta=0$), and (iii) electrical admittance $Y$. The vibrating length of the cantilever was $l = 26$ mm.

A block diagram of the experimental set-up is shown in Fig. 3. The tip displacement of piezoelectric cantilevers was measured by a photonic sensor MTI2000 (MTI Instruments). The actuator (in Fig. 3, bimorph is shown as an example) was mounted on a XYZ micropositioner (Ealing Electro-Optics, Inc.). To measure the blocking force, a special metal head of a load cell ELF-TC500 (Entran Devices, Inc.) was glued by Super Glue to the vibration end of the actuator. Power supply PS-15 (Entran Devices, Inc.) was used to drive the load cell. The electrical admittance was measured by means of a small (a few ohms) resistor $R$ connected in series with the actuator. The lock-in amplifiers (SR830 DSP, Stanford Research Systems, Inc.) used to measure the signals corresponding to the tip displacement and admittance were synchronized with the output voltage of the power amplifier (790 Series, PCB Piezotronics, Inc. or PA-250H, Julie Research Laboratories, Inc.). The input AC signal to the power amplifier was provided by a generator DS345 (Stanford Research Systems, Inc.). The developed experimental set-up can be used over a wide frequency range (from DC to several kHz). The maximum driving voltage this setup can handle is 300 Volt RMS.

![Block diagram of the experimental set-up](image)

**Fig. 3.** Block diagram of the experimental set-up adopted to measure electromechanical properties of piezoelectric actuators. $\mathbf{P_s}$ denotes the vector of spontaneous polarization.

Electromechanical characteristics were measured in the quasistatic regime at room temperature. The measurement frequency 100 Hz was at least ten times smaller than the fundamental resonant frequency.

### 4. EXPERIMENTAL RESULTS

The values of the piezoelectric $d_{33}$ coefficient of the CRESCENT and unimorph actuators are given in Table 1. It is evident that the $d_{33}$ coefficient decreases with increasing device fabrication temperature. Thus the CRESCENT actuator fabricated at 350°C has the lowest value whereas the unimorph actuator has the highest value.

<table>
<thead>
<tr>
<th>TYPE OF PIEZOELECTRIC ACTUATOR</th>
<th>piezoelectric $d_{33}$ coefficient</th>
</tr>
</thead>
<tbody>
<tr>
<td>CRESCENT (fabricated at 250°C)</td>
<td>518</td>
</tr>
<tr>
<td>CRESCENT (fabricated at 350°C)</td>
<td>512</td>
</tr>
<tr>
<td>Unimorph</td>
<td>570</td>
</tr>
</tbody>
</table>
Fig. 4 and Fig. 5 show the variation of quasistatic tip displacement, blocking force and admittance of the CRESCENT actuator fabricated at 250°C with increasing electric field up to about 2 kV/cm. Fig. 6 and Fig. 7 show the variation of these parameters with electric field for the CRESCENT actuator fabricated at 350°C. The variation of these parameters with electric field for the unimorph actuator is depicted in Fig. 8 and Fig. 9. As seen from the figures, for all the three actuators under consideration, the quasistatic tip displacement and blocking force vary linearly with electric field for low fields but the variation becomes increasingly non-linear at higher electric fields. The slope of tip displacement and blocking force vs. electric field monotonically increases with electric field at higher levels of field.

**Fig. 4.** Variation of tip displacement and blocking force of CRESCENT fabricated at 250°C with electric field.

**Fig. 5.** Variation of relative admittance of CRESCENT fabricated at 250°C with electric field.
g. 6. Variation of tip displacement and blocking force of CRESCENT fabricated at 350°C with electric field.

g. 7. Variation of relative admittance of CRESCENT fabricated at 350°C with electric field.
Fig. 8. Variation of tip displacement and blocking force of unimorph with electric field.

Fig. 9. Variation of relative admittance of unimorph with electric field.

5. DISCUSSION AND SUMMARY

The experimental results show that the CRESCENT actuator fabricated at 250°C exhibits a larger tip displacement but a smaller blocking force and electrical admittance than the unimorph actuator. However, the CRESCENT actuator fabricated at 350°C shows a smaller tip displacement, blocking force and electrical admittance than the unimorph. It is indeed useful to compare the various electromechanical parameters of the three actuators and arrive at an overall figure of
merit which can be used to evaluate their electromechanical performance relative to one another. This comparison can be done using the analysis given below. This analysis can be applied only to actuators consisting of plates with rectangular cross-section.

The actuators were studied in the cantilever configuration in which the mechanical load is usually applied to the vibrating end of the cantilever. Therefore the most important characteristics under quasistatic conditions are the free displacement $\eta$ (Fig. 1) of the vibrating end and the blocking force $F_m$ when $\eta=0$. For piezoelectric unimorph actuators the analysis can be applied only to actuators consisting of plates with rectangular cross-section.

$$\eta = \frac{3}{2} \frac{d_{31}^2}{t} k_x E,$$
$$F_m = \frac{3}{8} \frac{d_{31}^2}{s_{11}^m} \frac{w t}{l} k_x E,$$

where $d_{31}$ and $s_{11}^m$ are the piezoelectric coefficient of ceramics, and the mechanical compliance of ceramics in the direction X under the constant electric field $E$ (Fig. 1) respectively, $k_x$ and $k_m$ are displacement and blocking force coefficients respectively, and $E = U/t_c$.

For unimorph cantilevers (i.e. without internal mechanical stress) $k_x$ and $k_m$ can be expressed as

$$k_x = \frac{2xy(1+x)}{1+4xy+6x^2y+4x^2y+x^2y^2}, \quad x = \frac{t_m}{t_c}, \quad y = Y_m,$$
$$k_m = \frac{2xy}{1+xy},$$

where $Y_m$ is the Young’s modulus of the metal. As given in equation (2) factors $k_x$ and $k_m$ depend on the ratio of thicknesses $x$ and ratio of Young’s moduli $y$ of metal and piezoelectric plates.

For CRESCENT actuators, internal mechanical stress produced during device fabrication and poling, changes the electromechanical properties of ceramics. Therefore equation (1) for these actuators can be written as

$$\eta = \frac{3}{2} \frac{d_{31}^2}{t} k_x k_m E,$$
$$F_m = \frac{3}{8} \frac{d_{31}^2}{s_{11}^m} \frac{w t}{l} k_x k_m E,$$

where $k_p$ is equal to relative change in $d_{31}$ and $k_m$ is equal to the change in $d_{31}/s_{11}^m$ as a result of the internal stress bias. $k_x$ is not very sensitive to change in $s_{11}^m$. Hence $k_x$ can be calculated from the ratio of the tip displacements of the CRESCENT and unimorph actuators using equation (1) and equation (3). However, $k_m$ being a strong function of $s_{11}^m$, the value of $k_m$ in equation (3) should be first calculated using the value of $s_{11}^m$ under mechanical stress in equation (2). Then $k_m$ can be determined from the ratio of blocking forces and ratio of $k_m$ for unimorph and CRESCENT actuators using equation (1) and equation (3).

Another important quasistatic electromechanical characteristic of the actuators is their electrical admittance $Y$:

$$Y = j\omega \frac{\varepsilon_{11}'E_p}{t_c},$$

where $\omega$ is the angular frequency $= 2\pi v$, $\varepsilon_{11}'$ is the component of the tensor of the dielectric permittivity of ceramics and coefficient $k_T$ depends on the electromechanical coupling coefficient and a change in the dielectric permittivity due to device fabrication.

An overall figure of merit representing the ratio of the mechanical work to the input electrical energy can be expressed as

$$\text{overall figure of merit} = \frac{\eta F_m}{Y}.$$
quasistatic characteristics should be measured for the same amplitude and frequency of the applied electric field since electromechanical properties of piezoelectric ceramics depend on the amplitude and frequency of the electric field.\(^7\)

Calculated relative values with respect to the unimorph actuator representing electromechanical characteristics of the actuators studied are given in Table 2. These values were obtained at low applied electric field (less than 20 V/cm). As seen from Table 2, the CRESCENT actuator fabricated at 250°C has the highest figure of merit.

Table 2. Figures of merit of bending-mode piezoelectric actuators in the cantilever configuration.

<table>
<thead>
<tr>
<th>Type of Piezoelectric actuator</th>
<th>Tip displacement (relative)</th>
<th>Blocking force (relative)</th>
<th>Admittance (relative)</th>
<th>Overall figure of merit</th>
</tr>
</thead>
<tbody>
<tr>
<td>Unimorph</td>
<td>1</td>
<td>1</td>
<td>1</td>
<td>1</td>
</tr>
<tr>
<td>CRESCENT (fabricated at 250°C)</td>
<td>1.09</td>
<td>0.97</td>
<td>0.91</td>
<td>1.16</td>
</tr>
<tr>
<td>CRESCENT (fabricated at 350°C)</td>
<td>0.38</td>
<td>0.85</td>
<td>0.90</td>
<td>0.83</td>
</tr>
</tbody>
</table>

Based on the relative values of tip displacement and blocking force and equations (1) and (3) the values of \(k_p\) and \(k_{pm}\) under quasistatic conditions and low applied electric field can be calculated. The calculated values of these coefficients are given in Table 3.

Table 3. \(k_p\) and \(k_{pm}\) values for CRESCENT actuators

<table>
<thead>
<tr>
<th>Type of Piezoelectric actuator</th>
<th>(k_p)</th>
<th>(k_{pm})</th>
</tr>
</thead>
<tbody>
<tr>
<td>CRESCENT (fabricated at 250°C)</td>
<td>1.09</td>
<td>0.88</td>
</tr>
<tr>
<td>CRESCENT (fabricated at 350°C)</td>
<td>0.38</td>
<td>0.82</td>
</tr>
</tbody>
</table>

From Table 2, it is clear that the CRESCENT actuator fabricated at 250°C has a higher tip displacement and lower admittance than \(d_{31}\) unimorph fabricated from the same materials. This implies that the average piezoelectric \(d_{31}\) coefficient of the piezoelectric ceramic plate poled under a certain level of mechanical bending stress is higher and the corresponding dielectric permittivity \(\varepsilon_{33}\) lower than that of the starting material. This is very surprising because experimental results show that longitudinal stress decreases piezoelectric \(d_{31}\) coefficient. Also, it is interesting to note that for a piezoceramic plate poled under a certain level of mechanical stress, the piezoelectric \(d_{31}\) coefficient is lower but the piezoelectric \(d_{31}\) coefficient is higher than the starting material. This enhanced electromechanical response can be attributed to specific domain structures which are formed during poling. A very important fact supporting this hypothesis is that there was a significant increase in the radius of curvature of the device after poling. For the CRESCENT actuator fabricated at 250°C (above the optimum temperature), excessive residual mechanical stress may decrease the piezoelectric \(d_{31}\) coefficient. This explains the inferior electromechanical characteristics exhibited by the actuator. The blocking force of both CRESCENT actuators is less than that of unimorph. This can be due to the increase in \(s_n^8\) due to mechanical stress during poling of the device.

It is evident from the results that the electromechanical and dielectric properties of the CRESCENT actuators are dependent on the level of the applied electric field. At high electric fields, they exhibit a non-linear variation. This can be attributed to the non-linear behavior of the soft PZT ceramics.\(^7\) This non-linear behavior of ceramics has an extrinsic nature i.e. it is caused by domain wall and phase boundary motion.

Another important yardstick to evaluate the reliability of an actuator is the electric field at which mechanical failure occurs at resonance. Electromechanical resonance induces a very high level of mechanical vibrations which causes mechanical fracture. For unimorph and bimorph actuators, the fracture occurs at 30-50 MPa in the region where actuators were clamped since this region is subjected to the highest level of stress.\(^8\) CRESCENT actuators have metal plates, and hence even if mechanical failure of ceramics occurs the actuators do not fracture since metals like stainless steel have much
higher fracture toughness than ceramics. Hence they can be expected to have a higher reliability than bimorphs. Work to study the reliability aspects of the CRESCENT actuators is currently in progress.

In summary, the electromechanical characteristics of CRESCENT and unimorph actuators in cantilever configuration have been investigated and compared. The tip displacement, blocking force, and electrical admittance under quasistatic conditions were chosen to evaluate electromechanical performance. The experimental results show that the CRESCENT actuator fabricated at close to optimum temperature exhibits superior characteristics as compared to conventional unimorph. This also shows that though in most cases stress has an adverse effect on the performance of piezoelectric devices, it can enhance performance if prudently designed and properly directed.

ACKNOWLEDGMENTS

This work was supported by the Office of Naval Research under the contract N00014-94-1-1140.

REFERENCES

APPENDIX 5
Dynamic Characteristics of Rainbow Ceramics

Catherine Elissalde and Leslie Eric Cross

Materials Research Laboratory, The Pennsylvania State University, University Park, Pennsylvania 16802

The piezoelectric resonance behavior of end-clamped bimorph structures made from the rainbow monolithic ceramic have been evaluated. Thickness, width, width flexure, and length bending modes have been identified and measured. Using a very crude model in which the cermet component of the rainbow is assumed to have the same density \times\text{elasticity product as the bulk, surprisingly good agreement is obtained between observed and calculated frequency behavior. By appropriate processing it is possible to delaminate the bimorph, and work is now in progress to measure the properties of the two separate phases so that a proper composite resonator model may be developed.

I. Introduction

Piezoelectric and electrostrictive materials are used in a wide range of applications. Especially smart materials have aroused increasing interest. To improve the performance of piezoelectric materials (higher strain or higher displacement), new structures such as flextensional structures or "moonie" are developed. Recently a new type of monolithic ceramic bender, known as "rainbow" (reduced and internally biased oxide wafer), was developed by Haertling. This material presents a number of advantages and in particular the possibility to generate very high displacements. The piezoelectric material response depends not only on its piezoelectric properties but also on elastic and dielectric parameters. Elastic bodies show resonances, and the resonance method is convenient for evaluating such properties. In this way, the purpose of this work is to study the resonant behavior of rainbow materials. A good knowledge of the frequency characteristics is indeed, essential for many types of applications.

II. Experimental Procedure

(1) Sample Preparation

Rainbow piezoelectric PZT (lead zirconate titanate) materials were prepared using a conventional mixed oxide process. During the processing, an additional important step for the rainbow is the high-temperature chemical reduction of one surface of the wafer. The thickness of the reduced layer is 1/3 to 1/2 of the sample thickness. This single-sided reduction introduces a stress in the ceramic and the rainbow becomes a dome-shaped wafer.

The rainbow actuators were supplied by Aura Ceramics, Inc. The dimensions of the initial wafer were the following: 50.3 mm in diameter and 0.76 or 0.38 mm in thickness. A motional diamond saw was used to generate the rainbow cantilever for our experiments. One end of the sample was then clamped in a plastic support using 5 Minute Epoxy glue, and the other end was free.

(2) Frequency Measurements

Frequency measurements were performed using an HP4194 A impedance/gain-phase analyzer which covers a frequency range of 100 Hz to 40 MHz. A preliminary calibration was conducted (open and short circuit). On account of the initial bent shape of the rainbow and thus because of the high sensitivity of the sample measurement, two different cells were used. Both were connected to a test fixture with the four terminals attached directly to the terminals of the analyzer (BNC connectors). The so-called "soft cell" (SC) consisted of an HP16047 C test fixture. The so-called "hard cell" HC was made up of a clip mounted on an HP16047 A test fixture. For each cell, the contact was near the clamped end of the rainbow cantilever.

The reduced layer on the concave side of the rainbow served as one of the electrodes but it was also possible to electrode it only

---

G. H. Haertling—contributing editor

Manuscript No. 192775. Received March 5, 1995; approved June 5, 1995.

Member, American Ceramic Society.

---

Fig. 1. Impedance spectrum of rainbow cantilever, $L = 25.3$ mm, $w = 5$ mm, $t = 0.76$ mm.

---

Frequency (Hz)
Communications of the American Ceramic Society

Table I. Measured Resonant Frequency of Rainbow Cantilever

<table>
<thead>
<tr>
<th>Resonant frequency (Hz)</th>
<th>w = 5 mm</th>
<th>x = 2.5 mm</th>
<th>w = 1.5 mm</th>
</tr>
</thead>
<tbody>
<tr>
<td>f = 0.76 mm</td>
<td>93.732</td>
<td>274.861</td>
<td>678.735</td>
</tr>
<tr>
<td>f = 0.38 mm</td>
<td>52.046</td>
<td>158.197</td>
<td>426.264</td>
</tr>
</tbody>
</table>

structure of the rainbow, there are two materials involved, the unreduced PZT ceramic and the lead-based cermet of the reduced region. As a first approximation, assuming that the cermet behaves elastically like the PZT, the value obtained from experimental results is about $C'_{\text{er}} = 14.4 \times 10^6$ N/m$^2$ ($C'_{\text{er}} = 2.6$ MHz). Such a value is effectively the same as the value for common piezoelectric ceramics.

The frequency of the resonant mode observed close to $3 \times 10^5$ Hz in Fig. 1 depends only on the width of the sample and thus corresponds to the lateral extensional mode of the cantilever. Again, assuming the cermet behaves like PZT we could expect the frequency to be given by

$$f_x = 0.5(1/\omega)\sqrt{C_{\text{er}}'p}$$

where $\omega$ is the width of the sample, $p$ the density, and $C_{\text{er}}'$ the effective elastic constant measured at constant field. The mechanical conditions are $S = 0$, $T = 0$, and $C_{\text{er}}' = 1/S = 1 - \sigma^2$.

Figures 2 and 3 represent the measured frequency versus width, for width 0.76 and 0.38 mm thick rainbows. The plots clearly verify the $1/\omega$ dependence and from the slope a value of $S = 1.4 \times 10^{-13}$ m/N is obtained assuming that the value of the Poisson ratio, $\sigma$, is that of PZT, i.e., $\sigma = 0.36$. Such a $S$ value is in the range expected for softer PZT compositions.

For the resonance observed near $9 \times 10^4$ Hz (Fig. 1) the most probable origin is a width bending mode as the frequency scales with both width and thickness, but not with length of the rainbow. Table I lists the measured resonance frequencies as a function of these two controlling dimensions.

(2) Bending Mode

In this study we have focused our attention on the lowest frequency resonant mode, namely the bending mode. Figure 4 shows the measured bending resonant frequency versus the ratio $\omega L^2$ of the cantilever. By changing the length, the frequencies of the other resonant modes remain constant. Again, using the assumption of a homogeneous ceramic, the resonant frequency of a cantilever (one end clamped and the other free) can be expressed as
temperature were performed. The temperature range was from room temperature to 340 K. As an example, the temperature dependence of the frequency of the bending mode is shown in Fig. 7. For all the samples, a decrease of frequency is observed when the temperature increases. The corresponding variation of the resonant frequency can be well fitted using a law-type such as \( \ln f = A + B(1/T) \) (A and B are constants). Using Eq. (3) it is also possible to determine the temperature dependence of the elastic compliance. In each case, \( S''_t \) increases linearly with increasing temperature. The curve fit is a straight line (Fig. 8).

It is perhaps surprising that the assumption of a homogeneous ceramic in describing the elastic and density properties works very well and is probably adequate for many engineering purposes. Given the fact that oxygen is lost from the ceramic during the reduction process and that the resulting cermet puts the ceramic under compression, one would expect a higher density in this component. Possibly this change is partially compensated by a reduction in stiffness due to the metallic phase. We note that in some cases it is possible to cause a delamination between the ceramic and the cermet phases, so that with the possibility of explaining the elastic and density behavior of the two separate phases, it will be possible to model more exactly the precise behavior of this composite resonator.

### IV. Summary

The frequency behavior of rainbow cantilevers has been studied in a large frequency range (100 Hz to 4 MHz). Different modes of resonance have been observed. The stronger corresponds to the thickness mode and appears at high frequency (\( f > 2 \text{ MHz} \)). At low frequency, the bending mode can be characterized by an impedance-frequency peak with a value of
$Q$ close to 100. The corresponding resonant frequency increases when the length of the cantilever decreases and is also dependent on temperature. The behavior has been modeled on the simple assumption that the rainbow cermet has elastic and density properties similar to those of the bulk ceramic and there is surprisingly good agreement between measured and predicted properties. The temperature dependence of the resonance suggests that the $S_{T1}$ has a linear variation with temperature.

Using separated cermet and ceramic elements, measurements are now in progress to characterize exactly the separate phases so that a more precise composite resonator model can be developed.

References

APPENDIX 6
Characterization of the linear and non-linear dynamic performance of RAINBOW actuator

Sanjay Chandran, V. D. Kugel and L. E. Cross
Intercollege Materials Research Laboratory, The Pennsylvania State University
University Park, PA 16802, USA

Abstract - In the last few years the technology of using piezoelectric actuators for applications requiring large displacements such as loudspeakers and noise-canceling devices has undergone significant development. The RAINBOW (Reduced and Internally Biased Oxide Wafer) is a novel high displacement actuator and knowledge of its dynamic response is indeed essential for these applications. In an attempt to characterize the RAINBOW, measurements were made of important lumped mechanical and electrical parameters. Cantilevers of different dimensions were cut from RAINBOW discs. The data include measurements of the mechanical displacement (under both quasistatic conditions and electromechanical resonance), and its hysteresis, mechanical quality factor and the electrical impedance of RAINBOW cantilevers. These measurements demonstrate the macroscopic effects of the sinusoidal applied electric field and indicate significant non-linearities in the RAINBOW device.

I. INTRODUCTION

For many years, piezoelectric and electrostrictive ceramic materials are being increasingly investigated for use as solid-state actuators for small displacements (<10 μm) and precise mechanical movement devices. However, many applications like loudspeakers and noise-canceling devices require actuators producing larger displacements (>1 mm). The direct extensional strain in most active ceramic materials is quite small (<1%) and hence novel techniques of strain amplification are required to satisfy these demands for high displacement actuation. Recently, a new type of monolithic ceramic bender called RAINBOW was developed by Haertling [1]. Key features of the RAINBOW include quick processing, ease of fabrication, surface mountable configurations and above all, its ability to produce very high displacements at reasonable driving fields. A good knowledge of the dynamic behavior of the device is indeed essential for many applications. The purpose of this work is to study the dynamic characteristics of the RAINBOW.

II. EXPERIMENTAL METHODS

A. Sample Preparation

As it is known, piezoelectric PLZT ceramics are prepared by a conventional mixed oxide process. During the processing, an additional important step for making the RAINBOW is the high temperature chemical reduction process which involves the local reduction of one surface of the ceramic thereby achieving an anisotropic, stress-biased, dome-shaped wafer with significant internal tensile and compressive stresses which act to increase the overall strength of the material [1]. The thickness of the electrically conducting reduced layer is about 1/3 of the sample thickness. The RAINBOW ceramics used in this work were supplied by Aura Ceramics Inc. The original wafers were 50.8 mm in diameter and 0.38 mm in thickness. Cantilevers of various dimensions were cut from the original wafers using a Motional Diamond Saw. One end of the cantilever was clamped in a plastic support using Superglue, the other end was free.

B. Measurements

Frequency measurements under low electric field were performed using a HP4194A Impedance/Gain Phase analyzer in the frequency range 100 Hz - 4 MHz. To investigate the response of the RAINBOW cantilevers under varying frequency and amplitude of the driving field (RMS), sinusoidal signals varying in amplitude and frequency were applied with the signal generator. The current flowing through the sample was determined from the voltage drop across a small resistance which was measured on a lock-in amplifier. Hence impedance of the sample was determined. The tip displacement of the RAINBOW cantilever was measured by an optical fiber sensor. The amplitude of the tip displacement (RMS) was measured by a lock-in amplifier while the sinusoidal signal corresponding to the tip displacement was monitored on an oscilloscope. Complete description of the measurement setup used can be found in [2].
III. RESULTS

A. Resonance modes

The frequency spectrum of the electrical impedance of the RAINBOW cantilever with one end clamped shows four resonance modes in the frequency range 100 Hz - 4 MHz [3]. The lowest resonant mode is the bending mode. All the results discussed subsequently are pertinent to this bending mode. The resonant frequency of the bending mode can be expressed as [4]:

\[ f_b = \frac{m^2}{2\pi^2} \cdot \frac{aL^2}{\rho \times s_i^2} \]  

where \( m \) = eigen value of the resonant mode = 1.875
\( l \) = length of the sample
\( \rho \) = density of the sample
\( s_i^2 \) = elastic compliance of the sample

It is assumed in (1) that the elastic compliance values of both reduced and unreduced layers are equal. Work by Elissalde [3]; confirmed this aspect.

B. Determination of \( Q \)

The sharpness of the peak in the impedance frequency spectrum in the neighborhood of resonance is determined by losses in the material. Near the resonance the dominant factor may be related to the mechanical losses. Hence the mechanical quality factor \( Q \) is an important parameter to be determined in the characterization of resonating systems: it gives an idea of the damping, a narrow peak surrounding \( f_b \) suggests light damping and vice-versa. \( Q \) can be calculated from the minimum impedance \( Z_m \) at resonance as follows [4]:

\[ Q = \frac{1}{\sqrt{4\pi (C_f - C_j) \cdot (f_a \cdot f_j) \cdot |Z_m|}} \]  

where \( C_f - C_j \) = capacitance measured at frequency well below fundamental resonance
\( f_a \cdot f_j \) = frequencies corresponding to the maximum values of resistance \( R \) and conductance \( G \) respectively.

As evident from Fig. 1, \( Q \) (calculated from (2)) decreases almost linearly with the driving field over the 2 kV/cm range. This behavior can be attributed to the increase in the losses in the material.

C. Variation of \( f_b \) with driving voltage

Fig. 2 shows that the resonant frequency \( f_b \) decreases with increasing electric field. This can be attributed to change in the elastic properties of the material due to the increasing driving voltage, increasing losses and a change in the effective length of the curved RAINBOW cantilever. The curvature of the RAINBOW slightly decreases with increasing magnitude of the driving voltage resulting in a longer effective length. \( s_i^2 \) of the RAINBOW also may increase as a result of temperature increase due to Joule heating [3].

D. Variation of tip displacement \( w \) at resonant frequency with driving field

Fig. 3 shows the variation of the tip displacement of the RAINBOW cantilever with increasing driving field at resonant frequency. The tip displacement \( w \) increases in a non-linear manner with increasing field. A 5% hysteresis is observed and the slope of the curve shows a monotonic decrease over the 2 kV/cm field range. It can be observed that \( w \) is as high as 0.6 mm at a driving field slightly less than 2 kV/cm. This confirms the high displacement actuation of the RAINBOW.

E. Variation of tip displacement \( w^* \) under quasistatic conditions with driving field

As seen from Fig. 4, the quasistatic tip displacement \( w^* \) (at 10 Hz) increases non-linearly with electric field. The curve shows hysteresis and the slope of the curve monotonically decreases with increasing field.

F. Signal distortions

While traversing the frequency range 100 Hz to 1 kHz, small output signal distortions were always detectable. Significant distortions in the tip displacement signal \( w \) for a sinusoidal input were observed for frequencies close to the frequencies

\[ f_b = \frac{f_a}{j} \]  

where \( j = 2, 3, 4 \) etc.

This may be due to the presence of higher harmonic components which may grow and cause significant distortions.

IV. DISCUSSION

The results discussed in the previous sections suggest the presence of significant non-linearities in the RAINBOW device. These non-linearities include a 5% hysteresis and a change in the slope of tip displacement versus driving field curve. To understand the origin of distortions in the tip displacement signal \( w \) of the RAINBOW cantilevers it is
Fig. 1. Mechanical Quality factor as a function of applied field

Fig. 2. Resonant frequency as a function of applied field

Fig. 3. Tip displacement at resonance as a function of applied field

Fig. 4. Quasistatic tip displacement and F as a function of applied field
useful to analyze the constitutive equations of bending vibrations [5]. Externally applied voltage produces a bending moment in bimorph and unimorph cantilevers [5]. This bending moment $M$ is related as

$$M = d_{33} / s_{11} E$$

(4)

There is a linear relation between $d_{33}$ and $w^*$ and between $1/s_{11} E$ and $f^*$ [4]. Therefore the product $F^*$ given by

$$F^* = (f_{33}^*)^2 \cdot (w^*)$$

(5)

gives an idea of the component of the bending moment produced by external electric field. If $F^*$ is a non-linear function of the sinusoidal driving voltage, the corresponding excited mechanical vibrations can be expected to have higher harmonic components. The variation of $F^*$ as a function of the driving field is shown in Fig. 4. It is evident from Fig. 4 that $F^*$ varies non-linearly with the driving field especially at high fields. Consequently, at such high fields, the higher harmonic components in displacement $w$ may grow and significant distortions may result.

V. SUMMARY

The dynamic performance of the RAINBOW under an electric field varying in amplitude and frequency has been studied. As discussed in the earlier sections, the measurements of various parameters like mechanical displacement (under both quasistatic conditions and resonance), electrical impedance, mechanical quality factor and resonant frequency indicate significant non-linearities in the behavior of the RAINBOW.

VI. REFERENCES


APPENDIX 7
Reduced and internally biased oxide wafer (RAINBOW) actuators are fabricated by a controlled reduction of Pb(Zr,Ti)O₃-based piezoelectric material. The reduction process results in a conductive layer composed of interconnected metallic lead phase and refractory oxides (ZrTiO₃, ZrO₂, La₂O₃, etc.). The nature of the reduction is discovered to be the result of a complex volume change leading to a nanoscale interconnected metallic structure. The distribution of phases within the cermet vary within the thickness of the wafer. Within the piezoelectric ceramic phase, the reduction process modifies the grain-boundary structure to give two distinct types of fracture: transgranular and intergranular. The complex microstructures of the RAINBOW actuator materials are discussed in relation to their dielectric and piezoelectric properties.

I. Introduction

Recently, there has been a continuous effort to improve the performance of piezoelectric materials for electromagnetic actuator applications. Novel piezoelectric structures, such as unimorph or bimorph cantilevers and flextensional composite structures, have all been developed to produce higher strains than the basic monolithic materials. As an example, the ceramic-metal composite actuator, also called "Moonie," is able to provide large displacements and generative forces. A new type of monolithic ceramic, known as the reduced and internally biased oxide wafer (RAINBOW), is of extreme interest because it presents the advantage of a wide range of stress/strain characteristics. The RAINBOW device can be described as a monolithic structure with a piezoelectric layer (nonreduced) and a reduced cermet layer (electrically conductive): a cermet is a ceramic-metal composite material often used for mechanical applications. The controlled reduction of a stoichiometric Pb(Zr,Ti)O₃ (PZT) is achieved by placing the bottom surface of the ceramic on a carbon substrate and protecting the top surface with a ZrO₂ plate. The RAINBOW actuator is heated at high temperatures and cooled to room temperature. An internal radial stress develops during the reduction process and distorts the wafer to the domelike structure of the actuator. This transformation process gives the RAINBOW actuator unique elastodielectric properties. Our objective is to understand the microstructure of the RAINBOW material, with relation to the electrical and mechanical properties of the two phases (reduced and nonreduced).

II. Experimental Techniques

Lead lanthanum zirconate titanate (PLZT-) based RAINBOW ceramics were used (Aura Ceramics, Inc., Minneapolis, MN). The composition was given as 5.5/56/44, in terms of the respective constituent lanthanum/zirconium/titanium ions.

The electromechanical resonant behavior of the RAINBOW material was measured using an impedance/gain phase analyzer (Model HP4194A, Hewlett-Packard Co., Palo Alto, CA) in the frequency range 100 Hz-1 MHz. Dielectric measurements were conducted from room temperature up to 250°C at various frequencies using a multifrequency inductance-capacitance-resistance (LCR) meter (Model HP4274A, Hewlett-Packard). An acoustic microscope (Sontex) was used to determine longitudinal and transverse sound velocities of the samples.

Thermal expansion measurements were performed from room temperature up to 650°C to measure the thermal strains. A vertical push-rod dilatometer equipped with a high-sensitivity linear variable-differential transformer (LVDT) was used. X-ray diffractometry (XRD) analysis was conducted for phase identification using a diffractometer (Model PAD V, Scinmag, Sunnyvale, CA). Scanning and transmission electron microscopy studies. SEM and TEM, respectively, were conducted to evaluate the microstructural details of the RAINBOW structures. The SEM microscope was a field-emission model (Model 6300f, JEOL, Tokyo, Japan). and a scanning tunneling electron microscope (STEM) (Model 420, Philips Electronic Instruments, Mahwah, NJ) operated at 120 kV was used for TEM. The TEM samples were prepared in planar and cross-sectional views. The samples were polished to a thickness of 40 μm and then mounted on 3 mm copper grids with epoxy. Ion-beam thinning was performed on a dual mill (Gatan, Pleasanton, CA) at 4 kV at 12°.

III. Results and Discussion

(1) Microstructural Characterization

Figure 1(a) shows a typical XRD analysis for the reduced layer on the cermet phase. The dominant phase was identified as metallic lead: the presence of the additional oxide materials. The complex microstructures of the RAINBOW actuator materials are discussed in relation to their dielectric and piezoelectric properties.
Fig. 1. XRD patterns for (a) the reduced layer and (b) a different thickness region of the cermet.
which is caused somewhat by the unreduced oxide inclusions but predominantly is associated with the interconnected porosity, which is discussed below.

Figures 2(a)–(e) show the general microstructural characteristics of the cermet phase. Figure 2(a) is an SEM micrograph that shows the pore and channel structures contained within the cermet; these pores and channels have cross-sectional diameters of ~100 nm. Figure 2(b) shows the equivalent TEM micrograph of these structures; the TEM study revealed that these channels and pores are real features distributed homogeneously throughout the cermet up to the piezoelectric interface. Additionally, the channels are percolating throughout the cermet with a 3-3 connectivity. The size of the oxide phase inclusions is ~100 nm (Fig. 2(c)). Figure 2(d) shows a Moire fringe of the phases.

Fig. 2. (a) SEM image of the cermet; (b) TEM planar view image of the cermet; (c) TEM bright-field image of the multiple oxide inclusions within the metallic lead phase; (d) Moire patterns within the cermet lead crystallites; and (e) selected-area diffraction photograph of the cermet, showing ring patterns of the phases.
Table I. Comparison between Lattice Spacings
Obtained from TEM and International Centre
for Powder Diffraction Data Files

<table>
<thead>
<tr>
<th>Material</th>
<th>TEM d-spacing, Å</th>
<th>XRD d-spacing, Å</th>
<th>ICDD Card No.</th>
</tr>
</thead>
<tbody>
<tr>
<td>ZrTiO₃</td>
<td>3.59 3.61</td>
<td>2.93 2.95</td>
<td>34-415</td>
</tr>
<tr>
<td></td>
<td>2.53 2.516</td>
<td>2.39 2.33</td>
<td></td>
</tr>
<tr>
<td></td>
<td>2.11 2.15</td>
<td>1.81 1.806</td>
<td></td>
</tr>
<tr>
<td>PbO</td>
<td>3.87 3.856</td>
<td>2.879 2.872</td>
<td>26-577</td>
</tr>
<tr>
<td></td>
<td>2.33 2.335</td>
<td>2.09 2.09</td>
<td></td>
</tr>
<tr>
<td></td>
<td>1.89 1.884</td>
<td>1.879 1.872</td>
<td></td>
</tr>
<tr>
<td>Lead</td>
<td>2.879 2.855</td>
<td>2.49 2.475</td>
<td>4-086</td>
</tr>
</tbody>
</table>

*International Centre for Powder Diffraction Data, Newtown Square, PA.

pattern contained within the metallic lead crystallites. Closer inspection of the Moire fringe contrast reveals a dislocation structure associated with lattice parameter changes and very little suggestion of lattice rotation as the source of the internal stresses. Figure 2(e) is a selected-area diffraction pattern revealing ring patterns of the oxide phase ZrTiO₃. The systematic diffraction rows noted in Fig. 2(d) are associated with the larger metallic lead crystallites. The obtained lattice spacings are compared to the International Centre for Diffraction Data (Newtown Square, PA) files for these different phases and listed in Table I.

Electron micrographs of the cross-sectional region between the cermet and the ceramic are shown in Figs. 3(a)-(d). Figure 3(a) is an SEM micrograph of the cermet/ceramic interface. A typical bright-field micrograph of the interface is illustrated in Fig. 3(b): the volume reduction of the cermet phases, along with the pores and channel formation, is shown. No special crystallographic orientation relationships have been found in this study; we believe this is the result of the textured lead-metal crystallites being first nucleated and grown from the random orientation of the piezoelectric ceramic grains. Figure 3(c) shows a diffraction pattern from the piezoelectric grain: the elongated spots in the (110) direction are typical of the ferroelectric twin structures found in PZT grains. From the microscopy, it is clear that the interface between the cermet and ceramic is relatively uniform. The roughness of the interface, being only over a few grains along the length of the cermet-ceramic, is represented schematically in Fig. 3(d). Figure 4(a) shows that there are two regimes within the perovskite oxide resulting from the reduction process; these regions are labeled I and II in the SEM micrograph of a fractured...
cermet-ceramic interface. Region I reveals transgranular fracture, whereas region II reveals intergranular and transgranular fracture but is predominantly intergranular. Also observed within the grains is the complex ferroelastic domain structure typical of a PZT ferroelectric.

The difference in the fracture behavior is believed to be the result of an initial modification of the grain boundaries in the form of lead oxide loss and reduction from the grain boundaries. The process is the first stage of the reduction process, with the second stage being the full reduction of the PZT grains into the cermet.

The microstructure features observed above account for the evolution of the cermet phase in the reduction process, as noted by Haerling. Figure 5 summarizes the general features of the reduction process: the evolution of the thickness (t) versus reduction time (t) does not follow the ideal parabolic relation \( t = D t_0 \), where \( D \) is the diffusion coefficient. The micrographs above, with the volume reduction and the channel formation, could account for this departure. The chemical reduction processes also are listed.4

(2) Physical Properties

In our previous study of the resonance behavior of RAINBOW materials, we modeled the impedance data assuming similar elastic behavior between the ceramic and cermet phases. Considering the new microstructural evidence, we reconsidered this basic assumption.

The determination of the fundamental thickness mode allows precise measurement of the elastic stiffness coefficient, \( C_{33}^{0} \), using the following relationship:

\[
2f = \left( \frac{C_{33}^{0}}{\rho} \right)^{1/2}
\]

where \( f \) is the thickness of the sample, \( f \) the parallel resonance frequency, and \( \rho \) the density. Geometrically determined densities reveal similar values for sections of the cermet phase and ceramic phase: \( \rho = 7.5 \times 10^3 \) kg m\(^{-3}\). This similarity is a direct consequence of the porosity contained within the cermet phase. The obtained \( C_{33}^{0} \) value for the nonreduced material is \( 15.5 \times 10^9 \) N m\(^{-2}\), within the typical range expected for soft PZT.4

For the RAINBOW material, this value is only \( 14.4 \times 10^9 \) N m\(^{-2}\). By varying the relative thickness of the piezoelectric and modeling the composite elastic stiffness with the series mixing:

\[
\frac{1}{C_{33}^{0}_{\text{m}}} = \frac{t_{\text{m}}}{C_{33}^{0}_{\text{m}}} + \frac{t_{\text{p}}}{C_{33}^{0}_{\text{p}}} + \frac{t_{\text{cermet}}}{C_{33}^{0}_{\text{cermet}}}
\]

where \( t_{\text{m}} \) is the piezoelectric thickness, \( t_{\text{cermet}} \) the total thickness is simply the sum of the components: i.e., \( t_{\text{cermet}} = t_{\text{p}} + t_{\text{cermet}} \). \( C_{33}^{0}_{\text{m}} \) the piezoelectric elastic stiffness, and \( C_{33}^{0}_{\text{cermet}} \) the cermet elastic stiffness. Then, knowing \( C_{33}^{0}_{\text{m}} \) and varying \( t_{\text{cermet}} \), deduction of \( C_{33}^{0}_{\text{cermet}} \) is possible. An average value of 10.04 \( \times 10^9 \) N m\(^{-2}\) was obtained. Further verification of the elastic coefficients was obtained from acoustic microscopy. Acoustic microscopy allows direct determination of the elastic coefficients via the velocity of the acoustic wave transmitted through the material. A comparison of the two techniques is presented in Table II, where good agreement is shown between these measurements. A reduction in stiffness within the cermet structure is noted: the metallic lead and interconnected porous structure account for this result.

Additionally, the lateral extensional mode of the RAINBOW cantilever can be determined. Figure 6 shows the resonant frequency versus inverse width (1 w) of the nonreduced piezoelectric cantilever with both ends free. Under these boundary conditions, the resonant frequency can be expressed as
Effect of reduction time on the reduced layer thickness.

Table II. Elastic Stiffness Constant \( (C_{eff}^E) \) Values Obtained from Resonance Method and Acoustic Microscopy

<table>
<thead>
<tr>
<th>Technique</th>
<th>Cermet</th>
<th>Piezoelectric</th>
</tr>
</thead>
<tbody>
<tr>
<td>Resonance</td>
<td>10.04</td>
<td>15.49</td>
</tr>
<tr>
<td>Acoustic microscopy</td>
<td>9.93</td>
<td>15.52</td>
</tr>
</tbody>
</table>

\[
f_i = 0.5 \left( \frac{1}{w} \right) \left( \frac{C_{eff}^E}{p} \right)^{1/2}
\]

where \( w \) is the width of the piezoelectric sample, \( p \) the density, and \( C_{eff}^E \) is the effective elastic constant where

\[
C_{eff}^E = \frac{1}{S_f(1 - \sigma^2)}
\]

\( S_f \) is the elastic compliance, and \( \sigma \) is the Poisson ratio. From Fig. 6, we can obtain \( C_{eff}^E = 8.63 \times 10^{10} \text{ N m}^{-2} \). Also, via the bending mode, the scan is obtained directly:

\[
f_i = 1.0279 \left( \frac{1}{L^3 S_f p} \right)^{1/2}
\]

where \( L \) is the length of the cantilever. \( S_f \) was determined to be 13.36 \( \times 10^{-13} \text{ N m}^{-2} \). From Eq. (4), \( \sigma \) for the nonreduced phase was determined to be 0.366. Such a value is in good agreement with a previous reference and confirms the validity of the technique. In the lateral direction, the resonant frequency was independent of the reduced and nonreduced phase thicknesses (Fig. 7).

To complete this characterization, the coefficient \( d_{ij} \) of the piezoelectric element was determined. From the resonance method, the \( d_{ij} \) value obtained was \(-150 \times 10^{-12} \text{ C/N}\). A direct technique, the double-beam laser interferometer, was used to verify the \( d_{ij} \) value. Under an alternating-current-\( (ac-) \) driven electric field along the polarization direction \( (E_p) \), the sample deformation \( (\Delta L/L_i) \) and the strain \( (S_i) \) are measured. Through the converse piezoelectric effect:

\[
d_{ij} = \frac{S_i}{E_p} (\Delta L/L_i)
\]

The obtained \( d_{ij} \) value of \(-141 \times 10^{-12} \text{ C/N}\) is in good agreement with the previous value deduced from the resonance method. Such a value is more characteristic of a hard PZT than that expected for the soft PZT studied. This lower \( d_{ij} \) value can be explained by a reduction in the extrinsic domain wall contribution to \( d_{ij} \) in the soft PLZT due to the high transverse constraining stress. From these piezoelectric resonance measurements, the cermet microstructure clearly influences the density and elastic stiffness of the RAINBOW structures, which, in turn, controls the resonant modes.
structural-property relations in a reduced and internally biased oxide wafer (RAINBOW) actuator material

Fig. 6. Resonant frequency of the lateral extensional mode versus inverse width (nonreduced phase).

Fig. 7. Resonant frequency of the lateral extensional mode versus the nonreduced-layer-thickness:cermet-thickness ratio.

Fig. 8. Dielectric constant as a function of temperature for (a) RAINBOW material and (b) the nonreduced phase.

Fig. 9. Frequency dependence of (a) the real part of the permittivity, \( \varepsilon' \), and (b) the imaginary part of the permittivity, \( \varepsilon'' \), at room temperature.
Dielectric measurements have been performed from room temperature up to 250°C for nonreduced PLZT and RAINBOW samples. Figures 8(a) and (b) show the corresponding temperature dependence of the dielectric constant, $\varepsilon''$, at different frequencies. It is interesting to note that removing the electrode on the reduced face of the RAINBOW material does not change the obtained results. verifying the conductive cermet phase to be suitable as an electrode material. The Curie temperature is ~200°C and remains the same when the frequency increases. Such a result is in agreement with a nonrelaxor such as PLZT and corresponds to a composition similar to 6/65/35, which is that of a typical soft PZT. The dielectric permittivity of the RAINBOW material shows a dispersion typical of a space relaxation over the measured temperature range. The room-temperature dielectric behavior shows a relaxation, characterized by a simultaneous maximum of the imaginary part of the permittivity, $\varepsilon''$, and decrease of the real part of the permittivity, $\varepsilon'$, (Figs. 9(a) and (b)). The microstructure observed in Fig. 4(a) is believed to be the origin of this space-charge relaxation. By systematically polishing away the cermet layer and region I and making impedance measurements, the dielectric response confirms this hypothesis on each combination of layers (Fig. 9). The physical removal of region I, the prereduced grain-boundary piezoelectric region, leaves no space-charge polarization effects at room temperature in region II. Therefore, the differences in capacitance and resistance from these mixed regions give the space-charge contribution. The relaxation observed at ~1 kHz depends only on the electrical boundary conditions, and this is not believed to be the result of the stress inherent in the RAINBOW material. This last effect is considered only on the elastic properties.

IV. Summary and Conclusions

A structure-property relationship study of the RAINBOW cermet-ceramic has revealed new insights into the formation process: the microstructure, and electromechanical properties. The cermet microstructure consists of an interconnected metallic lead phase and an interconnected porous structure, with unreduced oxides ZrTiO$_3$ and La$_2$O$_3$ embedded in the lead metal matrix as inclusions. The cermet/ceramic interface is relatively sharp, but there is a prereduction of the grain-boundary phase in the ceramic. The prereduction modifies the grain boundary, as observed in the fracture properties of the ceramic. The prereduction and nonreduced regimes (regions I and II) in the ceramic create a space-charge polarization.

Physical properties that determine the resonance behavior, such as density and elastic stiffness, are given consideration with the observed microstructure. Measurements of the elastic stiffness of the cermet were obtained via a thickness resonance mode and acoustic microscopy and showed excellent agreement.

Acknowledgments: The authors wish to thank Michael Hill of NIST (Maryland) for the acoustic microscopy measurements. Many thanks also go to Mark S. Angeline for his technical input on the SEM microscope.

References


Determination of Young's modulus of the reduced layer of a piezoelectric RAINBOW actuator

Qing-Ming Wang and L. Eric Cross

Intercollege Materials Research Laboratory, The Pennsylvania State University, University Park, Pennsylvania 16802-4800

(Received 6 November 1997; accepted for publication 12 February 1998)

The reduced and internally biased oxide wafer (RAINBOW) actuator, a new type of monolithic piezoelectric bending device, is composed of a reduced electromechanically passive layer and an unreduced piezoelectric layer. The determination of the elastic properties of the reduced layer is important for optimizing actuator performance. In this paper, an analytical expression for bending resonant frequency is derived for the cantilever RAINBOW actuator by using composite beam theory. A resonance method is then described for determining Young's modulus of the reduced layer of RAINBOW actuator. As a piezoelectric bending actuator, RAINBOW can be excited by its electromechanically active PZT layer without using any external excitation method. Young's modulus can be calculated by measuring its resonant frequency, sample geometry, and densities of component parts. For comparison, the Young's modulus is also determined by measuring the resonant frequency of a completely reduced plate with an external acoustic excitation method. The results obtained by these two excitation methods show good agreement. © 1998 American Institute of Physics. [S0021-8979(98)05010-5]

I. INTRODUCTION

In past few years, there has been a considerable research interest in developing novel materials and devices for electromechanical applications. Piezoelectric ceramic actuators convert electrical energy into mechanical energy via the inverse piezoelectric effect, while piezoelectric ceramic sensors convert external mechanical energy into an electric signal via the direct piezoelectric effect. Due to these unique electromechanical properties, together with some other advantages such as light weight, distributed characteristics, high coupling factor, quick response, low energy consumption and low cost, piezoelectric or electrostrictive ceramic solid state actuators are receiving much attention for numerous applications including micromachines, acoustic sensing, loud speakers, active vibration control, etc. The most typical piezoelectric actuators include multilayer stacks with internal electrodes, bimorph/unimorph benders and flextensional composite actuators "moonie" and "cymbal." The advantages of the multilayer actuator are its large generative force and quick response speed. But small displacement and high capacitance make them impractical for certain applications. Bimorph or unimorph benders can be used in the cases where large displacement is desirable and low force can be tolerated. For applications where intermediate level displacement and generative force are required, composite actuator moonie and cymbal can be applied. However, in all these actuators reliability is always a concern. Performance degradation, delamination, as well as fracture are usually observed in these actuators in practical applications due to bonding or structure problems, especially when under high field and long time periodical driving. More recently, a new type of monolithic piezoelectric/electrostrictive bending actuator, namely RAINBOW (reduced and internally biased oxide wafer) device, was developed by Haertling in Clemson University. It was reported that the RAINBOW actuator offers some unique advantages over conventional bimorph and unimorph actuators such as: (i) monolithic composite structure providing good reliability, (ii) large axial displacement level due to its domelike configuration, and (iii) high mechanical strength due to the existence of internal compressive pre-stresses.

Since the appearance of the RAINBOW actuator, considerable research work has been conducted to characterize the electromechanical properties of this device. It is believed that the reduced layer has a different elastic modulus with the active piezoelectric lead zirconate titanate (PZT) or lanthanum modified lead zirconate titanate (PLZT) ceramic layer. To achieve maximum output displacement and force level from the RAINBOW actuator, a suitable thickness ratio of reduced layer and PZT or PLZT layer has to be designed and fabricated, which in turn requires the determination of the elastic modulus of the reduced layer in this composite. However, when discussing the characteristics of RAINBOW actuators, previous works simply assumed that the reduced layer has the same Young's modulus as the piezoelectric PLZT layer, which may lead to inaccurate results in actuator design and fabrication.

Several methods could be used for accurate determination of elastic modulus. One is ultrasonic pulse-echo technique. In this method, an ultrasonic beam is generated by a transducer which is attached to the measured materials. By measuring the sound velocity in the medium, the elastic properties can be determined. The other is the resonance method. By measuring the resonant frequency, elastic properties can be calculated because resonant frequency is related

---

a)Electronic mail: qxw4@psu.edu
b)Author to whom correspondence should be addressed.
to the structure, elastic properties and dimensions of material. For direct measurement of Young's modulus of the reduced layer of a thin RAINBOW actuator, the ultrasonic pulse-echo method may have some technical difficulties due to the small density difference between the reduced layer and remaining piezoelectric layer. The echo reflected from the interface between these two layers could be too weak to measure precisely.

In this paper, the resonance method is used to determine the Young's modulus of the reduced layer of a soft PZT-based thin cantilever RAINBOW actuator. Taking into account the composite beam structure, an analytical expression is derived for the bending resonant frequency. Conveniently, the RAINBOW actuator can be driven internally due to its piezoelectric nature. Young's modulus can be calculated by measuring its resonant frequency, sample geometry, and densities. For comparison, the Young's modulus of the transformed section is equal to the moments of the areas of the transformed layers divided by the total cross-sectional area:

$$n = \frac{(E_r/E_p)w_t(t_r/2)+wt_p(t_r+t_p/2)}{(E_r/E_p)w_t-wt_p}$$

where $n$ is the position of the transformed neutral axis measured with respect to the bottom of the reduced layer. Using the parallel-axis theorem,$^{13}$ the area moment of inertia $I_x$ is calculated for each layer as a moment of inertia plus a term relative to the transformed neutral axis, i.e.,

$$I_x = A \frac{wt_r^2}{12} + Aw_{t_p} \left[ n - \frac{t_r}{2} \right]^2 + \frac{wt_p^2}{12} + wt_p \left( t_r + t_p/2 - n \right)^2$$

where $A = E_r/E_p$, the equivalent flexural rigidity of the RAINBOW actuator is, therefore, $E_r I_x$, and this can be used in Eq. (3) to determine the natural bending resonance frequency,$^{14}$ $f_1$, for a beam fixed at one end and free at the other (i.e., cantilever) with uniform mass per unit length $m$ and the length $L$:

$$f_1 = \frac{\lambda_1^2}{2\pi L^2} \left( \frac{E_{p,c}}{m} \right)^{1/2}$$

Here, Young's modulus of the piezoelectric layer $E_p$ has to be used because the reduced layer has been transformed with respect to it. $\lambda_1$ is the eigenvalue where $i$ is an integer that describes the resonance mode number: for the first mode, $\lambda_1 = 1.43$, and for the second mode, $\lambda_2 = 4.69$.

For the uniform cantilever RAINBOW structure, the mass per unit length can be expressed as:

$$m = \rho_r + \rho_p$$

II. BENDING RESONANT FREQUENCY OF CANTILEVER RAINBOW ACTUATOR

The structure of a cantilevered RAINBOW actuator is shown schematically in Fig. 1(a). Bending deformation occurs during the reduction process in the RAINBOW actuator due to a thermal expansion coefficient difference between the reduced conductive layer and the remaining PZT or PLZT layer, which generates large internal stresses. Therefore, the structure of a cantilever RAINBOW actuator is basically a curved composite beam. We assume that the thickness of reduced and piezoelectric layer are $t_r$ and $t_p$, respectively. The transformed section model$^{15}$ is used to redefine the RAINBOW structure.

Figure 1(b) shows the actual cross section of a RAINBOW actuator. The effect of electrodes is ignored here because their thickness is much less than $t_r$ or $t_p$. The transformed section method for the composite beam allows the width of the beam layers to be proportioned by the ratio of their Young's modules, thereby defining the entire beam as having one Young's modulus. If the modulus for both reduced layer and piezoelectric layer are $E_r$ and $E_p$, respectively, the transformed width of the reduced layer is $E_r/E_p$ of its original width as shown in Fig. 1(c). The transformed structure behaves as though it has a single modulus of $E_p$. The neutral axis of the transformed RAINBOW cross section is equal to the moments of the areas of the transformed layers divided by the total cross-sectional area:

$$n = \frac{(E_r/E_p)w_t(t_r/2)+wt_p(t_r+t_p/2)}{(E_r/E_p)w_t-wt_p}$$

The neutral axis of the transformed RAINBOW cross section is equal to the moments of the areas of the transformed layers divided by the total cross-sectional area:

$$n = \frac{(E_r/E_p)w_t(t_r/2)+wt_p(t_r+t_p/2)}{(E_r/E_p)w_t-wt_p}$$

where $n$ is the position of the transformed neutral axis measured with respect to the bottom of the reduced layer. Using the parallel-axis theorem,$^{13}$ the area moment of inertia $I_x$ is calculated for each layer as a moment of inertia plus a term relative to the transformed neutral axis, i.e.,

$$I_x = A \frac{wt_r^2}{12} + Aw_{t_p} \left[ n - \frac{t_r}{2} \right]^2 + \frac{wt_p^2}{12} + wt_p \left( t_r + t_p/2 - n \right)^2$$

where $A = E_r/E_p$, the equivalent flexural rigidity of the RAINBOW actuator is, therefore, $E_r I_x$, and this can be used in Eq. (3) to determine the natural bending resonance frequency,$^{14}$ $f_1$, for a beam fixed at one end and free at the other (i.e., cantilever) with uniform mass per unit length $m$ and the length $L$:

$$f_1 = \frac{\lambda_1^2}{2\pi L^2} \left( \frac{E_{p,c}}{m} \right)^{1/2}$$

Here, Young's modulus of the piezoelectric layer $E_p$ has to be used because the reduced layer has been transformed with respect to it. $\lambda_1$ is the eigenvalue where $i$ is an integer that describes the resonance mode number: for the first mode, $\lambda_1 = 1.43$, and for the second mode, $\lambda_2 = 4.69$.

For the uniform cantilever RAINBOW structure, the mass per unit length can be expressed as:

$$m = \rho_r + \rho_p$$

Substituting Eq. (1) into Eq. (2), we have Eq. (5):

$$I_x = A \frac{wt_r^2}{12} + aw_{t_p} \left[ n - \frac{t_r}{2} \right]^2 + \frac{wt_p^2}{12} + 4 \frac{Aw_{t_p} \left( t_r+t_p \right)^2}{At_r+t_p}$$

Therefore, for the first bending resonance mode, the resonant frequency can be expressed as:

$$f_1 = \frac{3.52 \left( \frac{E_p}{2\pi L^2} \left( \frac{t_r+t_p}{At_r+t_p} \right)^{1/2} \left( \frac{t_r}{12} + \frac{At_r+t_p}{12} \right)^{1/2} \right.}{\left. \frac{t_r+t_p}{At_r+t_p} \right)^{1/2} \left( \frac{t_r}{12} + \frac{At_r+t_p}{12} \right)^{1/2} \right) \left( \frac{t_r}{12} + \frac{At_r+t_p}{12} \right)^{1/2} \right) \left( \frac{t_r}{12} + \frac{At_r+t_p}{12} \right)^{1/2} \right)$$

Assuming the Young's modulus of the piezoelectric layer, the thickness of both reduced layer and piezoelectric layer, and materials densities are known, from the measured bending resonance frequency, Young's modulus of the reduced layer can be calculated using Eq. (6).

As a special case, when $E = E_p = E_r$, $\rho = \rho_p = \rho_r$ and $t = t_r+t_p$, Eq. (6) can be simplified to:

$$f_1 = \frac{3.52 \left( E \right)^{1/2}}{4 \pi L^2}$$
This is exactly the fundamental bending resonance frequency for a cantilever single beam structure and for a cantilever bimorph actuator.

III. EXPERIMENT

A. Preparation of RAINBOW actuator

The preparation processing for the RAINBOW actuators is similar to that reported by Haertling. Commercial Motorola soft PZT 3203 HD (5H-type) ceramics were purchased for fabricating the RAINBOW actuator. The rectangular ceramic block was cut into thin plates with dimension of 55.0 mm × 15.0 mm × 1.01 mm. After cutting, the ceramic plate was placed on a piece of high density flat carbon block with ground surface. This assembly was then heated to 975 °C with a heating rate of 300 °C per hour in a furnace at normal air atmosphere. The PZT ceramic plate was treated at a temperature of 975 °C for 8 h and then cooled down to room temperature rapidly. Here, to achieve a certain reduced layer thickness, a suitable reduction temperature and time should be chosen, which depend on material properties such as grain size, density and chemical nature. The reduction of the PZT ceramic plate occurs as a result of oxidation of the solid carbon block. As a consequence of this heat treatment, an electrically conductive but electromechanically inert layer with black color was formed. Due to the thermal expansion mismatch between the reduced layer and the PZT layer, large internal stress was generated in the the ceramic part, which results in bending deformation of ceramic plate after cooling. A very sharp reduced layer/PZT interface was observed by using an optical microscope and the thickness of the reduced layer and PZT layer, \( t_r = 0.42 \) mm, \( t_p = 0.60 \) mm, are measured. The RAINBOW actuator was then electroded by sputtering gold on the major surfaces and poled under a dc field of 2.0 kV/mm in fluorinert poling oil at 90 °C for 1 min. After poling, it was observed that the RAINBOW actuator become flatter, indicating that the internal thermal stress was partially released.

B. Measurement

Usually there are three methods that may be used to excite the thin cantilever structure for resonance measurement: (1) acoustic excitation using a loud-speaker; (2) mechanical excitation using a piezoelectric transducer; (3) photothermal excitation using a laser beam. In the case of resonance measurement of the RAINBOW actuator, due to the intrinsic piezoelectric activity of poled PZT ceramic layer, when the electric field is applied on the PZT layer, lateral dimensional change will be induced. However, this dimensional change is opposed by the reduced layer. As a consequence, a bending vibration is generated. As in any other piezoelectric resonator, drastic changes of both electrical impedance and phase can be observed at bending resonance. Therefore, the resonant frequency of the RAINBOW actuator can be obtained through impedance/phase spectrum measurement under electric field excitation. An impedance/gain phase analyzer (Model HP4194A, Hewlett-Packard Co., Palo Alto, CA) was used for resonant measurement. To measure the Young’s modulus of the PZT layer, a cantilever PZT bimorph actuator with similar dimension was prepared by bonding two thin PZT plates with opposite polarization directions (series connection). The dimension of each PZT thin plate is 60.0 mm in length, 6.68 mm in width and 0.55 mm in thickness. The thickness of silver epoxy bonding layer is less than 10 μm. Thus its effect can be ignored. The bending resonance frequency of the bimorph actuator was also obtained through the impedance/phase measurement. Densities of completely reduced samples and PZT ceramic samples were measured by using a water immersion method.

For comparison, the Young’s modulus of the reduced layer was also determined by measuring the resonant frequency of a completely reduced thin plate with the external acoustic excitation method. In this case Eq. (7) was used. The experimental setup is shown in Fig. 2. A variable-frequency signal generator is connected to a small loudspeaker to produce a mechanical driving vibration. One end of the sample is clamped. An optical displacement measurement system (MTI 2000 Fotonic Sensor, MTI Instrument) is used to detect the amplitude of vibration of the cantilever. A very small mirror, made by sputtering thin gold film on a small glass piece, is attached to the cantilever tip for reflecting the incident light from an optic probe. A lock-in amplifier (SR830 DSP, Stanford Research System, Inc.) which synchronized with the output voltage of the signal generator is used to measure the output signal from the MTI sensor.

IV. RESULTS AND DISCUSSION

Densities of reduced layer and PZT ceramic are listed in Table I. It is found that the reduced layer has a higher density than PZT ceramics. The density change is due to the loss of oxygen and the formation of heavy metal Pb during the reduction processing. X-ray diffraction analysis indicated that the reduced layer is a mixture of metal oxides such as ZrO₂ and TiO₂, metal Pb, and partially reduced PZT ceramic phases. Scanning electron microscopy (SEM) observation demonstrated that a continuous lead metal phase is located...
The bending resonant frequency of the bimorph actuator was also measured by the same technique for the determination of Young's modulus of PZT ceramics.

To reduced errors in the measurement of resonant frequency, several RAINBOW and bimorph actuators with different lengths but the same thickness and width are used in the measurement. In Eqs. (6) and (7), the bending resonant frequency \( f_r \) is proportional to \( 1/L^2 \):

\[
f_r = k \times \frac{1}{L^2}
\]

by plotting the bending resonant frequency \( f_r \) against \( 1/L^2 \), a straight line should be obtained with a slope:

\[
k = \frac{3.52f}{2\pi} \left( \frac{E_p}{t_p \rho_p + t_p \rho_f} \right) \left[ \frac{A r}{12} \frac{r_p^2}{12} + \frac{A t f_p}{4} \right]^{\frac{1}{2}}
\]

for cantilever RAINBOW actuators, or:

\[
k = \frac{3.52f}{4\pi} \left( \frac{E_p}{3\rho_f} \right)^{\frac{1}{2}}
\]

for bimorph actuators so that the Young's modulus can be calculated.

The bending resonant frequencies \( f_r \) as a function of \( 1/L^2 \) for both cantilever bimorph and RAINBOW actuators are depicted in Figs. 4 and 5, respectively. As expected, straight lines are obtained in both cases. The Young's modulus of PZT ceramics was first calculated from the slope of Fig. 4. The result is \( E_p = 3.4 \times 10^{10} \text{ N/m}^2 \). This value and the slope obtained from Fig. 5 are then substituted into Eq. (9), and A
can be calculated through nonlinear numerical fitting. The result is $A = 0.84739$, thus the Young's modulus of reduced layer is obtained: $E_r = 2.88 \times 10^{10}$ N/m².

The value of Young's modulus of PZT ceramic thus obtained is on the same order of magnitude but lower than the data provided by the manufacturer, which is $E_{f1} = 6.2 \times 10^{10}$ N/m². The reason for this may be due to the difference measurement technique. It was found that if we substituted these data to calculate the Young's modulus of the reduced layer, the result obtained was unreasonably low. For comparison purposes, a completely reduced thin sheet was cut for the direct measurement of Young's modulus. A small loud-speaker was used as an external acoustic excitation source for the determination of the fundamental bending resonant frequency of a cantilever beam made from this completely reduced plate. In Fig. 6(a), the vibrational amplitude and phase of this cantilever beam is plotted against the driving frequency of the loud-speaker. Its bending resonant frequency can be determined, which corresponds to the amplitude maximum of the cantilever vibration. Fig. 6(b) shows the plot of resonant frequency against $1/L^2$. From the slope, the Young's modulus of this reduced sheet can be determined by Eq. (6). The result is $E_r = 2.98 \times 10^{10}$ N/m², which is in good agreement with the calculated value, indicating that the calculated value is valid. The small difference may be accounted for by the fact that the reduced layer in a RAINBOW actuator is subjected to internal stress, while the completely reduced sample is free of internal stress. The measurement error is another possible reason.

On the other hand, because the RAINBOW actuator can be regarded as two different layers bonded mechanically in parallel to the length direction. The Young's modulus of the whole composite beam structure along the length can be represented by the empirical mixing rule:

$$E_{ct} = E_{cp} t_p + E_{cr} t_r,$$  \hspace{1cm} (11)

where $E_c$ and $t_c$ are the effective Young's modulus and total thickness of composite beam. We can calculate $E_c$ by Eq. (10) using the slope of Fig. 5 and the density of composite $\rho_c$, which is obtained by $\rho_c = \rho_f t_f + \rho_p t_p$. The result is $E_c = 3.125 \times 10^{10}$ N/m². Substituting this result into Eq. (11), $E_c$ can be calculated as: $2.73 \times 10^{10}$ N/m², which agrees quite well with the value calculated by Eq. (9). However, it is believed that Eq. (9) is more accurate for Young's modulus determination.

The accuracy in the determination of Young's modulus depends on the errors introduced in the measurements of density, thickness, length and resolution of resonant fre-
frequency. It is estimated that the inaccuracy is less than 10%, indicating the obtained Young’s modulus is quite reliable and can be used for optimization of the RAINBOW actuator performance.

V. CONCLUSION

An expression relating bending resonant frequency with Young's modulus of a RAINBOW structure is derived using composite beam theory. The Young’s modulus of the reduced layer of the RAINBOW actuator is then determined by the bending resonant frequency measurements through both piezoelectric and external acoustic excitation methods. The results obtained by these two methods showed good agreement. Compared with PZT ceramics, the reduced layer has a lower Young’s modulus, which is probably due to the presence of metal Pb phase in the reduced layer and relative loose structure after the high temperature reduction process. The electrical impedance/phase measurement technique has been shown to be a reliable nondestructive method for the Young’s modulus determination.

ACKNOWLEDGMENTS

This work was supported by the Office of Naval Research under Contract No. N00014-94-1-1140, which is gratefully acknowledged.

APPENDIX 9
A piezoelectric bimorph transducer utilizing piezoelectric $d_{33}$ coefficient was developed. This bimorph consists of piezoelectric segments bonded by a polymeric agent and was fabricated by a diving and layering technique. The transducer has superior piezoelectric characteristics compared to standard piezoelectric $d_{31}$ bimorphs. Piezoelectric coefficients, electrical admittance, mechanical compliance, and losses of the actuator were found to increase with increasing driving electric field.

Piezoelectric bimorph actuators have been known for many years and are widely used for efficient interconversion of electrical and mechanical energy. A typical piezoelectric bimorph transducer consists of two similar piezoelectric plates poled along their thickness (Z axis) and adhesively bonded together (Fig. 1): in the case of a piezoelectric/metal bimorph a metal plate is replaced for one of the piezoelectric plates. Instead of a "piezoelectric/metal bimorph" notion, we will use "unimorph" in the sense that this type of bimorph has only one piezoelectric plane. Clearly, the flexural displacement in this actuator is caused by the piezoelectric effect ($d_{31}$) in the direction perpendicular to the polar P axis. The static tip displacement $\eta$ and corresponding blocking force $F_{31}(\eta=0)$ of the piezoelectric bimorph cantilever can be written as

$$\eta = \frac{3}{2} \frac{d_{31}}{\varepsilon} \frac{E}{Z}$$

$$F_{31} = \frac{3}{2} \frac{d_{31}}{\varepsilon} \frac{E}{Z}$$

where $d_{31}$ is the piezoelectric coefficient, $\varepsilon$ is the mechanical compliance in the direction Y under the constant electric field, and $Z$, $w$, and $t$ are the dimensions of the cantilever (Fig. 1). Mechanical stress $T_1$ arising in the unloaded piezoelectric cantilever under the applied electric field is directed along the X axis.

Commercial piezoelectric bimorph actuators are usually made from soft PZT ceramics with the composition near the morphotropic phase boundary. For these compositions, piezoelectric coefficient $d_{33}$ along the polar axis and corresponding coupling factor $k_{33}$ are 2-12 times larger than $d_{31}$ and $k_{31}$. Studies of the effect of the mechanical stress on piezoelectric properties of the ceramics also show that piezoelectric coefficient $d_{33}$ is much less sensitive to the compressive stress $T_1$ along the polar axis than coefficient $d_{31}$ is to the compressive stress $T_1$ perpendicular to the polar axis. Thus, the piezoelectric bimorph actuator utilizing $d_{31}$ coefficient can be expected to have superior characteristics compared to a standard $d_{31}$-type bimorph transducer.

In this Letter, we report on a novel type of piezoelectric bimorph and unimorph in which the piezoelectric effect along the polar axis is used to generate flexural displacement. A schematic view of the developed caterpillar-type $d_{33}$ unimorph is shown in Fig. 2. From Fig. 2 it is clear that, when the piezoelectric plate changes its length under the applied voltage $V$, it causes bending of the cantilever. Unlike in the standard $d_{31}$-type bimorph actuator (Fig. 1), the bending moment in these transducers is caused by piezoelectric $d_{33}$ coefficient. To fabricate the $d_{33}$ bimorph the metal plate is replaced by an identical piezoelectric plate. In this case the driving voltage is applied in such a way that mechanical strains generated in the top and bottom piezoelectric plates have opposite signs.

The experimental process adopted for fabricating the $d_{33}$-type transducer is described below. Initially, a stack of similar piezoelectric electrode plates poled along their thickness is fabricated. Adjacent plates have opposite directions of spontaneous polarization $P$. The plates are bonded to each other by a very thin layer of conductive polymer. Then the stack is sliced by means of a diamond saw in the plane parallel to the direction of spontaneous polarization. To fabricate a unimorph transducer, the sliced plate is glued to a metal substrate (see Fig. 2). Conducting electrodes are deposited on the surface of the piezoelectric plate in such a way that voltage $V$ applied to these electrodes produces the same piezoelectric strain (positive or negative) in each of the piezoelectric segments comprising this plate. To make a $d_{33}$ bimorph, a metal plate is replaced by an identical piezoelectric plate. It is important to note that the conductive polymer serves not only for electric connecting, but also for improving fracture toughness. Therefore, the transducer developed

![Figure 1: Bimorph piezoelectric $d_{33}$ element with series connection. The vector of the spontaneous polarization $P$ is directed along the Z axis.](image-url)
can sustain a much larger bending force without mechanical failure.

In our experiments we used soft piezoelectric ceramics PK1550 (Piezo Kinetic, Inc.) and stainless steel SS302. The ceramic plates in the stack were bonded using commercial conductive adhesives EP41TDC (Master Bond, Inc.) and E-Soilder 3025 (Insulating Materials, Inc.). J-B Weld epoxy (J-B Weld Company) was used for bonding metal and sliced ceramic plates. Each piezoelectric segment in the piezoelectric plates (Fig. 2) had the following dimensions: \(t_1 = t_2 = 1.39 \text{ mm}, w = 11 \text{ mm}\). Stainless steel plates had thicknesses: \(t_a = 0.025, 0.029, \text{ and } 0.37 \text{ mm}\). For comparison, standard \(d_{33}\)-type bimorphs and unimorphs of identical dimensions were also fabricated.

To characterize the transducer developed, the following parameters of these transducers in the unimorph configuration \(t_a = 25.0 \text{ mm}\) were measured: \(i\) displacement \(\gamma\) of the free end, \(ii\), blocking force \(F_b\), and \(iii\), electrical admittance as a function of the applied electric field. All measurements were done at a frequency of 100 Hz which is about 1 kHz the first resonant frequency of the bending vibrations.

The displacement and blocking force (rms) for bimorphs and unimorphs are shown in Fig. 3 and 4, respectively. It is clear that the displacement of the \(d_{33}\) bimorph is about 2.5 times larger than that of the \(d_{31}\) bimorph and the blocking force is about 1.5 times larger. Although the displacement of the \(d_{33}\) unimorph is smaller than the displacement of the \(d_{31}\) bimorph, it is approximately 1.3 times that of the \(d_{31}\) unimorph. The blocking force of \(d_{33}\) unimorphs is about 2.0 times that of the \(d_{31}\) unimorph. Thus, the developed \(d_{33}\) bimorphs and unimorph actuators have superior piezoelectric characteristics compared to those of \(d_{31}\) actuators. Relative characteristics of the above mentioned transducers for a low electric field \((-10 \text{ V/cm})\) are given in Table I.

There is clearly an optimal ratio \(t_a/t_2\) for unimorphs that gives maximum displacement. Equation (1) for \(d_{33}\) unimorphs can be written (through analysis will be given elsewhere) as

\[
F_b = \frac{\pi t_a^2}{2}\frac{\epsilon_0\sigma E^2}{d_{33}^2}
\]

\[
\gamma = \frac{2t_a d_{33}^2}{\pi t_a^2}\frac{\epsilon_0\sigma E^2}{d_{33}^2}
\]

Figure 4. The dependence of displacement and blocking force for unimorphs on the electric field. The thickness of the metal plate is 0.37 mm. The thickness of the \(d_{33}\) bimorph is \(t_a = 0.029 \text{ mm}\).

Table I. Relative characteristics of various types of transducers.

<table>
<thead>
<tr>
<th>Type of Piezoelement</th>
<th>Displacement</th>
<th>Blocking Force</th>
</tr>
</thead>
<tbody>
<tr>
<td>(d_{31}) bimorph</td>
<td>1</td>
<td>10</td>
</tr>
<tr>
<td>(d_{31}) unimorph</td>
<td>0.4</td>
<td>1.30</td>
</tr>
<tr>
<td>(d_{33}) bimorph</td>
<td>2.4</td>
<td>1.52</td>
</tr>
<tr>
<td>(d_{33}) unimorph</td>
<td>0.7</td>
<td>5.30</td>
</tr>
</tbody>
</table>

Kapel, Chandran, and Gross
where $E = V/a_0$, $Y_m$ is the Young's modulus of the metal plate, and $s_{ij}$ is the mechanical compliance of ceramics in the direction of the polar axis. For bimorph cantilevers the coefficients $k_d$ and $k_d'$ in Eq. (2) are equal to 1. Theoretical values of $k_d$ and $k_d'$ and corresponding experimental data obtained from the comparison of displacements and blocking forces of the $d_{33}$ bimorph and unimorphs are given in Fig. 5. A relatively large difference between the theoretically derived and experimentally observed values for the blocking force can be attributed to the effect of the bonding layer (Fig. 3) whose thickness has been neglected in the theoretical calculations.

A close analysis of the displacement, blocking force (Figs. 3 and 4) by means of Eqs. (1) and (2), and electrical impedance of the actuators studied shows that piezoelectric, mechanical, and dielectric properties of the transducers fabricated from soft piezoelectric ceramics depend upon the magnitude of the electric field (Fig. 6). We also found that the imaginary parts of $d_{31}$, $d_{33}$, $s_{11}$, $s_{33}$, and the electrical admittance increase with increasing electric field (through analysis will be given elsewhere). Clearly, the relative displacement is proportional to the piezoelectric coefficients and the blocking force is proportional to the ratio $d_{ij}$/ where $\gamma$ is the mechanical compliance of the whole transducer. From Fig. 6 it can be seen that the increase in $d_{31}$ and $\gamma$ with increasing electric field is more than that in $d_{31}$ and $s_{33}$. According to known results the nonlinear behavior of PZT ceramics has an intrinsic nature, i.e., it is caused by domain wall and interphase interface motion. In a bimorph transducer, two factors can affect the behavior of piezoelectric coefficients under an external electric field. The first one is an increase in the piezoelectric coefficients with increasing electric field of the second one is the effect of the internal stress that exists in the bimorph even without the external load. It is known that the decrease in $d_{31}$ due to the stress $T_3$ is much greater than the increase in $d_{33}$ due to $T_3$. Therefore, it is reasonable to expect that the enhancement of the displacement of the $d_{31}$ bimorphs under the electric field will be larger than that of the $d_{33}$ bimorph; this is consistent with the experimental data (Fig. 5). The same two factors also affect the mechanical compliance and dielectric permittivities: in addition, the conductive polymer causes mechanical "softening" of $d_{33}$ actuators as well.

In summary, we have developed and studied in significant detail, a novel type of piezoelectric bimorph transducer based on the piezoelectric $d_{31}$ coefficient. Superior piezoelectric characteristics compared to piezoelectric $d_{33}$ bimorphs were obtained. Piezoelectric coefficients, electrical admittance, mechanical compliance, and losses of the actuators increase with increasing driving electric field.

This work was supported by the Office of Naval Research and under Contract No. N00014-84-1-1140.

CHARACTERISTICS OF SHEAR MODE PIEZOELECTRIC ACTUATORS

Qing-ming Wang, Baomin Xu, V. D. Kugel and L. E. Cross

187 Intercollege Materials Research Laboratory
The Pennsylvania State University
University Park, PA 16802

Abstract: Characteristics of cantilever shear mode piezoelectric actuator have been investigated. In this actuator configuration, soft PZT ceramic plate was poled along length and driven across its thickness, with one end mechanically clamped and the other free. Experimental results indicated that relative large tip displacement can be obtained through nonlinear piezoelectric response at high driving field. Due to lateral shear force, mechanical bending also contributes to the tip displacement of shear mode actuator. The fundamental bending resonant frequency was observed in the frequency range from 150 Hz to 600 Hz, depending on thickness and length of actuator. We also found that the resonant frequency of bending vibration is dependent on the driving field because of elastic nonlinearities. Two layer and multilayer shear mode actuators were also developed to reduce the driving voltage, while shifting the bending resonance to higher frequency range by increasing the total thickness of actuator.

Key words: PZT, Ceramic actuator, Shear mode

I. Introduction

Piezoelectric ceramic materials such as lead zirconate titanate (PZT) have been designed as solid-state actuators and sensors for many applications such as precision positioning, noise and vibration sensing and cancellation, underwater speaker, linear motor, and many others (1). The three most common types of piezoelectric actuators are multilayer ceramic actuators, bimorph or unimorph actuators and extensional composite actuators. Multilayer actuators, in which about 100 thin ceramic sheets are stacked together with inter-electrodes utilizing the direct extensional mode ($d_{31}$ mode), are characterized by large generated force, high electromechanical coupling, high resonant frequency, low driving voltage and quick response but small displacement level. On the other hand, bimorph or unimorph actuators consist of two thin ceramic sheets or one ceramic and one metal sheet bounded together with the poling and driving sections normal to the interface. When driving, the ceramic extension and shrinkage of ceramic sheets due to inverse mode ($d_{33}$) result in a pure bending vibration. Bimorph and unimorph actuators can generate large displacement level but low electro-mechanical coupling, low resonant frequency and low driving force. For flexextensional composite actuator, two typical examples are so-called "moonie" and "cymbal" actuators (2), (3) which consist of a piezoelectric or electrostrictive ceramic disk and metal end-caps. The ceramic is excited in an extensional mode and the metal caps in a flexure mode. The metal end-caps act as a mechanical transformer converting and amplifying the radial displacement ceramic disk into linear axial motion. Very large effective $d_{33}$ coefficient can be obtained in "moonie" and "cymbal" actuators which results in quite large displacement levels.

Monomorph actuator (4), basically a modified unimorph actuator, made from a semiconductive piezoelectric ceramic thin sheet coated with certain electrodes. Quite a large tip displacement (100um) arises from the non-uniform distribution of the electric field which occurs at semiconductor-metal electrode interface. The monomorph actuator has a simple structure in which bonding problems usually found in unimorph or bimorph structure can be avoided. A more recent device, Rainbow actuator, developed by Haertling(5), is also a monolithic unimorph-type ceramic actuator which is produced by selectively reducing one surface of a PZT ceramic wafer on carbon black at high temperature. The reduced layer is a good electrical conductor and it acts as both the electrodes and inert part of the actuator. There is a very large internal pre-stress developed in the Rainbow actuator during reducing processing. It was reported (6) that Rainbow actuator is capable of achieving very high axial displacement (>1000 um) and sustaining moderate pressures.

Obviously, mechanical motion generated by all the above ceramic actuators utilizes or is related to either $d_{33}$ longitudinal mode or $d_{31}$ transverse mode. It is interesting to note that although the $d_{33}$ which couples to the shear mode is the highest coefficient in soft PZT ceramic materials, piezoelectric shear vibration mode is seldom used in actuator and transducer applications. In this paper, cantilever shear mode actuator and its characteristics will be presented. In this actuator, ceramic plate is poled along length and driven across its thickness with one end clamped and the other end free. The advantage of cantilever shear mode actuator is its simple, monolithic structure which allows for easy fabrication. However, its displacement level is relatively low comparing with bimorph or unimorph actuators.
II. Experimental Procedures

(1) Sample preparation:

In this study, Soft PZT 3203HD (Motorola Ceramic Product, Albuquerque, New Mexico) ceramic material was used to prepare shear mode actuator. Rectangular ceramic block with dimensions of 38.1 mm x 38.1 mm x 25.4 mm was electrodeposited with air-dry silver paste and poled at a temperature of 90°C in oil bath for 45 minute under dc electric field of 1.8 kV/mm along its length. After poling, the ceramic block was aged for more than one week at room temperature. Then d$_{33}$ coefficient was measured by using Berlincourt d$_{33}$ meter. The measured value, 684 x 10$^{-12}$ m/V, comparing with 650 x 10$^{-12}$ m/V provided by manufacturer data sheet, indicates the ceramic block was well poled. Then the silver electrodes were removed and the poled ceramic block was cut into thin plates with 38.1 mm length, 12.5 mm width and thickness from 0.3 mm to 1.6 mm. Gold electrodes were sputtered on the major surfaces for applying electric field across thickness. One end of ceramic plate was clamped with rigid plastic support which is mounted on an optic plate with micropositioner (Ealing Electro-Optics, Inc.) while the other end free.

(2) Measurement

To characterize the shear mode actuators, the tip displacement was measured as function of both frequency and driving field. The measurement system was schematically shown in Fig. 1. MTI 2000 Fotonic Sensor (MTI Instrument) was used for displacement measurement. A small mirror was attached on the actuator tip for reflecting the incident light from optic fiber. When actuator vibrates under ac field driving, the reflected light is detected and transferred to voltage signal or displacement signal by the sensor. The optic fiber head was mounted on an XYZ micropositioner which provides distance adjustment against the measured actuator. Manual calibration was performed to determine the correspondence of displacement and output signal of the sensor. The AC signal generated by DS345 function generator (Stanford Research System, Inc.) was amplified through a powder amplifier (790 series, PCB Piezotronics, Inc.). The output of the powder amplifier was then applied on the ceramic actuator. A lock-in amplifier (SR330 DSP, Stanford Research System, Inc.) which synchronized with the output voltage of the power amplifier was used to measure the output signal from MTI sensor. An Oscilloscope was also used to monitor the applied voltage and output signal from MTI sensor. The maximum output voltage of the power amplifier is 300 volts RMS. When even higher driving field is needed, a power source with maximum output voltage of 500 Volts is used.

III. Results and Discussions

Linear piezoelectric shear strain, tip displacement and blocking for pure shear mode actuator can be written as:

\[ \varepsilon = d_{33} E_1, \]  
\[ x = d_{33} E_1 L. \]  

![Figure 2. Driving field dependence of tip displacement of shear actuators with different thickness (L = 32.2 mm, w = 12.3 mm)](image-url)
and piezoelectric coefficients will increase. Our experimental results on various PZT ceramic materials indicate that piezoelectric $d_{31}$ coefficient increases as driving field amplitude. Especially for soft PZT, nonlinearity occurs even at relatively low electric field level which consequently contributes to the nonlinear electric field response of tip displacement of the cantilever shear mode actuators. On the other hand, for cantilever shear mode actuator, one end is mechanically clamped thus can not generate pure shear motion under high driving field. Due to this boundary condition, a lateral force which perpendicular to the actuator surface will be produced when driving under electric field. In the case of thin ceramic plate, mechanical bending will then be easily resulted which enhances the tip displacement of ceramic actuator. The thinner the ceramic plate, which is easier to be bent thus the larger the tip displacement. The induced mechanical bending mode can be further clarified by Fig. 3 which shows frequency response of tip displacement for shear actuator with 0.9 mm thickness under driving field of 100 V/mm. The observed fundamental bending resonant frequency is 273 Hz. Experimental results also shows that as a given driving field level, as actuator thickness increases, resonant frequency linearly increases, as shown in Fig. 4. Using the assumption of a homogeneous ceramic plate, the fundamental bending resonant frequency of a cantilever actuator can be expressed as:

$$f_0 = 0.161 \left( \frac{L}{L_0} \right) \frac{1}{\sqrt{d_{31}}}$$

(4)

For ferroelectric ceramics such as PZTs, linear piezoelectric response is observed only under very small signal level. The materials properties provided by manufacturers are all measured under small signal level. However, as driven field amplitude increases, the charge output or strain response due to electric field will increase nonlinearly. Consequently, the effective dielectric constant and piezoelectric coefficients will increases. Our experimental results on various PZT ceramic materials indicate that piezoelectric $d_{31}$ coefficient increases as driving field amplitude. Especially for soft PZT, nonlinearity occurs even at relatively low electric field level which consequently contributes to the nonlinear electric field response of tip displacement of the cantilever shear mode actuators. On the other hand, for cantilever shear mode actuator, one end is mechanically clamped thus can not generate pure shear motion under high driving field. Due to this boundary condition, a lateral force which perpendicular to the actuator surface will be produced when driving under electric field. In the case of thin ceramic plate, mechanical bending will then be easily resulted which enhances the tip displacement of ceramic actuator. The thinner the ceramic plate, which is easier to be bent thus the larger the tip displacement. The induced mechanical bending mode can be further clarified by Fig. 3 which shows frequency response of tip displacement for shear actuator with 0.9 mm thickness under driving field of 100 V/mm. The observed fundamental bending resonant frequency is 273 Hz. Experimental results also shows that as a given driving field level, as actuator thickness increases, resonant frequency linearly increases, as shown in Fig. 4. Using the assumption of a homogeneous ceramic plate, the fundamental bending resonant frequency of a cantilever actuator can be expressed as:

$$f_0 = 0.161 \left( \frac{L}{L_0} \right) \frac{1}{\sqrt{d_{31}}}$$

(4)

For ferroelectric ceramics such as PZTs, linear piezoelectric response is observed only under very small signal level. The materials properties provided by manufacturers are all measured under small signal level. However, as driven field amplitude increases, the charge output or strain response due to electric field will increase nonlinearly. Consequently, the effective dielectric constant and piezoelectric coefficients will increase. Our experimental results on various PZT ceramic materials indicate that piezoelectric $d_{31}$ coefficient increases as driving field amplitude. Especially for soft PZT, nonlinearity occurs even at relatively low electric field level which consequently contributes to the nonlinear electric field response of tip displacement of the cantilever shear mode actuators. On the other hand, for cantilever shear mode actuator, one end is mechanically clamped thus can not generate pure shear motion under high driving field. Due to this boundary condition, a lateral force which perpendicular to the actuator surface will be produced when driving under electric field. In the case of thin ceramic plate, mechanical bending will then be easily resulted which enhances the tip displacement of ceramic actuator. The thinner the ceramic plate, which is easier to be bent thus the larger the tip displacement. The induced mechanical bending mode can be further clarified by Fig. 3 which shows frequency response of tip displacement for shear actuator with 0.9 mm thickness under driving field of 100 V/mm. The observed fundamental bending resonant frequency is 273 Hz. Experimental results also shows that as a given driving field level, as actuator thickness increases, resonant frequency linearly increases, as shown in Fig. 4. Using the assumption of a homogeneous ceramic plate, the fundamental bending resonant frequency of a cantilever actuator can be expressed as:

$$f_0 = 0.161 \left( \frac{L}{L_0} \right) \frac{1}{\sqrt{d_{31}}}$$

(4)
indicating \( f_s \) is proportional to actuator thickness, which is in good agreement with experimental data. By Eq. (4), using the measured resonant frequency, mechanical compliance \( s_{11} \) could be calculated. However, further experimental results (Fig. 5) show that the bending resonant frequency is strongly dependent on driving electric field: as the amplitude of driving field increases, resonant frequency decreases. Therefore mechanical compliance \( s_{11} \) in Eq. (4) is dependent on driving field amplitude. This is due to elastic nonlinearities of piezoelectric ceramics. In Eq. (4) \( s_{11} \) is the linear elastic compliance. When piezoelectric ceramic is driven under high electric field, nonlinear elastic coefficients have to be taken into account.

\[
x_i = s_{ij}T_j + s_{jkl}T_jT_k + s_{ijkl}T_jT_kT_l
\]

The nonlinear elastic coefficients are directly related to and account for the shift of resonant frequency \( f_s \).

Quite a large blocking force can be generated in pure shear mode vibration and this shear force is independent of actuator length (Eq. (3)). However, when induced bending vibration occurs under high driving field level, blocking force will be greatly reduced and can be written as:

\[
F_s = \frac{Yr^2w\delta}{4L^3}
\]

Where \( Y \) is Young's modulus of ceramic actuator, \( \delta \) is tip displacement. The blocking force is now dependent on electric field in a quite complex way since tip displacement is not a linear function of driving field for cantilever shear mode actuator.

It should be noted that in piezoelectric shear mode actuator, large tip displacement can only be achieved when using thin ceramic plate, under high driving field. In some applications which require low displacement level but large forces, two-layer or multilayer shear mode actuator could be used. In these actuators, thin ceramic plates with antiparallel poling direction are bonded together mechanically in series and electrically in parallel. Low driving voltage could be used due to reduced individual layer thickness. Experimental results on two-layer actuators indicate that bending resonant frequency is proportional to the total actuator thickness thus can be shift to higher frequency range by using more layers.

IV. Summary

Characteristics of piezoelectric shear mode actuator have been investigated. Due to both nonlinear piezoelectric response and electric field induced mechanical bending, quite a large tip displacement can be obtained. The resonant frequency of bending vibration is dependent on the driving field because of elastic nonlinearities.

ACKNOWLEDGMENT

This work was supported by the Office of Naval research under the contract N00014-94-14140

Reference

(8) Qing-Ming Wang, L. E. Cross, "Electric Field and Temperature Dependence of Piezoelectric d31 coefficient of PLZT Ceramics", to be published.
APPENDIX 11
A comparative analysis of piezoelectric bending-mode actuators

V. D. Kugel, Sanjay Chandran, and L. E. Cross
Intercollegiate Materials Research Laboratory, The Pennsylvania State University, University Park, PA 16802

ABSTRACT

During the last several years novel piezoelectric bending actuators have been developed: RAINBOW, CERAMBOW, CRESCENT, $d_{31}$ bimorph and THUNDER. A comparative experimental investigation of electromechanical characteristics of these devices along with conventional $d_{31}$ bimorph and unimorph actuators was conducted in this work. All transducers were fabricated from soft piezoelectric ceramics. The experimental results show that $d_{31}$ bimorph and unimorph elements have superior quasi-static characteristics as compared to other type of bending-mode actuators. All these piezoelectric devices demonstrate a significant dependence of electromechanical performance on the magnitude of the driving electric field. It was found that the decrease in the mechanical quality factor and resonant frequency of bending vibrations in $d_{31}$ unimorph, RAINBOW, CRESCENT (CERAMBOW) and THUNDER with increasing electric field is much smaller than that in bimorph and $d_{31}$ unimorph actuators. The dependence of the behavior of these devices on the operating conditions governs the selection of a particular device for a specific application.

Keywords: piezoelectric actuators, bimorph, unimorph, RAINBOW, CRESCENT, CERAMBOW, THUNDER

1. INTRODUCTION

Piezoelectric ceramics have a relatively high electromechanical coupling coefficient which makes them very attractive for applications involving efficient transformation of electrical energy into mechanical energy and vice versa. Therefore piezoelectric actuators fabricated from these ceramics have significant potential for use as sensors and actuators in smart systems. Some of the most popular types of piezoelectric devices are bending actuators, in which applied electric field causes mechanical bending because of the piezoelectric effect. A classical example of such a device is piezoelectric bimorph actuator consisting of two similar piezoelectric plates poled along their thickness and adhesively bonded together. A schematic view of the piezoelectric bimorph cantilever with rectangular cross-section is shown in Fig. 1a. The lateral displacement $u$ under applied voltage $V$ (Fig. 1b) is caused by the piezoelectric effect in the direction perpendicular to the polar $(Z)$ axis (piezoelectric $d_{31}$ coefficient).

![Schematic view of the piezoelectric bimorph cantilever](image)
Another widely used bending actuator is unimorph (Fig. 2).\textsuperscript{2} The unimorph actuator consists of the non-piezoelectric and electrode piezoelectric plates bonded together. Here, like in the piezoelectric bimorph, the piezoelectric effect in the direction perpendicular to the polar axis (coefficient $d_{31}$) generates flexural displacement.

![Fig. 2. A schematic view of the piezoelectric $d_{31}$ unimorph cantilever with rectangular cross-section.](image)

A new type of bending piezoelectric actuator named RAINBOW (standing for Reduced And Internally Biased Oxide Wafer) has recently been developed.\textsuperscript{3} This is a monolithic bender in which the ceramic is subjected to a selective high temperature reduction with graphite in an oxidizing atmosphere resulting in a reduced non-piezoelectric layer with metallic electrical conductivity and an unreduced piezoelectric layer. The resulting stress-biased monolithic unimorph has domelike structure (Fig. 3) because of difference in the thermal contraction of reduced and non-reduced parts of the ceramic plate. It was stated that the actuator can generate significant axial displacement.\textsuperscript{4}

![Fig. 3. A schematic view of RAINBOW actuator.](image)

CERAMBOW (stands for CERAMic Biased Oxide Wafer) piezoelectric actuator is another stress-biased unimorph actuator in which metal and electrode ceramic plates are cemented together at an elevated temperature using appropriate solder.\textsuperscript{3} Curvature develops as the CERAMBOW is cooled to room temperature since the metal and ceramics have different coefficients of the thermal expansion. CERAMBOW has the same shape as RAINBOW (Fig. 4). A much more reliable way for stress-biased actuators fabrication has been suggested by the authors. Dome shape of this stress-biased unimorph named CRESCENT (Fig. 4) is achieved by bonding metal and electrode piezoelectric plates at a high temperature using special epoxies (curing temperature is 200-400 °C). If fabricated at the same temperature, the CERAMBOW and CRESCENT have similar electromechanical properties.

Another new class of bimorph and unimorph actuators utilizing piezoelectric $d_{33}$ coefficient has recently been developed.\textsuperscript{5} This caterpillar-type piezoelectric $d_{33}$ transducer consists of piezoelectric segments bonded by a polymeric agent by a dicing and layering technique (Fig. 5). It is the piezoelectric effect along polar axis $P_{3}$ (coefficient $d_{33}$) that causes flexural displacement in these transducers. Since piezoelectric $d_{33}$ coefficient in commercial piezoelectric ceramics is 2-2.2 times larger than $d_{31}$, this transducer generates significantly higher displacement than conventional piezoelectric $d_{31}$ bimorph and unimorph actuators.
Recently, another type of unimorph stress-biased piezoelectric actuator - THUNDER - (stands for THin layer Unimorph Driver and sensor) has been reported. The technique of THUNDER fabrication consists of high temperature bonding (300-350 °C) of an elecroded ceramic plate with metal foils using LARC®-SI polyimide adhesive developed at NASA. The foils are cemented from both surfaces of the ceramic plate and the thickness of the metal foils on one surface of the ceramic plate is much thicker than on the other (Fig. 6). After the high temperature bonding the structure is given an additional beat by mechanical pressing to increase curvature. No published data are available to date for evaluating electromechanical properties of THUNDER.

Thus, several new types of piezoelectric bending actuators have emerged during the last several years. A comparative analysis of their electromechanical properties is indeed the need of the hour. Therefore this work was undertaken in attempt to find and use unified criteria for the device characterization. The approach used is described in
section 2. Experimental procedure and results are presented in section 3. Finally, discussion and summary are given in section 4.

2. REPRESENTATION OF EXPERIMENTAL DATA

Actuators consisting of plates with rectangular cross-section have only been considered in this work. Since in many instances piezoelectric bending actuators are used in the cantilever configuration where one end of the actuator is clamped and the second one vibrates under or without an external load (Fig. 1), this configuration was used for the actuator characterization.

In the cantilever configuration mechanical load is usually applied to the vibrating end of the cantilever. Therefore the most important quasistatic characteristics are the free displacement \( \eta \) (Fig. 1) of the vibrating end and the blocking force \( F_0 \) when \( \eta = 0 \). For piezoelectric bimorph and unimorph actuators⁴

\[
\eta = \frac{3}{2} \frac{d_3^1 l^2}{t} k_x E,
\]

\[
F_0 = \frac{3}{8} \frac{d_3^1 w t^4}{s} k_x E,
\]

where \( d \) is the piezoelectric coefficient of ceramics, \( s \) is the mechanical compliance of ceramics in the direction \( X \) under the constant electric field \( E \) (Fig. 1), \( E = U/\ell \), for all actuators except for \( d_3^1 \) bimorph and unimorph for which \( E = U/\ell \) (Fig. 5), \( L, W, \) and \( t \) are the dimensions of the piezoelectric plate of the cantilevers (Figs. 1 and 2), and \( k_x \) and \( k_m \) are displacement and blocking force coefficients, respectively. For conventional piezoelectric \( d_3^1 \) bimorph and unimorph actuators (Figs. 1 and 2) piezoelectric coefficient \( d_3^1 \) and mechanical compliance \( s_m^0 \) should be used in equation (1); for \( d_3^2 \) transducers \( d_3^2 \) and \( s_m^2 \) should be used.

For bimorph cantilevers displacement and blocking force coefficients in equation (1) are equal to 1, for unimorph cantilevers without internal stress bias (Figs. 2 and 5) they can be expressed as

\[
k_1 = \frac{2\gamma (1+\nu)}{1+4\nu+5\gamma^2(\gamma+4\nu^2+\nu^2)} = \frac{x}{t}, \quad k_2 = \frac{y}{t},
\]

\[
k_m = 2\gamma \frac{1+\nu}{1+\nu^2},
\]

where \( \nu \) is the thickness of the non-piezoelectric plate (Fig. 2) and \( Y_m \) is the Young's modulus of the non-piezoelectric plate. As follows from the analysis of equation (2) factors \( k_1 \) and \( k_m \) depend on the ratio of thicknesses \( \nu \) and Young's moduli \( y \) of non-piezoelectric and piezoelectric plates. The larger \( \nu \) is, the higher the value of \( k_1 \) and \( k_m \) for optimum \( x \). Analysis shows that maximum value of \( k_1 \) is 0.5 and corresponding \( k_m \) is 2.

For stress-biased unimorph actuators like RAINBOW, CERAMIC, and CRESSENt, mechanical stress arising in these structures during device fabrication and poling, changes electromechanical properties of ceramics. Therefore equation (1) for these actuators can be written as

\[
\eta = \frac{3}{2} \frac{d_3^1 l^2}{t} k_x k_x E,
\]

\[
F_0 = \frac{3}{8} \frac{d_3^1 w t^4}{s} k_x k_m E,
\]

where \( d_3^1 \) and \( s_m^0 \) are the piezoelectric and mechanical compliance coefficients of the piezoelectric ceramics before the device fabricating, \( k_x \) and \( k_m \) are coefficients equal to relative change in \( d_3^1 \) and \( d_3^2 / s_m^0 \) respectively as a result of the acting mechanical stress. Coefficients \( k_1 \) and \( k_2 \) in equation (3) should be calculated using equation (2) for values of piezoelectric and electromechanical coefficients of the piezoelectric plate poled under mechanical stress. Equation (3) is also valid for THUNDER, but in this case \( k_1 \) and \( k_2 \) are not described by equation (2).

One more important quasistatic electromechanical characteristic of the actuators is their electrical admittance \( Y \):

\[
Y = \frac{j \omega \ell w}{t} \frac{d_3^1}{s} k_x k_m E.
\]
where \( \omega \) is the angular frequency, \( \varepsilon_{ij} \) is the component of the tensor of the dielectric permittivity of ceramics and \( k_y \) is the factor depending on the corresponding mechanical coupling coefficient and a change in the dielectric permittivity as a result of the devices fabricating.

Thus as is seen from equations (1-4) quasistatic electromechanical characteristics of actuators with the same dimensions of piezoelectric part of the devices, the same electric field, and the same frequency can be normalized with respect to actuator chosen as a standard. It is convenient choose \( d_{31} \) bimorph actuator (Fig. 1) as this standard. Therefore for any given actuators can be expressed as

\[
\eta = \eta_0 \alpha \varepsilon_{ij} f_{\alpha}^i f_{\alpha}^j \]

\[
F_a = F_a_0 \varepsilon_{ij} f_{\alpha}^i f_{\alpha}^j \]

\[
Y = Y_0 \varepsilon_{ij} f_{\alpha}^i f_{\alpha}^j \]

(5)

where \( f_{\alpha}^i \), \( f_{\alpha}^j \), \( f_{\alpha}^k \) are individual figures of merit characterizing the tip displacement, blocking force and electrical admittance of a specific bending actuator, respectively.

The mechanical work \( W_m \) that can be produced by the bending-mode actuators is proportional to the product of the multiplication of the tip displacement \( \eta \) and the blocking force:

\[
W_m = \eta F_a
\]

Quasistatic electrical energy \( W_d \) accumulated in the transducer is proportional to its dielectric permittivity \( \varepsilon \), the volume of the piezoelectric plate and the square of electric field \( E \). Using equation (4), the following relation is obtained:

\[
W_d = \frac{\varepsilon_0 \varepsilon E^2}{\varepsilon_0}
\]

(7)

The ratio of the mechanical work to the input electrical energy can serve as an overall figure of merit for the electromechanical efficiency of the piezoelectric actuator,

\[
\text{integral figure of merit} = \frac{W_m}{W_d} = \frac{\eta F_a}{\varepsilon_0 \varepsilon E^2}
\]

(3)

Using equation (5) the overall figure of merit relative to that of piezoelectric \( d_{31} \) bimorph, \( f_\alpha \), can be written as

\[
f_\alpha = \frac{\eta_0 \alpha \varepsilon_{ij} f_{\alpha}^i f_{\alpha}^j}{f_{\alpha}^i f_{\alpha}^j}
\]

(9)

Thus, the relative factors \( f_{\alpha}^i \), \( f_{\alpha}^j \), \( f_{\alpha}^k \), and \( f_\alpha \) make it possible to describe quasistatic electromechanical characteristics of bending-mode piezoelectric actuators in the cantilever configuration. It is important to note that the actuators under consideration should be fabricated using the same piezoelectric ceramics and should have the same dimensions of the active piezoelectric part. All quasi-static characteristics should be measured for the same amplitude and frequency of the applied electric field since electromechanical properties of piezoelectric ceramics depend on the amplitude and frequency of the electric field. Experimental results show that despite a significant dependence of quasistatic electromechanical characteristics of piezoelectric actuators fabricated from soft piezoelectric ceramics on the applied electric field, the changes of these characteristics for different actuators show almost the same variation. Therefore it is enough to find \( f_{\alpha}^i \), \( f_{\alpha}^j \), \( f_{\alpha}^k \), and \( f_\alpha \) for one value of the applied electric field.

Since in many cases bending-mode actuators are operated near the fundamental frequency of bending vibrations, resonant characteristics of the actuators are also important. The values of the fundamental resonant frequency \( \nu_r \) and the mechanical quality factor \( Q_m \) can be chosen to characterize resonant properties. The fundamental resonant frequency of the piezoelectric \( d_{31} \) bimorph cantilever with rectangular cross-section is

\[
\nu_r = \frac{1875}{16\pi^2} \frac{1}{l^2} \frac{1}{\sqrt{3d_{31}}} \frac{1}{\rho_s}
\]

(10)

where \( \rho_s \) is the density of the ceramics. The resonant frequency of the piezoelectric \( d_{31} \) unimorph with rectangular cross-section is expressed as

\[
\nu_r = \frac{1875}{16\pi^2} \frac{1}{l^2} \frac{1}{\sqrt{3d_{31}}} \frac{1}{\rho_s} \frac{1}{\sqrt{k_z(1+\varepsilon)}}
\]

(11)

where \( \rho_s \) is the density of the non-piezoelectric plate. Thus, it is evident that the resonant frequency of the unimorph cantilever is a function of the resonant frequency of the piezoelectric plate (bimorph). For calculating the fundamental
resonant frequency of piezoelectric d_{33} bimorph and unimorph, \( s_3^e \) should be used in equations (10) and (11) instead of \( s_3^e \). Equation (11) is also valid for RAINBOW, CERAMBOW, and CRESCENT cantilevers but in this case, the mechanical compliance of the stress-biased piezoelectric plate should be used in the calculations. For THUNDER too, the resonant frequency is a function of the resonant frequency of piezoelectric plate which is described by equation (10) but the actual dependence is a more complicated function of \( x, y \) and \( z \). Based on the analysis described above, the resonant frequency of the bending mode actuators can be related to the resonant frequency of the piezoelectric \( d_{33} \) bimorph which has the same dimensions of the active piezoelectric plate, using the equation

\[ v_r = \frac{v_{res}}{f_n^*}, \]  

(12)

where \( f_n^* \) is the figure of merit characterizing the fundamental resonant frequency of the bending-mode transducer. It is known\(^5\) that the fundamental resonant frequency depends on the magnitude of the applied electric field. Therefore another important parameter to be considered is the relative change in the resonant frequency \( \Delta v_r/v_r \), as a function of the electric field.

The mechanical quality factor \( Q_m \) is another important resonant characteristic. Since mechanical vibrations of the piezoelectric bending-mode actuators are described by fourth-order differential equation\(^3\) and not by a second order one, \( Q_m \) should be defined appropriately. By analogy with the definition of \( Q_m \) for damped harmonic vibrator without frequency dispersion of the relevant electromechanical parameters of the system, \( Q_m \) can be expressed as\(^11\)

\[ Q_m = \frac{n_r}{n}, \]  

(13)

where \( n_r \) is the amplitude of resonant vibrations. The amplitude of resonant vibrations depends on mechanical losses in the actuator\(^6\) and is also very sensitive to the way the cantilever is clamped. Therefore it is preferable to compare the relative change \( \Delta Q_m/Q_m \) as a function of the electric field instead of \( Q_m \).

Thus, relative factors \( f_n^*, f_n^1, f_n^2, \) and \( f_n \) are used to characterize quasistatic electromechanical characteristics and \( f_n^*, \Delta v_r/v_r, \) and \( \Delta Q_m/Q_m \) are used to characterize resonant properties of bending-mode piezoelectric actuators in the cantilever configuration.

3. EXPERIMENTAL PROCEDURE AND EXPERIMENTAL RESULTS

All transducers investigated were fabricated from soft piezoelectric ceramics and had a rectangular cross-section and the following dimensions: 0.4-2.5 mm in thickness, 5-15 mm in width, and 15-35 mm in length. Piezoelectric bimorph and metal/piezoelectric unimorph actuators were fabricated from PKI550 (Piezo Kinetic, Inc.) ceramic plates poled along their thickness. This category of piezoelectric ceramics is analoguous to soft piezoelectric ceramics PZT. Stainless steel SS302 was used to make the unimorphs because of its very high Young's modulus and, consequently, high theoretical ratio of Young's moduli \( y=1.05 \) (equation (2)). The plates were bonded using commercial J-B Weld epoxy (J-B Weld Company). Piezoelectric d_{33} bimorphs were fabricated by a dicing and layering technique.\(^6\) Theoretical ratio of Young's moduli for the d_{33} unimorph is \( y=3.90 \). The ceramic plates in the stack were bonded using commercial conductive adhesives EP21TDCS (Master Bond, Inc.) and E-Solder 5025 (Insulating Materials, Inc.). J-B Weld epoxy was used for bonding metal and sliced ceramic plates. Each piezoelectric segment in the piezoelectric plates (Fig. 5) had the following dimensions: \( t_1 = t_2 = 1.09 \) mm, \( w = 11 \) mm. As follows from equation (2), electromechanical characteristics of \( d_{31} \) and \( d_{33} \) unimorph actuators depend on the ratio of thicknesses \( x \) and Young's moduli \( y \) of non-piezoelectric and piezoelectric plates. Theoretical analysis shows\(^9\) that maximum value of the displacement factor \( f_n^* \) and overall figure of merit \( f_n \) correspond to different values of \( x \). Therefore, only devices exhibiting maximum quasistatic tip displacement were chosen. Experimental study showed\(^4\) that for SS302/PKI550 unimorphs optimum \( x \) lies between 0.2 and 0.35.

CRESCENTs were fabricated from PKI550 and SS302 using several types of high temperature epoxies. Since the Cure temperature ~200 °C was lower than the device fabrication temperature, the actuators were poled after their fabrication. It was found that tip displacement factor \( f_n^* \) for CRESCENT depends not only on the \( x \) and \( y \) factors but also on the curing temperature \( T_c \), therefore only CRESCENTs possessing maximum \( f_n^* \) were used for a comparative study. For actuators with \( x=1 \) mm and \( y=0.37 \) mm the optimal curing temperature was around 250-260 °C. The radius of the curvature of the transducer before poling was 0.4 m and after poling it increased to 0.8-0.9 m.
Since figures of merit are defined for elements with the same dimension of the piezoelectric part, the experimental data obtained for actuators fabricated from PK1550 with different dimensions were recalculated for a device with standard dimensions of piezoelectric plate.

RAINBOW actuators were cut from piezoelectric RAINBOW disks which were purchased from Aura Ceramics, Inc. RAINBOW disks are fabricated from C3900 ceramics which is analogous to PZT5H. The thickness of the device was 0.46-0.48 mm, the thickness of the piezoelectric (unreduced) part \( t_e \) was approximately 0.27-0.29 mm. The thickness of the reduced layer was 0.12 mm and the thickness of conductive epoxy layer which served as electrode was 0.07 mm.

THUNDER actuators were fabricated from soft piezoelectric ceramics PZT5A and Al foil. Three layers of the foil were cemented on one side of the ceramic plate and one layer on the other side. Curing temperature was 300-320 °C. The thickness of the THUNDERs was 0.41 mm, the thickness of piezoelectric plates was 0.2 mm. The radius of the actuator curvatures after poling was approximately 0.33 m. After the high temperatures bonding followed by poling (first stage) the devices were additionally bent by mechanical pressing (second stage) and the radius of the curvature decreased to 0.14 m. Measurements of piezoelectric characteristics were conducted after the first and second stages.

Individual figures of merit of RAINBOW and THUNDER actuators were calculated using their experimental data and theoretical calculations for bimorph actuators from the same piezoelectric ceramics having the same dimensions.

The tip displacement of actuators was measured by a photonic sensor MII 2000 (MTI Instruments Division). The blocking force was measured by means of a load cell ELF-TC500 (Entran Devices, Inc) and the electrical admittance was measured by means of a lock-in amplifier SR830 DSP (Stanford Research Systems, Inc). A complete description of the experimental set-up is given elsewhere. Electromechanical characteristics were measured in a quasistatic regime and at the fundamental frequency of bending vibrations. In the quasistatic regime the measurement frequency was at least ten times smaller than the fundamental resonant frequency.

Experimental figures of merit \( f_d, f_b, f_a, f_m \) representing electromechanical characteristics of the actuators studied are given in the Table. These values were obtained for low applied electric field (less than 20 V/cm). Dependencies of the resonant frequency and mechanical quality factors on the electric field are shown in Figs. 7 and 8, respectively. Resonant characteristics of CRESCENT actuators were similar to that of \( d_{33} \) unimorphs. No significant difference in almost all measured electromechanical properties of THUNDER actuators after the first and second stages of their fabrication was observed. Only mechanical quality factor decreased by 23% after the second stage.

Table. Figures of merit of bending-mode piezoelectric actuators in the cantilever configuration.

<table>
<thead>
<tr>
<th>TYPE OF PIEZOELEMENT</th>
<th>Tip displacement factor ( f_d )</th>
<th>Blocking force factor ( f_b )</th>
<th>Admittance factor ( f_a )</th>
<th>Overall figure of merit ( f_m )</th>
<th>Resonant frequency factor ( f_m )</th>
</tr>
</thead>
<tbody>
<tr>
<td>( d_{31} ) Bimorph</td>
<td>1</td>
<td>1</td>
<td>1</td>
<td>1</td>
<td>1</td>
</tr>
<tr>
<td>( d_{31} ) Unimorph, ( x=0.34, \ y=3.05 )</td>
<td>0.41</td>
<td>1.3</td>
<td>1.0</td>
<td>0.74</td>
<td>1.7</td>
</tr>
<tr>
<td>RAINBOW</td>
<td>0.19-0.22</td>
<td>0.1-1.2</td>
<td>0.66</td>
<td>0.03-0.40</td>
<td>1.2-1.4</td>
</tr>
<tr>
<td>CRESCENT ( 250 ) °C (CERANOVID) ( x=0.34, \ y=3.05 )</td>
<td>0.44</td>
<td>1.75</td>
<td>0.91</td>
<td>0.83</td>
<td>1.7</td>
</tr>
<tr>
<td>THUNDER ( (\text{GAP/PZT5A/Al}) ) ( d_{33} ) Bimorph</td>
<td>0.12</td>
<td>0.36-1.0</td>
<td>0.90</td>
<td>0.05-0.13</td>
<td>2.1</td>
</tr>
<tr>
<td>( d_{33} ) Unimorph, ( x=0.34, \ y=3.90 )</td>
<td>2.5</td>
<td>1.52</td>
<td>1</td>
<td>3.80</td>
<td>0.84</td>
</tr>
</tbody>
</table>

Analysis of \( d_{31} \) bimorphs experimental data showed that the experimental tip displacement and its theoretical value calculated according to equation (1) are in good agreement. Averaged experimental value of the blocking force was 33% less than theoretical one given by equation (1) and the averaged experimental value of the resonant frequency of cantilevers was 13% smaller than theoretical one (equation (10)). Nevertheless, the resonant frequency of these bimorph actuators with free-free boundary conditions (non of the ends is clamped) coincided with the theoretical one. For \( d_{33} \) unimorph actuators
the same tendency was observed: the blocking force was smaller by 13% and the resonant frequency was smaller by 12% than corresponding theoretical values. In $d_{31}$ bimorphs, the blocking force was smaller by 42% and the resonant frequency was smaller by 17% than corresponding theoretical values. The discrepancy was smaller for $d_{33}$ unimorphs: the blocking force was smaller by 7% and the resonant frequency was smaller than the corresponding theoretical values by 3%. There may be several reasons for these discrepancies. Firstly, under the applied electric field, piezoelectric cantilever bends not only along the $X$ axis but also along the $Y$ axis (Fig. 1). In theoretical calculations, bending along the $Y$ axis was neglected. Bending along the $Y$ axis may affect the blocking force and resonant frequency of actuators. The second reason is that the cementing epoxy whose thickness was neglected in the calculations may also change electromechanical properties of the actuators. We were unable to conduct the same analysis for RAINBOW, CRESCENT, and THUNDER actuators since exact electromechanical properties of these devices are unknown.

As seen from the Table, the blocking force factor for RAINBOW does not have an exact value. This is because the blocking force showed significant dependence on the external load. The force increased markedly with increasing mechanical prestress which can be generated externally by the horizontal displacement of the load cell stuck to the vibrating end of the actuator. It was also found that if the load cell was pressed against the vibrating end, the measured blocking force was an order of magnitude larger than that for the case when the load cell was glued to the vibrating end of RAINBOW cantilevers. A significant scattering in the measured blocking force factor $f_b$ of THUNDERS (see Table) is probably caused by experimental limitations since it is very difficult to attach the vibrating end of THUNDER having a curved shape and the load cell head having a flat surface.

![Graph](image1)

**Fig. 7.** Dependence of the resonant frequency $v_r$ of bending vibrations on the electric field (rms). Low-field resonant frequency is: 1394 Hz ($d_{31}$ bimorph), 1015 Hz ($d_{31}$ unimorph), 1029 Hz ($d_{33}$ unimorph), 595 (RAINBOW), and 227 Hz (THUNDER).

![Graph](image2)

**Fig. 8.** Dependence of the relative mechanical quality factor on the electric field (rms). Low-field quality factor $Q_m$ is: 55 ($d_{31}$ bimorph), 48 ($d_{31}$ unimorph), 25 ($d_{33}$ unimorph), 62 (RAINBOW), and 137 (THUNDER, after first stage).
4. DISCUSSION AND SUMMARY

As follows from the Table, $d_{31}$ bimorph and unimorph actuators have the best quasi-statuistic electromechanical characteristics with respect to the blocking force and overall figure of merit. The reason for this is that piezoelectric $d_{31}$ coefficient and corresponding coupling factor $k_{31}$ are 2-2.2 times larger than $d_{31}$ and $k_{31}$. $d_{31}$ bimorph generates higher tip displacement than $d_{31}$ unimorph but its blocking force is significantly lower. $d_{31}$ bimorph is followed by CRESCENT (CERAMBOW) and $d_{31}$ unimorph actuators. It is interesting that the CRESCENT fabricated at the optimal temperature has a higher tip displacement, and overall figure of merit than $d_{31}$ unimorph fabricated from the same materials. It means that the average piezoelectric $d_{31}$ coefficient of the piezoelectric plate poled under certain mechanical bending stress is higher and corresponding dielectric permittivity $e_r$ lower than that of the starting material. A probable reason is that there are specific domain structures that are formed during poling. Two experimental facts support this hypothesis; the first one is a significant increase in the radius of the transducer curvature after poling. The second fact is that after separation of the metal and ceramic plates in the poled CRESCENT, the ceramic plate retained its curved shape which implies that there is no mechanical stress in the transducer. Moreover, effect of the longitudinal stress cannot explain the increase in the tip displacement since experimental results show that longitudinal stress decreases piezoelectric $d_{31}$ coefficient. If the CRESCENT is prepared above the optimal temperature residual mechanical stress may decrease piezoelectric $d_{31}$ coefficient.

As is seen from the Table, RAINBOW and THUNDER actuators have lowest quasi-statuistic figures of merit. Since reduced layer in RAINBOW actuator has a Young's modulus much lower than stainless steel used for $d_{31}$ unimorph fabrication, displacement factor $k_3$ for this actuator is less than that for unimorphs. In addition as follows from the analysis of equation (2), the ratio of thickness of the reduced and active piezoelectric layers is less than the optimal one. The same reason can explain inferior quasi-static behavior of THUNDERs. Also, these devices have metal foils from both sides that decrease tip displacement. Since the thickness of the piezoelectric plates in the device was relatively small, the adhesive layers can also decrease $f_m$.

As seen from Fig. 7, the fundamental resonant frequency of bending-mode actuators depends on the applied electric field. Large sensitivity of the resonant frequency of $d_{31}$ unimorph to electric field can be due to the fact that at a high level of mechanical stress which exists at a high level of resonant vibrations the epoxy bonding in piezoelectric segments becomes soft due to non-linear strain-stress relationship in the polymer materials. Resonant properties of $d_{31}$ bimorph were not measured but based on the above discussion, this device should be even more sensitive to high electric field because it does not have the stabilizing metal plate. Relatively high dependence of $v_f$ of $d_{31}$ bimorph can be caused by an increase of the mechanical compliance of the piezoelectric ceramics with increasing electric field. The effect of mechanical "softening" of the piezoelectric ceramics in all $d_{31}$-type unimorph devices is less significant since they have non-piezoelectric part whose properties do not depend on the electric field.

All actuators demonstrate a significant decrease of the mechanical quality factor with increasing electric field (Fig. 8). The most sensitive device studied was the $d_{31}$ bimorph. Since the amplitude of resonant vibrations is inversely proportional to the mechanical losses at the resonance, the mechanical losses increase significantly under high electric field. At the electric field 1 kV/cm the mechanical quality factor decreases by an order of magnitude compared to its low-field value. Since in unimorph-type structures there is non-piezoelectric part in which the mechanical losses do not depend on the applied electric field, the decrease of $Q_e$ in these actuators is more gradual.

It should be noted that there is one more important figure of merit that has not been considered in this work. It is mechanical failure at resonance. Our results showed that at a high level of mechanical vibrations at resonance the actuators fracture. The fracture occurred at the surface of ceramic plates in the region where actuators were clamped since this area is subjected to the highest level of stress. Analysis of experimental data showed that fracture of $d_{31}$ bimorph and unimorph transducers at resonance occurs if the maximum stress at the clamped surface reaches 30-50 MPa. In this case mechanical failure occurs in several seconds. Clearly, unimorph actuators having metal plates, such as $d_{31}$ and $d_{31}$ unimorph, CRESCENT, CERAMBOW, and especially THUNDER, are more reliable in a sense that even if mechanical failure of ceramics occurs the actuators do not fracture since metals like steel or Al have much higher fracture toughness than ceramics.

It is worthwhile to note that the straight or slightly curved shape of bending-mode actuators is not optimal in terms of overall figure of merit $f_m$. For instance, theoretical calculations show that L-shaped $d_{31}$ bimorph cantilever (Fig. 9) has higher blocking force factor $f_m'$ and overall figure of merit $f_m$ than straight $d_{31}$ bimorph with the same dimensions. This is because in a conventional straight structure bending moment generated in the actuator works against the blocking force...
applied to the vibrating end. Therefore, this force blocks the movement of the actuator's tip only. In the L-shaped structure, application of the horizontal force produces a mechanical moment on the horizontal part of the actuator. Thus, the blocking force should almost prevent displacement in the whole bottom part of the transducer; consequently, the magnitude of the corresponding blocking force should be higher in this case.

<table>
<thead>
<tr>
<th>ELEMENT</th>
<th>Tip displacement factor</th>
<th>Blocking force factor</th>
<th>Admittance factor</th>
<th>Overall figure of merit</th>
<th>Resonant frequency factor</th>
</tr>
</thead>
<tbody>
<tr>
<td>$d_{31}$ Bimorph</td>
<td>$f_0$</td>
<td>$f_0$</td>
<td>$f_0$</td>
<td>$1$</td>
<td>$1$</td>
</tr>
<tr>
<td>L-shaped $d_{31}$ Bimorph</td>
<td>$0.75$</td>
<td>$1.5$</td>
<td>$1.0$</td>
<td>$1.125$</td>
<td>$1.32$</td>
</tr>
</tbody>
</table>

Fig. 9. Schematic view of L-shaped $d_{31}$ bimorph and theoretical figures of merit of this device with $l = l_o = 0.5l$.

In summary, a comparative experimental investigation of electromechanical characteristics of piezoelectric $d_{31}$ and $d_{33}$ bimorph and unimorph actuators, RAINBOW, CRESCENT (CERAMIC), and THUNDER actuators in cantilever configuration has been conducted. The tip displacement, blocking force, and electrical admittance were chosen to characterize quasi-static properties and the resonant frequency and mechanical quality factor were chosen to characterize the behavior at the fundamental frequency of bending vibrations. The experimental results show that $d_{31}$ bimorph and unimorph actuators have superior quasi-static characteristics as compared to other types of bending-mode actuators. It was found that resonant frequency and especially mechanical quality factor of all actuators depend on the applied electric field. $d_{31}$ unimorph, RAINBOW, CRESCENT (CERAMIC), and THUNDER were found to be less dependent on the applied electric field than $d_{33}$ bimorph and $d_{33}$ bimorph and unimorph actuators. These results indicate that the choice of devices for a particular application depends on conditions under which the device will operate.

ACKNOWLEDGMENTS

The authors would like to thank R. L. Fox, R. G. Bryant, and J. M. Bacon for supplying THUNDER samples. This work was supported by the Office of Naval Research under the contract N00014–94–1–1140.

REFERENCES

7. R. L. Fox, R. G. Bryant, and J. M. Bacon “THUNDER ACTUATOR” (NASA Langley Research Center), private communication, 1996.
Fig. 4. Amplitude and phase characteristics of RAINBOW cantilever.

Fig. 5. Amplitude and phase characteristics of shear-mode cantilever.

Fig. 6. Dependence of the resonant frequency $v$ of bending vibrations on electric field. Low-field resonant frequency is 1394 Hz (bimorph), 1015 Hz (unimorph), 595 (RAINBOW), and 286 Hz (shear-mode).

Fig. 7. Dependence of the ratio of amplitudes at resonant and low frequencies on electric field. Low-field quality factor $Q_L$ is 55 (bimorph), 48 (unimorph), 62 (RAINBOW), and 31 (shear-mode).

weak dependence of $v$, and $Q_L$ in shear-mode actuator can be related to the lack of the electromechanical coupling for the bending mode and to relatively low level of resonant vibrations. We assume that bending vibrations in the transducer appear as a result of elastic instability [10] of pure shear vibrations because of the action of inertia forces.

In summary, we have studied piezoelectric properties of bimorph, unimorph, RAINBOW, and shear-mode actuators in a wide electric field and frequency range. An appropriate experimental measurement system has been developed. The quasi-static and resonant piezoelectric properties of these structures have been found to be highly dependent on the magnitude of driving field.

ACKNOWLEDGMENT

This work was supported by the Office of Naval Research under the contract N00014-94-1-1140.

REFERENCES


Behavior of Piezoelectric Actuators under High Electric Field

V. D. Kugel, Q. M. Zhang, Baomin Xu, Qing-ming Wang, Sanjay Chandran, and L. E. Cross

187 Materials Research Laboratory, The Pennsylvania State University, University Park, PA 16802

Abstract—Behavior of piezoelectric actuators (bimorph, unimorph, RAINBOW, and shear-mode) fabricated from soft ceramics has been investigated in a wide electric field and frequency range. The electrical admittance, mechanical displacement, and blocking force of these transducers have been found to be highly dependent on the magnitude of driving field. The resonant frequency and mechanical quality factor of bending vibrations for all but shear-mode actuators decreases significantly with increasing driving field. Analysis shows that despite a large variation in the quasi-static electrical admittance and reduced tip displacement of bimorph, unimorph, and RAINBOW cantilevers with driving field, their ratio is almost a constant, which characterizes the ratio of dielectric permittivity \( \varepsilon_{33} \) to piezoelectric coefficient \( d_{31} \).

I. INTRODUCTION

Studies of last few years have demonstrated that piezoelectric transducers have significant potential as actuators for acoustic noise control [1]. These transducers are operated at such high power levels that the ferroelectric ceramic used in them begins to exhibit nonlinear behavior. It is known that piezoelectricity-related properties of ceramic materials are highly dependent on the level of driving electric field and mechanical vibrations [2-6]. A common feature in this behavior of piezoelectric ceramics is an increase in dielectric and piezoelectric coefficients, and losses even at electric fields much lower than the coercive one. It is believed that the nonlinear behavior has extrinsic nature, i.e., is related to the domain wall and interphase boundaries motion [4]. Till now, most of the studies were focused on the investigation of materials properties. The purpose of this work was to study the behavior of the potential piezoelectric transducers such as bimorph, unimorph, RAINBOW, and shear-mode vibrators under high electric field.

II. EXPERIMENTAL PROCEDURE

All transducers investigated had a rectangular cross-section and the following dimensions: 0.45-2 mm in thickness, 5-15 mm in width, and 15-35 mm in length. Piezoelectric bimorph and metal/piezoelectric unimorph actuators were fabricated from PKI550 (Piezo Kinetic, Inc.) ceramic plates poled along their thickness. This category of piezoelectric ceramics is analogous to "soft" PZT5H ceramics. Stainless steel SS302 was used to make the unimorphs. The plates were bonded using commercial J-B Weld epoxy (J-B Weld Company). Rainbow transducers were cut from piezoelectric RAINBOW disks which were purchased from Aura Ceramics, Inc. Shear-mode actuators were fabricated from 3203HD ceramic plates (Motorola). The plates were poled along their length and the driving electric field was applied across the thickness of the plates. Clearly, the flexural displacement generated in the bimorph, unimorph and RAINBOW actuators is caused by piezoelectric \( d_{31} \) coefficient while linear displacement in the shear-mode actuators is caused by piezoelectric \( d_{31} \) coefficient.

To characterize these transducers, their electromechanical properties as a function of the driving electric field well below and close to the fundamental frequency of bending vibrations were investigated. The following parameters of these transducers in the cantilever configuration were measured: i) displacement \( \eta \) of the free end; ii) blocking force \( F_{bl} (\eta = 0) \), and iii) electrical admittance as a function of the applied electric field and frequency.

A block diagram of the experimental set-up is shown in Fig. 1. The tip displacement of piezoelectric cantilevers was measured by a photonic sensor MTI 2000 (MTI Instruments). The measuring head of the sensor was mounted on a manual micropositioner which provided the linear displacement and rotation for adjusting height and angle of the head against the measured transducer. The transducer (in the Figure, bimorph is shown as an example) was mounted on XYZ micropositioner (Ealing Electro-Optics, Inc.). To measure the blocking force, a special metal head of a load cell ELF-TC500 (Entran Devices, Inc.) was glued by Super Glue to the vibration end of the transducer. The load cell was mounted on a micropositioner which provided a horizontal displacement for adjusting the position of the load cell against the measured transducer. The load cell was driven by power supply PS-15 (Entran Devices, Inc.). The electrical admittance was measured by means of a small (several ohm) resistor connected in series with the transducer. All lock-in amplifiers (SR830 DSP, Stanford...
Research Systems, Inc) were synchronized with the output voltage of the power amplifier (790 Series, PCB Piezotronics, Inc or PA-250H, Julie Research Laboratories, Inc). The input AC signal to the power amplifier was supplied by a generator DS345 (Stanford Research Systems, Inc). The developed experimental set-up made it possible to measure the mechanical displacement and electrical admittance in the frequency range of 0-20 kHz and the blocking force in the frequency range from DC to several kilohertz. The maximum driving voltage was 300 Volt RMS.

III. REPRESENTATION OF EXPERIMENTAL DATA

The quasi-static tip displacement \( \eta \) and corresponding blocking force \( F_m \) of unimorph actuators can be written as [7]

\[
\eta = \frac{3}{2} \frac{d_{31}}{t_e} \frac{l^2}{E} k_x, \quad k_x = \frac{2\pi(1+x)}{1+4\pi x-6x^2+4\pi x^2+y+x^2y},
\]

\[ x = \frac{t_0}{t_e}, \quad y = Y_{\text{eff}}, \]

\[ F_m = \frac{3}{8} \frac{d_{31}}{E} \frac{w r^2}{l} k_y, \quad k_y = \frac{2\pi}{1+xy}.
\]

(1)

where \( l \) and \( w \) are the length and width of the cantilever, correspondingly; \( t_e \) and \( t_m \) are the thickness of the ceramic and metal plates, correspondingly; \( E \) is the driving electric field, \( Y_{\text{eff}} \) is the Young's modulus of the metal plate, and \( Y_{\text{eff}} \) is the mechanical compliance of ceramics in the direction perpendicular to the polar axis. For bimorph cantilevers the coefficients \( k_x \) and \( k_y \) in (1) are equal to 1. Corresponding equations for pure shear-mode actuator can be written as:

\[
\eta = d_{33}E,
\]

\[
F_m = \frac{1}{4} \frac{d_{33}}{E} \frac{w r^2}{l^2} \epsilon_y
\]

(2)

where \( d_{33} \) is the component of the mechanical compliance of ceramics in the direction of polar axis. It should be noted that blocking force in shear-mode thin plates causes bending. As follows from (1, 2) reduced amplitudes \( \eta/E \) and \( F_m/E \) are proportional to \( d_{33} \) and \( d_{33}/E \) correspondingly (in the case of bimorph and shear-mode cantilevers \( k_x \) and \( k_y \) are equal to 1). Consequently, a relative change in the ratios \( \eta/E \) and \( F_m/E \) as a function of the driving electric field gives information about the change in the piezoelectric and mechanical properties of the actuators.

The electrical admittance of the transducers at frequencies well below the fundamental bending resonance can be written as

\[
Y = j\omega \frac{S}{t_e} \epsilon_y,
\]

(3)

where \( S \) is the area of the one of the electrodes, \( \epsilon \) is the component of the tensor of the dielectric permittivity (\( \epsilon_{11} \) for bimorphs and unimorphs and \( \epsilon_{11} \) for shear-mode transducers) and \( k_y \) is the coefficient depending the corresponding electromechanical coupling coefficient. For bimorphs this coefficient equals \( 1 - 0.5 d_{31}^2/\epsilon_{31} \epsilon_{11} \), for shear-mode actuators it equals 1. As follows from (3) the relative change in the electrical admittance as a function of the driving electric field gives information about the change in the dielectric permittivity and electromechanical coupling coefficient of the transducer.

To characterize the change in the properties of transducers near fundamental resonance, the following parameters as a function of the driving electric field were measured: i) a change in the resonant frequency \( \Delta \nu \), ii) ratio of vibration amplitudes at resonant and low frequencies, \( \eta/\nu \). The relative change in \( \Delta \nu \) gives an idea of the change in the mechanical compliance and losses at the resonance. The relative change in \( \eta/\nu \) encompasses the change in the mechanical quality factor \( Q_m \), since the amplitude of the damped harmonic vibrations is proportional to this factor [8]

\[
\eta \approx \eta Q_m.
\]

Thus, the relative values of \( \eta, \eta/\nu, F_m/\nu \) and \( \eta/\nu \) as a function of electric field calculated from experimental data were chosen to characterize nonlinear properties of transducers. It should be noted that the lock-in amplifiers used (Fig. 1) made it possible to measure complex values of \( \eta, F_m \), and \( Y, \nu \), i.e., the amplitude and phase characteristics.

IV. EXPERIMENTAL RESULTS

All chosen functions (RMS) were normalized relatively their values at a low electric field (-5-10 V/cm). The dependence of relative values of \( Y, \eta/\nu, F_m/\nu \) on the electric field is shown in Figs. 2-5. Instead of the phase of the electrical admittance the phase of \( \eta/\nu \) was plotted since it is directly related to tangent of the dielectric losses (see (3)). All measurements were done at a frequency at least 10 times lesser than the fundamental frequency of bending vibrations. Maximum electric fields used in these experiments were much less than the coercive field of the ceramics (8-9 kV/cm). As is seen from these graphs, an increase in the driving electric field causes an increase in the amplitude and phase delay of mechanical displacement, electrical admittance, and blocking force. For bimorphs, the dependencies of \( Y \) and \( \eta/\nu \) on the electric field are almost identical and are more pronounced than the dependence of corresponding reduced blocking force \( F_m/\nu \) (Fig. 2). For unimorph cantilevers the effect of the driving electric field is stronger and function \( \eta/\nu \) increases more rapidly at a high electric field than \( Y \) (Fig. 3). A close analysis of experimental data presented in Figs. 2 and 3 shows that
there is a certain threshold electric field (10-50 V/cm) above which measured parameters begin to increase monotonically. As follows from Fig. 4, the dependence of piezoelectric properties of RAINBOW actuators on the electric field is extremely high. Functions $Y$ and $\eta/E$ behave almost in the same manner and, unlike for unimorphs, have convex shape. This type of actuators is characterized by very high losses. It should be noted that the blocking force for RAINBOW cantilevers has been found to be increased markedly with increasing shear force, which can be generated externally by the horizontal displacement of the load cell (Fig. 1). Behavior of $F_\eta/E$ was similar to that of $Y$. Experimental data for shear-mode cantilevers are given in Fig. 5. Clearly, in this case the electrical admittance increases much more rapidly with electric field than the reduced displacement.

Resonant characteristics of transducers are presented in Figs. 6 and 7. All actuators demonstrate a decrease in the resonant frequency of bending vibrations with increasing electric field (Fig. 6). The most pronounced decrease is shown by bimorph cantilevers and the smallest one is shown by shear-mode cantilevers. Measurements of the electrical impedance of the shear-mode vibrator show that there is no resonance in the impedance despite the fact that mechanical resonance of bending vibrations does occur. As is seen from Fig. 7, the relative decrease in the ratio of amplitudes at resonance and low frequency, $\eta/E$ (and, consequently, in $Q_d$), is highly dependent on the type of transducers. A drastic change is obtained with bimorph actuators while a moderate change is observed with shear-mode actuators. It is important to note that the magnitude of the change in $\eta/E$ has been found to depend on its value at low electric field.

V. DISCUSSION AND SUMMARY

The results presented here distinctly show that electromechanical properties of the actuators fabricated from soft ceramics depend on the level of the driving electric field. Data for bimorph transducer (Fig. 2) demonstrate that the electrical admittance $Y$ and reduced tip displacement $\eta/E$ behave in a very similar manner. This functions are proportional to $e_{33}^T(1 - 0.5d_{31}^2 / e_{33}^T)$ and $d_{31}$ correspondingly. Since the factor in parentheses does not change much it means that there is a close relation between electric field dependencies of $e_{33}^T$ and $d_{31}$. Our studies of the material properties of various soft PZT ceramics also demonstrate the same results [9]. As follows from the graph of the reduced blocking force $F_\eta/E = d_{31} / s_{31}$ (Fig. 2), an increase in $s_{31}$ is more moderate than in $e_{33}^T$ and $d_{31}$. This graph also shows the behavior of electromechanical coupling coefficient $d_{31}^2 / e_{33}^T$ because the ratio $d_{31}^2 / e_{33}^T$ is a constant. As follows from the comparison of Figs. 2 and 3 a change in coefficients $k_b$, $k_{bb}$, and $k_c$ with electric field takes place for unimorphs. Data for RAINBOW actuators (Fig. 4) also confirm that there is a close relation between $e_{33}^T$ and $d_{31}$. As follows from Fig. 5 and (2.3), $d_{31}$ increases much more rapidly with the electric field than $e_{33}^T$.

Data of resonant measurements show that bimorph vibrator is more sensitive to the electric field than other transducers. In unimorph actuator there is a non-piezoelectric metal plate which stabilize the behavior of the transducer since its properties do not depend on the electric field. In RAINBOW actuator, in addition to this effect, the internally biased compressive stress may affect resonant behavior. A relatively

![Fig. 2. Amplitude and phase characteristics of bimorph cantilever.](image)

![Fig. 3. Amplitude and phase characteristics of unimorph cantilever with $e_{33}^T = 0.34$.](image)
weak dependence of $v$, and $Q_0$ in shear-mode actuator can be related to the lack of the electromechanical coupling for the bending mode and to relatively low level of resonant vibrations. We assume that bending vibrations in the transducer appear as a result of elastic instability [10] of pure shear vibrations because of the action of inertia forces.

In summary, we have studied piezoelectric properties of bimorph, unimorph, RAINBOW, and shear-mode actuators in a wide electric field and frequency range. An appropriate experimental measurement system has been developed. The quasi-static and resonant piezoelectric properties of these structures have been found to be highly dependent on the magnitude of driving field.

ACKNOWLEDGMENT

This work was supported by the Office of Naval Research under the contract N00014-94-1-1140.

REFERENCES

APPENDIX 13
Behavior of soft piezoelectric ceramics under high sinusoidal electric fields

V. D. Kigel and L. E. Cross
The Materials Research Laboratory, The Pennsylvania State University, University Park, Pennsylvania 16802
(Received 3 July 1997; accepted for publication 2 June 1998)

The behavior of piezoelectric, dielectric, and elastic characteristics of soft piezoelectric lead zirconate titanate (PZT) ceramics was investigated under sinusoidal electric field $E$ applied along the poling direction and under mechanical stress-free conditions for a frequency range 10 Hz–1 kHz. Electrical displacement $D_1$ along the poling direction, mechanical strain $S_1$ in the direction perpendicular to the poling direction, and the resonant frequency $\nu_r$ of electromechanically uncoupled bending vibrations of polarized plates were measured. Commercial ceramics PZT5H, 3203-HD, and PK550 with chemical composition near the morphotropic phase boundary were used in the study. It was found that the amplitude and phase of the first harmonic of the relative strain $S_1/E_\infty$ and of the relative electrical displacement $D_1/E_\infty$ increase similarly with increasing amplitude $E_\infty$ of the electric field if the amplitude is less than coercive field. The corresponding increase in the square of the resonant frequency is more moderate. The dependencies are described well by linear functions except for low electric fields. The functions $S_1/E_\infty$ and $D_1/E_\infty$ demonstrate frequency dispersion as well. Relative amplitude of the second harmonic of the electrical displacement, $D_{2H}/D_{1H}$, which is polar, also increases almost linearly with increasing $E_\infty$. The relative third harmonic $D_{3H}/D_{1H}$ demonstrates saturation-like behavior. By means of the electric pulse technique, it was found that irreversible changes in the remnant polarization take place even at electric fields much smaller that the coercive field. A mathematical model of a hysteretic transducer, describing the electric field dependence of electromechanical properties of soft piezoelectric PZT ceramics, was suggested. According to this approach, the response depends not only on the instantaneous magnitude of the input signal (e.g., electric field) but also on its past extreme values. It was shown that the experimental Rayleigh law is a particular case of the suggested approach. The model relates directly electric field dependencies of complex piezoelectric coefficient $d_{33}(E_\infty)$ and $S_1/E_\infty$ of complex dielectric permittivity $\varepsilon_{33}(E_\infty)$ and $D_{1H}/E_\infty$, and of the amplitude of elastic compliance $s_{11}(E_\infty)$ and $\nu_r$. Application of the model to experimental data showed that the model describes well the first three complex harmonics of $D_1$ and the irreversible change in the remnant polarization. Physical causes of the observed behavior were analyzed. As an alternative to the model based in the $90^\circ$ polarization reorientation and tetragonal/cubic phase boundary motion, a new approach was suggested. In this model, the observed hysteretic changes in the electromechanical properties are assumed to be caused by the electric field dependency of the mechanical stress acting at interdomain boundaries in the partly constrained crystallites of these ceramics. © 1998 American Institute of Physics [S0021-8979(98)05217-6]

I. INTRODUCTION

Piezoelectric ceramics possess excellent electromechanical properties that make them very attractive for application in electromechanical and electroacoustical transducers. One important class of these transducers is high power sound and ultrasound transducers, which are operated under high electric field. A new type of such a high-power sound transducer for active noise control has been developed recently.\(^1\) Principal elements of the transducer are piezoelectric ceramic bimorph oscillators that are driven by high electric field (Fig. 1). Small-signal electromechanical properties of the piezoelectric bimorph structures can be characterized using piezoelectric $d_{33}$, dielectric $\varepsilon_{33}$, and elastic $s_{11}$ constants.

while corresponding high-signal coefficients are highly dependent on the electric field.\(^1\) Study of the high-signal coefficients is technically important. It has been widely acknowledged that electromechanical properties of piezoelectric ceramics depend on the applied electric field.\(^2\) Nevertheless, because of extremely complicated domain structure of piezoelectric ceramics, our knowledge in this is far from complete. Most of the data available are related to the amplitude and phase of $\varepsilon_{33}(E)$ and amplitude of almost no data concerning complex $d_{33}(E)$ and $s_{11}(E)$. Furthermore, there is no comprehensive theoretical model capable of explaining the behavior electromechanical characteristic under high electric field was suggested. In one approach,\(^3\) the Taylor series expansion of thermodynamic potentials leads to the following relation between components of the instantaneous mechanical stress $T$ and strain $S$, electrical displacement $D$ and field $E$.\(^4\)
\[ S_i = s_{ij}^E T_j + d_{im} E_m + s_{ij}^E T_j T_e + 2d_{im} T_e E_m + R_{mn} E_m E_n + \]
\[ + R_{mn}^E E_m E_n E_p + s_{ij}^E T_j T_e T_i + ... \]  
\[ D_n = d_{im} T_i + e_{an}^E E_n + d_{im} T_i T_j - 2R_{mn}^E T_e E_n + \]
\[ + e_{an}^E E_n E_p + e_{an}^E E_n E_p E_q + ... \]  

where \( s, d, R, \) and \( e \) are electromechanical characteristics (constants) of a piezoelectric material. Superscripts \( E \) and \( T \) denotes boundary conditions \( E = \text{const} \) and \( T = \text{const} \), respectively. It is assumed in Eq. (1) that the piezoelectric media does not exhibit losses. The principal drawback of this approach is that the model (1) does not take into account hysteretic behaviors between \( T, S, \) and \( E \) for macroscopically controlled boundary conditions like \( T = 0 \). Experimental data for PZT ceramics\(^{3,8-10} \) suggest that the hysteresis contributes significantly to the electromechanical response of the piezoelectric ceramics at high electric fields. Recently, the experimental Rayleigh law\(^{11} \) describing the hysteresis dependence between the magnetization and magnetic field, has been applied to the mechanical stress dependence of piezoelectrically induced charge on polar surfaces of piezoelectric ceramics.\(^{13,14} \) According to Lord Rayleigh,\(^{12} \) the hysteresis dependencies between \( Y \) (the change in the magnetization) and \( X \) (the magnetic field) when \( X \) varies in the range from \(-X_m \) to \( +X_m \) can be written as

\[
Y = x_0 X - \beta_0 X^2
\]

where \( x_0 \) and \( \beta_0 \) are constants describing reversible and irreversible components, respectively. According to experimental data,\(^{13,14} \) function (2) describes very well the stress dependence of the amplitude of the piezoelectrically induced charge. Nevertheless, as follows from the expansion of Eq. (2) in Fourier's series,\(^{15} \) function \( Y \) does not have the second harmonic while experimental data\(^ 3 \) show that the second harmonic of electrical displacement \( D_1 \) increases significantly with increasing electric field. The suggested functional dependence cannot also explain peculiarities of a gradual increase of the dielectric permittivity in the region where the permittivity begins to grow upon increasing amplitude of the electric field.\(^ 8 \) It should be noted that Eq. (2) is purely intuitive and it is not a result of the Taylor series expansion.

This work was undertaken in an attempt to investigate experimentally and theoretically the behavior of complex piezoelectric \( d_{ij}^j(E_m) \) and dielectric \( e_{ij}^j(E_m) \) coefficients, and the amplitude of elastic \( s_{ij}^E(E_m) \) coefficient of soft PZT under high sinusoidal electric fields and stress-free conditions \((T = 0)\) in the frequency range of 10 Hz-10 kHz. The approach adopted is described in Sec. II. Experimental techniques are presented in Sec. III. Experimental results are analyzed in Sec. IV. Mathematical and physical models and their comparison with experimental data are considered in Sec. V. Summary and conclusions are given in Sec. VI.
$D_3, T_1, S_1$. In the final part of this section, limitations of the suggested measuring procedure related to possible inaccuracy of Eq. (1) for describing hysteretic behavior will be considered.

As follows from Eq. (1) for a particular case when the electric field is applied along the polar $Z$ axis ($E_1 = E_2 = 0, E_3 = E$), and the mechanical stress acts only along the $X$ axis (only $T_1 = 0$), the relations between $S_1, T_1, D_3$, and $E$ are written as

$$S_1 = s_{11}^T T_1 + d_{31} E + d_{111}^T T_1^2 + 2 d_{3111} T_1 E$$

$$- R_{3333} E^3 + R_{3331} E^2 + s_{1111} T_1^3 + \ldots$$

$$D_3 = d_{31} T_1 + e_{33}^T E + d_{3111} T_1^2 + 2 R_{3331} T_1 E$$

$$- e_{3333} E^3 + e_{3331} E^2 + \ldots$$

By grouping all terms, which depend only on the electric field, and introducing

$$d_{31}(E) = d_{31} + R_{3333} E + R_{3331} E^2 + \ldots$$

$$e_{33}^T(E) = e_{33}^T + e_{3333} E^2 + e_{3331} E^2 + \ldots$$

and

$$s_{11}^T(E) = s_{11}^T + s_{1111} T_1^3 - 2 d_{3111} E - s_{1111} T_1^2 + \ldots$$

$$d_{3111}(E) = d_{3111} - d_{3111} T_1 - 2 R_{3331} E$$

Eq. (3) can be written as

$$S_1 = s_{11}^T(E) T_1 + d_{31}(E) E$$

$$D_3 = d_{3111}(E) T_1 + e_{33}^T(E) E$$

Thus, electromechanical coefficients $s_{11}^T(E), d_{31}(E), d_{3111}(E)$, and $e_{33}^T(E)$ are functions of the electric field and mechanical stress in a nonlinear piezoelectric media. In Eq. (6), terms describing the interaction between $E$ and $T_1$ (e.g., $d_{3111}$ and $R_{3331}$) are included in $s_{11}^T(E), d_{31} (E), T_1$); instead of this they can be introduced in $d_{31}(E)$ and $e_{33}^T(E)$, respectively.

As follows from Eq. (6), strain $S_1$ and electrical displacement $D_3$ are directly related to $d_{31}(E)$ and $e_{33}^T(E)$ under stress-free conditions ($T_0 = 0$).

$$S_1 = d_{31}(E) E$$

$$D_3 = e_{33}^T(E) E$$

Thus, coefficients $d_{31}(E)$ and $e_{33}^T(E)$ can be determined by measuring the mechanical strain in the transverse direction and electrical displacement in the polar direction under stress-free conditions. If the sinusoidal electric field is used, the amplitude and phase characteristics of harmonics of $S_1$ and $D_3$ can be gauged.

Measuring $s_{11}^T(E) T_1$ and $d_{31}(E) T_1$ is a significantly more complicated problem since coefficients describing interaction between $E$ and $T_1$ should be determined. Therefore, the experimental technique, which gives only information about $s_{11}^T(E), d_{31}(E)$ (stress-free conditions), was adopted in the present study: measuring the fundamental resonant frequency $\nu_r$ of bending vibrations of a poled piezoelectric ceramic plate in the cantilever configuration (Fig. 2).

Measuring resonant frequency of longitudinal mechanical vibrations excited by the electric field through the primary piezoelectric effect has been used by several authors.$^{7,15}$ Since the resonant frequency is a function of the corresponding mechanical compliance, the dependence of the frequency on the electric field gives information about nonlinearities in the mechanical compliance.$^{7,15}$ If the mechanical quality factor is high enough, mechanical nonlinearities are dominant because of a very high level of vibrations. Therefore this method is not applicable for finding $s_{11}^T(E)$. Moreover, the frequency range of primary interest in this study was 100 Hz–1 kHz while the frequency range of longitudinal vibrations is usually far above 10 kHz. Thus another type of vibration should be considered. The bending mode, which is excited by the electric field in bimorph cantilevers (Fig. 1) may be a possible alternative. The fundamental resonant frequency of bending vibrations is

$$\nu_r = \frac{1.375^2 c_r}{1 + \frac{1}{12} \frac{1}{\sqrt{3}} \frac{1}{\rho}}$$

where $c_r$ is the thickness, $l$ is the length of the bimorph cantilever (Fig. 1), and $\rho$ is the density of the material. Nevertheless, it is quite difficult to analyze the dependence of $\nu_r$ on $E$ in piezoelectric bimorphs since the mechanical stress can reach a very high level, and, in addition, the stress in nonuniform. Preliminary experimental studies showed that bending resonant vibrations can be excited in piezoelectric plates poled along their thickness (Fig. 2). These bending vibrations arise due to nonuniform distribution of spontaneous polarization of domains. Because of a relatively low level of the bending vibrations in the piezoelectric plate (Fig. 2) it can be assumed that the mechanical compliance depends only on the electric field, $s_{11}^T(E, 0) = s_{11}^T(0)$ i.e., the dependence of $\nu_r$ on $E$ is caused mainly by interaction nonlinearities, e.g., $d_{3111}$ in Eq. (5). The corresponding wave equation coincides with the same one describing bending vibrations along the $X$ axis in a linear piezoelectric beam

$$\frac{\partial^2 \eta}{\partial x^2} - \frac{12 \rho \rho s_{11}^T(E) \partial^2 \eta}{c_r^2 \sigma} = 0$$

FIG. 2. Schematic view of the poled piezoelectric ceramic plate in the cantilever configuration. $P_s$ denotes the vector of spontaneous polarization directed along the polar $Z$ axis. Voltage $U$ is applied to polar surfaces of the plate.
where \( \eta \) is the neutral surface displacement of a bent cantilever (Fig. 1).\(^3\) Assuming that the plate is driven at the resonant frequency \( \nu_r \), \( E = E_m \sin(\omega t) \) and all other harmonics of the displacement have a much smaller amplitude, the displacement is written as \( \eta = \eta_0 \sin(\omega t) \). Expansion of the mechanical compliance in the Fourier's series yields

\[
\frac{\partial^2 \eta_0}{\partial y^2} \sin(\omega t + \varphi) - \omega^2 \eta_0 = 12 \rho \left( \int E_0(\omega) \sin(\omega t - \varphi_0) \right) \sin(\omega t - \varphi) \eta_0 = 0.
\]

(11)

Since for a low level of losses, \( \varphi = \pi/2 \), Eq. (11) reduces to a similar one describing the amplitude of vibrations of a linear piezoelectric beam.\(^4\) The fundamental resonant frequency of bending vibrations of the cantilever is written as

\[
\nu_r = \frac{1.875 \, Y_0}{10} \sqrt{3 \left( \int E_0(\omega) \sin(\omega t - \varphi_0) \right) \sin(\omega t - \varphi) \eta_0}.
\]

(12)

Thus, measurement of the resonant frequency as a function of the electric field gives an idea of the field dependence of resonant frequency. Measurement of the resonant frequency as a function of the electric current density provides information about which of Eqs. (6) and (10) is the amplitude of the \( i \)th harmonic. Substituting Eq. (10) into Eq. (9) and taking into account only terms contributing to the fundamental vibrations, one obtains

\[
s_{11}^E = \left( s_{11}(E_m) \right)^{\text{st}} + \sum_{i=1}^{\infty} \left( \varepsilon^{(i)}(E_m) \right) \left( \cos(i\omega t) \right) \\eta_0 = 0.
\]

(10)

Thus, measurements of \( S_i \) and \( D_i \) as a function of \( E \) under \( T_i = 0 \) can provide information about which of Eqs. (4) and (7) is a correct description of piezoelectric and dielectric properties under high electric field. In contrast, the suggested procedure of measuring mechanical properties by means of Eq. (12) is a limited one since it is based on the assumption that Eqs. (5) and (10) are valid. Nevertheless, it is reasonable to assume that these equations are correct if the electric field and losses are not too high. Moreover, the validity of Eq. (5) is not restricted to the validity of Eqs. (4) and (7), therefore function \( \left( s_{11}(E_m) \right)^{\text{st}} \) is obtained from measuring \( \nu_r \) and can provide information about nonlinear or hysteresis origin of these dependencies.

Thus, the mechanical strain \( S_1 \) and dielectric displacement \( D_1 \) of unconstrained piezoelectric ceramic plates and the fundamental resonant frequency of bending vibrations \( \nu_r \) of piezoelectric ceramic plates in the cantilever configuration were chosen for characterizing the behavior of piezoelectric ceramics under high electric fields. \( S_1 \) and \( D_1 \) were determined by measuring harmonics of strain \( S_1 \) along the X axis and of the electric current density \( J \) along the polar Z axis under the sinusoidal electric field and stress-free conditions. In ceramics with low electric conductivity the harmonics of the electric current density \( J \) are directly related to the harmonics of the electrical displacement \( D_1 \).

\[
D_1^{(i)} = \int n_1^{(i)} \sin(i\omega t - \varphi) \, dt.
\]

(15)

where \( \omega \) is the angular frequency of the applied electric field \( E = E_m \sin(\omega t) \). \( n_1^{(i)} \) and \( \varphi \) are the amplitude and phase of the \( i \)th harmonic, respectively. In addition to \( S_1 \), \( D_1 \) and \( \nu_r \), irreversible change \( \Delta P \) in the remanent polarization in the polar direction under pulsed electric field was measured since this method also provides information regarding hysteresis properties of piezoelectric ceramics. It should be stressed that this irreversible change \( \Delta P \) can be caused by both the spontaneous polarization reorientation and hysteresis in the mechanical boundary conditions for ferroelectric domains under applied electric field.

III. EXPERIMENTAL TECHNIQUES

Soft piezoelectric lead zirconate titanate ceramics PZT5H (Morgan Matroc, Inc.), 3203HD (Motorola, Inc.), and PK1550 (Piezo Kinetic, Inc.) were used in the experiments. The ceramics were donor-doped PZT and have a chemical composition near the morphotropic phase boundary. Prepared samples had a shape of parallelepipeds with rectangular cross section. The thickness of the samples was 0.5–3 mm, the width was 3–10 mm, and the length was 20–50 mm. Surfaces having the largest area were electroded with electrolized nickel or sputtered gold and the samples were poled along their thickness at 90–120 °C with \( E = 15–20 \text{ kV/cm} \) in the silicon oil bath. All measurements were conducted on samples aged for at least 24 h.

The harmonic spectrum of the electric current density \( J \) through samples under the applied sinusoidal voltage \( E \) was...
measured by means of the experimental setup shown in Fig. 3. The ceramic sample was chosen such that \( 100R_m < 1/\omega^2 C_m \) where \( C_m \) is the capacitance of the sample. Voltage across the sample was delivered from the power amplifier (790 Series, PCB Piezotronics, Inc., or PA-250H, Julie Research Laboratories, Inc.). The input ac signal to the power amplifier was supplied by a generator (Stanford Research Systems, Inc.). Voltage across resistor \( R_m \) was gauged by a lock-in amplifier SR810 (Stanford Research Systems, Inc.). The input resistance \( R_m \) was chosen such that 100R, and \( C_m \) was used (Fig. 5). A photonic sensor MITI 2000 (MTI Instruments Division) with a corresponding set of electronic equipment was used. A complete description of the experimental setup is given elsewhere.17

To measure the irreversible change in the remnant polarization under high electric field the electric pulse technique was used (Fig. 5). A pair of single rectangular pulses having opposite polarities or single sinusoidal pulse consisting of one period of sinusoidal voltage were delivered to the sample by means of the generator DS345 and the amplifier 790. A digital oscilloscope LaCroy 3310A (LaCroy Corporation) with probe 10:1 (10 MΩ input resistance) were used to measure the induced on capacitor \( C_m = 0.015 \mu F \) electric charge. The amount of the irreversible change in the remnant polarization, \( \Delta P_r \), is equal to:

\[
\Delta P_r = \frac{\Delta U_c C_m}{A}
\]

where \( \Delta U_c \) is residual voltage remaining on the capacitor \( C_m \) after the second of two consecutive rectangular pulses with opposite polarities applied to the sample. The capacitor was short-circuited after the first pulse. Presumably, the second voltage pulse is the one causing the irreversible change in the polarization. The duration of each rectangular pulse was 50 ms. Equation (17) is also valid for the sinusoidal pulse if the irreversible change \( \Delta P_r \) takes place during the second part of the pulse only. The duration of the sinusoidal pulse was 100 ms.

To characterize \( D_1(E) \), \( S_1(E) \), and \( S_1(D_2) \) hysteretic loops, a modified Sawyer-Tower circuit with \( C_m = 9.64 \mu F \) was used (Fig. 6). Generator DS345 and an amplifier BOP 1000M (KEPCO Inc.) with an output voltage up to 1 kV delivered high sinusoidal voltage to the sample under test.

Crystallographic structure of the ceramics used was determined by the x-ray diffraction (XRD) method (Cu Kα.
characteristic line) using powder diffractometer SCINTAG VAX 3100 system. A spectrum produced by the $K \alpha_2$ line of a Cu source was stripped off.

All measurements were carried out at room temperature. A small-signal complex dielectric constant $\varepsilon'_3$ at frequencies far from resonances was calculated by neglecting nonlinearities and using Eqs. (7), (15), and (16).

$$\varepsilon'_3 = \frac{D^{(1)}_3 E^{(1)}_m}{E_m} = \frac{\varepsilon'_3 D^{(1)}_3}{\varepsilon'_3 E^{(1)}_m} = \frac{E^{(1)}_m}{E_m}.$$  \(18\)

where $E_m$ is the amplitude of the applied voltage and $d_3$ is the thickness of the ceramic sample. Since $\varepsilon'_3$ is complex, it also contains information about losses: dissipation factor $\varepsilon'_3$ is written as

$$\varepsilon'_3 = \frac{E^{(1)}_m}{E_m}.$$  \(19\)

IV. EXPERIMENTAL RESULTS

All measured values, except for those characterizing hysteresis loops (e.g., coercive field $E_c$ and remnant polarization $P_r$) were root-mean-square (rms) magnitudes. The results presented here were obtained under alternating electric field without dc bias. Measured small-signal dielectric and piezoelectric characteristics of PZTSH, 3203HD, and PK550 ceramics are shown in Table 1. Since $D^{(1)}_3 / E_m$ and $S^{(1)}_3 / E_m$ at a low level of $E_m$ are equal to $E'_3$ and $d_3$, respectively, it is convenient to analyze ratios

$$\frac{S^{(1)}_3 / E_m}{E'_3}.$$  \(20\)

![FIG. 7. Field dependencies of the relative amplitudes and phases of $D^{(1)}_3 / E_m$ and $S^{(1)}_3 / E_m$ as a function of $E_m$. Relative change in the amplitudes of these ratios and their phase characteristics as a function of $E_m$ for the ceramics studied are shown in Figs. 7-9. $D^{(1)}_3 / E_m$ and $S^{(1)}_3 / E_m$ were normalized with respect to their value at low electric field, $E_m < 20 \text{ kV/cm}$. As is seen from these graphs, for all studied polarized ceramics, the field dependencies of the amplitudes and phases of $D^{(1)}_3$ and $S^{(1)}_3$ for each type of ceramic](https://example.com/fig7)

![FIG. 8. Field dependencies of the relative amplitudes and phases of $D^{(1)}_3 / E_m$ and $S^{(1)}_3 / E_m$ as a function of $E_m$.](https://example.com/fig8)

**TABLE 1.** Small-signal parameters of investigated commercial ceramics.

<table>
<thead>
<tr>
<th>Material</th>
<th>$E_m$ (kV/cm)</th>
<th>$P_r$ (μC/cm²)</th>
<th>$D_{33}$ (nC/V)</th>
<th>Relative $\varepsilon'_3$ (100 Hz)</th>
<th>DF (100 Hz)</th>
</tr>
</thead>
<tbody>
<tr>
<td>PK250</td>
<td>8.2</td>
<td>20.0</td>
<td>-226</td>
<td>3235</td>
<td>0.012</td>
</tr>
<tr>
<td>PZTSH</td>
<td>8.4</td>
<td>29.7</td>
<td>-258</td>
<td>3294</td>
<td>0.017</td>
</tr>
<tr>
<td>3203HD</td>
<td>8.6</td>
<td>38.3</td>
<td>-240</td>
<td>3564</td>
<td>0.016</td>
</tr>
</tbody>
</table>
was very similar in the range of 0–1.4 kV/cm. Additional experiments showed that this is also valid for electric fields at least up to 2.1 kV/cm. A small phase shift between \(D_{31}^{(1)}\) and \(S_{31}^{(1)}\) can be attributed to the experimental limitations in determining the strain phase by means of the strain gauges. Amplitudes \(D_{31}^{(1)}\) and \(S_{31}^{(1)}\) increased almost linearly with increasing \(E_m\), except for the initial part \(E < 0.1 \text{kV/cm rms}\) where the dependencies were more complicated. It was found that there was small hysteresis in \(D_{31}^{(1)}(E_m)\) and \(S_{31}^{(1)}(E_m)\) dependencies: upon decreasing the amplitude of the applied electric field from the maximum to a small value (less than 10 V/cm) the value of \(D_{31}^{(1)}(E_m)/E_m\) and \(S_{31}^{(1)}(E_m)/E_m\) exceeded their initial amplitudes by several percent. The relaxation time was on the order of dozens of minutes.

A study of the effect of dc bias was also conducted. It was found that the shape of functions \(D_{31}^{(1)}(E_m)/E_m\) and \(S_{31}^{(1)}(E_m)/E_m\) depended significantly on biasing dc electric field. If the biasing electric field was high enough, functions \(D_{31}^{(1)}(E_m)/E_m\) and \(S_{31}^{(1)}(E_m)/E_m\) had a bell shape, i.e., they even decreased at a certain amplitude of the electric field that was not the case for zero dc bias.

The field dependencies of the ratio \(\nu(0)/\nu(E_m)\) and of the tip displacement of the piezoelectric cantilever (Fig. 2) at the fundamental resonant frequency of bending vibrations are shown in Fig. 10. The measured phase shift between the applied electric field and tip displacement was \(\phi = -75^\circ \pm 15^\circ\). As is seen from the figure, \(\nu(0)/\nu(E_m)\) increased almost linearly with increasing electric field except for low electric field. From a comparison of Figs. 7–9 and 10, the field dependence of \(\nu(0)/\nu(E_m)\) was more moderate than that of functions \(D_{31}^{(1)}(E_m)/E_m\) and \(S_{31}^{(1)}(E_m)/E_m\) for low electric field at resonance for PZT550 ceramic plate in the cantilever configuration. The length of cantilever is 27 mm, the end is 11 mm, and thickness is 0.55 mm.

Frequency dependencies of functions \(D_{31}^{(1)}(E_m)\) and \(S_{31}^{(1)}(E_m)\) in the range of 10 Hz–10 kHz in small electric fields (less than 10 V/cm rms) are presented in Fig. 11. The functions were normalized with respect to their value at 100 Hz. As is seen in the figure, the functional dependencies of the electric displacement and strain had the same tendency of slow decrease with increasing frequency. Scattering in the magnitude of the strain measured by means of the laser interferometer was probably caused by instrumental limitations and spurious vibrations due to the sample clamping in the holder. Despite a good stability of results obtained with the strain gauge, it was found that the strain gauge cannot be
used at frequencies above 500 Hz. This was caused by increasing phase delay introduced by the strain gauge.

Frequency dispersion of the amplitudes of the real and imaginary parts of $D^{(1)}_{1m}/E_m$ for PK1550 was also studied under various levels of the applied sinusoidal field. Since at small electric fields the real part of the relative dielectric permittivity is equal to $(D^{(1)}_{1m}/E_m)\cos(\omega t)$ and the dielectric losses are expressed by Eq. (19), the electric field dependencies of these quantities were calculated and plotted in Fig. 12. As is seen in the figure, the shape of the frequency dispersion curves depended on the amplitude of the applied field. Field dependencies of the normalized amplitudes of $D^{(1)}_{1m}/E_m$ at 100 Hz and 1 kHz for the ceramics studied are given in Fig. 13. As shown in the graph, a nonlinear increase in the first harmonic $D_{1m}$ at 100 Hz was large than at 1 kHz.

The initial part of the electric field dependence of $D^{(1)}_{1m}/E_m$ at several frequencies and the corresponding graph of $-D^{(1)}_{1m}\sin(\omega t)$ vs $D^{(1)}_{1m}\cos(\omega t)$ at 100 Hz for the 3203HD sample is shown in Fig. 14. As can be seen in the first graph, the dependence of $D^{(1)}_{1m}/E_m$ on $E_m$ at an electric field smaller than 0.1 kV/cm had significant nonlinearity and frequency dispersion. It was also found that there was a significant time instability in the absolute value of the dielectric response at this very low scale. Moreover, hysteresis was observed: even if the amplitude of the electric field increased from zero to a relatively low level of 50–100 mV/cm, the initial values of $D^{(1)}_{1m}$ and of the slope of $D^{(1)}_{1m}/E_m$ changed during immediate consecutive measurements. Nevertheless, if second measurements were repeated for several hours, the slope of $D^{(1)}_{1m}/E_m$ was almost consistent with that in the initial measurements. The second graph of Fig. 14 shows that the ratio between the amplitudes of imaginary and real parts of $D^{(1)}_{1m}$ was nonlinear.

Field dependencies of amplitudes and phases of the first three harmonics of the electrical displacement for a 3203HD sample are shown in Fig. 15. Field dependencies of amplitudes and phases of the first and second harmonics of the electrical displacement for another 3203HD sample are shown in Fig. 16. It should be noted that the level of the second and third harmonics of the electrical displacement caused by the instrumental nonlinearities was at least three times smaller than the measured values of $D^{(2)}_{1m}$ and $D^{(3)}_{1m}$ for the ceramics studied. The experimental data of Figs. 15 and 16 demonstrate that despite an increase of $D^{(1)}_{1m}/E_m$ 1.5–1.6 times at 2 kV/cm rms, the relative level of the second and third harmonics was less than several percent. As shown in Figs. 15 and 16, the second harmonic was polar and it phase...
was close to $-90^\circ$ or $-90^\circ$ depending on the mutual direction of the spontaneous polarization and electric field. It is also follows that the field dependence of the amplitude of the second harmonic varied from sample to sample. Moreover, experimental results also showed that the harmonic was unstable in time. Surprisingly, the third harmonic demonstrated saturation-like behavior (Fig. 15). The phase of the harmonic was close to $-90^\circ$.

Dependencies of the irreversible change $\Delta P_r$ in the remnant polarization on the electric field for a 3203HD sample are shown in Fig. 17. As shown, the change in the polarization took place even at electric fields much smaller than the coercive equal to 8.6 kV/cm. From a comparison of the amount of the changed polarization for sinusoidal and rectangular pulses, the change in the polarization depended on the amplitude and not on the rms value of the applied electric field (for pulses with the same duration).

Dependencies of the irreversible change $\Delta P_r$ in the remnant polarization on the electric field for a 3203HD sample are shown in Fig. 17. As shown, the change in the polarization took place even at electric fields much smaller than the coercive equal to 8.6 kV/cm. From a comparison of the amount of the changed polarization for sinusoidal and rectangular pulses, the change in the polarization depended on the amplitude and not on the rms value of the applied electric field (for pulses with the same duration).
FIG. 19. $D$ vs $E$, $S_1(E)$, and $S_1(D)$ hysteresis loops for 3203HD sample in electric field 13 kV/cm rms. The frequency of the field is 10 Hz.

![Graph](image)

harmonics of $D_2$ and $S_1$ (Fig. 9), which demonstrated the same behavior. This simple relation between $S_1$ and $D_3$ broke under an electric field that was higher than the coercive field (Fig. 19): the behavior became hysteretic. It was also found that under this level of the field, hysteretic loops decreased in time but they could be restored by holding the sample under zero field for a time depending on the sample.

Part of XRD spectra (2θ scan, Cu Kα1 line) and their deconvolutions for poled and depoled 3203HD ceramics are shown in Fig. 20. This range of 2θ was chosen since the tetragonal (T) and rhombohedral (R) phases are distinguishable on X-ray powder diffractiongrams at a 200/020/002 set of lines. The 200 reflections form a doublet in the tetragonal phase while 200 is a singlet in the rhombohedral phase. As is seen from the results of the deconvolution, there was a triplet at the 200/020/002 set. Therefore, 3203HD ceramics had the chemical composition at the morphotropic phase boundary (NBP) since both tetragonal and rhombohedral phases were presented in the ceramics. Calculated from the data the $c/a$ ratio for the tetragonal cell was equal to 1.0133. Comparison of XRD spectra for depoled and poled ceramics showed that there was a significant amount of 90° domain walls in the tetragonal phase of the poled ceramics since the 200 T line did not disappear upon poling the sample.

V. DISCUSSION

In the discussion of the experimental data, it is convenient to begin from the mathematical analysis of field dependencies of the electrical displacement $D_1$ and strain $S_1$. This makes it possible to find an adequate mathematical description of the experimental data. Then, physical causes of the observed behavior of the electromechanical properties of the soft piezoelectric ceramics under the sinusoidal electric field are analyzed.

A. Nonlinear model

As seen from Figs. 7-9 and 13-16, field dependencies of $D_1(E)$, $S_1(E)$, and $S_1(D)$ for all soft piezoelectric ceramics studied show an almost linear increase with increasing amplitude of the sinusoidal electric field except for low electric fields (Fig. 14). Relative values of the second harmonic of the electrical displacement $D_2$ also demonstrate almost linear dependence, while the third harmonic of the electrical displacement $D_3$ shows a saturation-like behavior.
Mathematical analysis of these dependencies can provide information as to whether nonlinear or hysteretic models are adequate for describing the electromechanical properties. As follows from the nonlinear model of the field dependence of $S_1$ and $D_1$ under stress-free conditions. [Eqs. (4) and (7)], corresponding dependencies can be written as

$$S_1 = d_{31}E - R_{331}E^2 + R_{333}E^3 - \ldots,$$
$$D_1 = e_{33}E - T_{332}E^2 + T_{333}E^3 - \ldots.$$  \hspace{1cm} (21)

As follows from the spectral analysis of Eq. (21) for sinusoidal applied electric field $E = E_m \sin(\omega t)$, the first three harmonics of $S_1$ and $D_1$ (not including the dc component) can be written as

$$S_1^{(1)} = d_{31} - \frac{3}{4} R_{331}E_m^2 - \ldots |E_m \sin(\omega t)|,$$
$$S_1^{(2)} = \frac{1}{2} R_{331}E_m^2 - \ldots |E_m \sin(2\omega t - \pi/2)|,$$
$$S_1^{(3)} = \frac{1}{2} R_{333}E_m^3 - \ldots |E_m \sin(3\omega t - \pi)|,$$
$$D_1^{(1)} = e_{33} - \frac{3}{4} T_{332}E_m^2 - \ldots |E_m \sin(\omega t)|,$$
$$D_1^{(2)} = \frac{1}{2} T_{332}E_m^2 - \ldots |E_m \sin(2\omega t - \pi/2)|,$$
$$D_1^{(3)} = \frac{1}{2} T_{333}E_m^3 - \ldots |E_m \sin(3\omega t - \pi)|.$$  \hspace{1cm} (23)

As seen from Eqs. (22) and (23), the series in brackets consists of odd powers of $E_m$ for even harmonics and of even powers of $E_m$ for odd harmonics of $S_1$ and $D_1$. Keeping only two first terms in the series expansion of the harmonics of $D_1$, corresponding approximating functions can be written as

$$\frac{D_1^{(1)}}{E_m} = \frac{\alpha_1 - \alpha_2 E_m^2}{\alpha_1},$$
$$\frac{D_1^{(2)}}{E_m} = \frac{\alpha_1 E_m^2 - \alpha_2 E_m^4}{\alpha_1 - \alpha_2 E_m^2},$$
$$\frac{D_1^{(3)}}{E_m} = \frac{\alpha_1 E_m^4}{\alpha_1 - \alpha_2 E_m^2},$$

where $\alpha_i$ are fitting parameters. Thus, five constant parameters are used for approximate of the amplitudes of the first three harmonics of $D_1$ with polynomials. Fitting experimental data of Fig. 15 for the first three harmonics of $D_1$ according to Eq. (24) is shown in Fig. 21. Clearly, the functional dependencies of the first and third harmonics do not describe satisfactorily the experimental data. The first equation in set (24) corresponds to a quadratic dependency while experimental data for $D_1^{(2)}E_m^2/E_m$ demonstrate an approximately linear dependence of the amplitude on the applied electric field except for low electric fields. Moreover, as seen from Eqs. (22) and (23), the fact that there are no losses in the nonlinear model of $D_1^{(2)}E_m^2/E_m$ and $S_1^{(1)}E_m^2/E_m$ contradicts experimental results, which showed a steady increase in the phase shift with increasing amplitude of the electric field (Figs. 15–16). The nonlinear function for approximating the third harmonic [last equation in set (24)] departs significantly from the experimental data. In addition, the phase of the third harmonic according to the model should be $\pi$ [Eq. (24)] while the experimental values are close to $-\pi/2$ (Fig. 15). Using Eqs. (5), (11), and (12), it can be also shown that for the nonlinear model the relative change in the resonant frequency squared, $(\nu_r/\nu_0(E_m))^2$, is described by a quadratic law similar to one for $D_1^{(2)}E_m^2/E_m$ and $S_1^{(1)}E_m$. Nevertheless, corresponding experimental data (Fig. 10) demonstrate approximately linear dependence. Thus, as follows from the analysis conducted above, the nonlinear model does not provide an adequate description of the electric field dependencies of the studied electromechanical properties of the soft piezoelectric ceramics under stress-free conditions.

### B. Hysteretic model

As was discussed in Sec. I, a hysteretic model can be suggested as an alternative for describing the behavior of the soft piezoelectric ceramics under applied electric field. The main peculiarity of hysteresis is multivaluedness in functional dependencies. It should be noted that from the mathematical viewpoint the nonlinear model is a partial case of the hysteretic model.

As was shown in Sec. II, Eq. (4) is the most general hysteretic function which can characterize electric field dependencies of $D_1$ and $S_1$ under stress-free conditions. There is an experimental confirmation that soft piezoelectric ceramics demonstrate hysteretic behavior in the electric field. As follows from Fig. 17, an irreversible change $\Delta \rho$, in the remnant polarization takes place in the periodic electric field that is much smaller than the coercive one (Fig. 17). This irreversible change, i.e., hysteretic process, contributes to the observed increase in the amplitude and phase delay of harmonics of $D_1$ with increasing electric field. A qualitative analysis regarding the correlation in the behavior of $D_1$ and $\Delta \rho$ will be given later.
In the periodical electric field with two extreme values $-E_m$ and $+E_m$, the following initial conditions are fulfilled for steady-state hysteresis loops:

$$D^2_3(E_m) = D^2_3(-E_m).$$  \hspace{1cm} (26)

$$D^2_3(-E_m) = D^2_3(E_m).$$

In addition, in this case $\alpha, \beta,$ and $\gamma$ are functions of $E_m$ only. Using Eq. (26), Eq. (25) reduces to

$$D^2_3(E) = D^2_3(0) + \alpha(E_m)E - \gamma(E_m)E^3$$

$$+ \left( \beta'(E_m)E^3 \right) \text{ up,}$$

$$+ \left( \alpha(E_m)E - \gamma(E_m)E^3 + \beta'(E_m)E^3 \right) \text{ down.}$$ \hspace{1cm} (27)

where

$$\alpha(E_m) = \alpha^E = \alpha^d,$$

$$\gamma(E_m) = \gamma^E = \gamma^d.$$  \hspace{1cm} (28)

Thus, as seen from a comparison of Eqs. (21) and (27), the Taylor's expansion coefficients in the hysteretic model become a function of $E_m$ and the relation between $D_3$ and $E$ becomes a multiple valued function. Fourier's analysis of harmonics of $D_3$ for the hysteretic model (27) with $E = E_m\sin(\omega t)$ gives

$$D^{0,1}_3 = D^{0,1}_3(0) - \frac{\beta'(E_m)E^3}{2}E_m^2,$$

$$D^{1,1}_3 = \left( \alpha(E_m) + \frac{\gamma(E_m)}{2} \right)E_m^2 \sin(\omega t)$$

$$- \frac{1}{3\pi} \left[ -\beta'(E_m)E_m^2 \right] E_m^2 \cos(\omega t),$$ \hspace{1cm} (29)

$$D^{2,1}_3 = \frac{1}{2} \left[ \beta'(E_m)E_m^2 - \beta''(E_m)E_m^4 \right] \sin \left( 2\omega t - \frac{\pi}{4} \right),$$

$$D^{3,1}_3 = - \frac{1}{15\pi} \left[ -\beta'(E_m)E_m^2 \right] E_m^2 \cos(3\omega t).$$

In ferroelectrics such as soft piezoelectric ceramics, which have a significant amount of domain walls and very complex elastic boundary conditions, it is reasonable to assume that functions $\alpha(E_m), \beta(E_m),$ and $\gamma(E_m)$ are rather smooth and are well described by truncated Taylor's series at amplitudes of the electric field much smaller than the coercive one. Assuming that $\gamma(E_m) = 0$ and expanding $\alpha(E_m)$ and $\beta(E_m)$ in Taylor's series and using only the first two terms in these series, Eq. (27) reduces to

$$D_3(E) = D_3(0) - \alpha E - \beta E^2 - \gamma E^3,$$ \hspace{1cm} (25)

where $\alpha, \beta,$ and $\gamma$ are functions depending on the corresponding sequence of past extremum values of the electric field; superscripts "u" and "d" describe ascending and descending branches of $D_3$, respectively (Fig. 22). Clearly, strain $S_3$ can be described by the equations analogous to Eq. (25).
where \( \alpha_0, \alpha_1, \beta_0, \) and \( \beta_1 \) are the expansion constants. It can be shown that Eq. (2) introduced by Lord Rayleigh for describing the hysteresis in magnetic media is a partial case of Eq. (30) when the reference level and it's change in the electric field are neglected in both equations and \( \alpha_0 = \alpha_R, \)
\[ \alpha_1 = 2\beta_0 = \beta_R, \] and \( \beta_2 = \beta_1 = 0. \)

Corresponding to Eq. (30) the first four harmonics (including the dc component) are
\[ D_j^{(3)} = D_j^{(3)}(0) - \frac{2}{\lambda} \left[ \beta_0^{(1)} - \beta_1^{(1)} E_m \right] \cos(\omega t), \]
\[ D_j^{(2)} = \frac{4}{\lambda^2} \left[ \beta_0^{(2)} + \beta_1^{(2)} E_m \right] \sin(\omega t), \]
\[ D_j^{(1)} = \frac{4}{\lambda^3} \left[ \beta_0^{(3)} + \beta_1^{(3)} E_m \right] \sin(3\omega t - \frac{\pi}{2}), \]
\[ D_j^{(0)} = \frac{4}{\lambda^4} \left[ \beta_0^{(4)} + \beta_1^{(4)} E_m \right] \sin(3\omega t + \frac{\pi}{2}), \]

where
\[ \beta_0^{(1)} = \beta_1^{(1)} E_m, \]
\[ \beta_0^{(2)} = \beta_1^{(2)} E_m, \]
\[ \beta_0^{(3)} = \beta_1^{(3)} E_m, \]
\[ \beta_0^{(4)} = \beta_1^{(4)} E_m. \]  

Thus, Eq. (31) can be considered as a first order approximation for describing harmonics of the hysteresis dependence of \( D_j \) on the sinusoidal electric field \( E = E_m \sin(\omega t). \) A set of equations analogous to Eq. (31) can be also written for \( S_j(E). \) As is seen in Eqs. (29) and (31), there is a phase delay \( \omega \) in the first harmonic \( D_j^{(1)} \), which appears owing to the hysteresis. It should be noted that in the type of hysteresis loop schematically presented in Fig. 22, \( \beta_0^{(1)}(E_m) > 0 \) and \( \beta_0^{(2)}(E_m) < 0. \) Therefore, the phase of the first harmonic is always negative. The sign of the phase of the second harmonic depends on the sign of the sum \( \beta_0^{(2)}(E_m) = \beta_0^{(2)}(E_m). \) Depending on the mutual direction of the remnant polarization and the applied electric field, the sum can be positive or negative. The sign of the third harmonic phase can be negative or positive, but for the dependence represented by Eq. (30) it should be \(-\pi/2. \) Thus, in the first approximation the phase shift of the second harmonic is \( \pm \pi/2 \) and of the third one is \( -\pi/2. \) (Eq. (31)). This qualitative analysis of phase characteristics of the first three harmonics of \( D_j \) is in agreement with the experimental data of Figs. 15 and 16. It should be noted that the nonlinear model (Eqs. (21)–(23)) fails to explain the phase shift in the first harmonic and gives a wrong phase of the third harmonic.

As follows from Eq. (29), \( D_j^{(3)} \sin(\phi_j^{(3)}) = 5 \) and corresponding values calculated from the experimental data of Fig. 15 are given in Fig. 22. The theoretical value is smaller than the experimental ratio, which equals approximately 13. We assume that the discrepancy is related to the limited accuracy of the presentation of the hysteresis by the cubic polynomials in \( E \) (Eq. (25)), which gives the hysteretic part represented by quadratic terms in Eq. (27). Analysis shows that if the hysteretic part of the dependence \( D_j(E) \) is described by the term including \( E^4 \) only, the ratio
\[ D_j^{(3)} \sin(\phi_j^{(3)}) / D_j^{(2)} \sin(\phi_j^{(2)}) = 5 \]
and corresponding values calculated from the experimental data of Fig. 15 are given in Fig. 23. The theoretical value is smaller than the experimental ratio, which equals approximately 13. We assume that the discrepancy is related to the limited accuracy of the presentation of the hysteresis by the cubic polynomials in \( E \) (Eq. (25)), which gives the hysteretic part represented by quadratic terms in Eq. (27). Analysis shows that if the hysteretic part of the dependence \( D_j(E) \) is described by the term including \( E^4 \) only, the ratio
\[ D_j^{(3)} \sin(\phi_j^{(3)}) / D_j^{(2)} \sin(\phi_j^{(2)}) = 5 \]
and corresponding values calculated from the experimental data of Fig. 15 are given in Fig. 23. The theoretical value is smaller than the experimental ratio, which equals approximately 13. We assume that the discrepancy is related to the limited accuracy of the presentation of the hysteresis by the cubic polynomials in \( E \) (Eq. (25)), which gives the hysteretic part represented by quadratic terms in Eq. (27). Analysis shows that if the hysteretic part of the dependence \( D_j(E) \) is described by the term including \( E^4 \) only, the ratio
\[ D_j^{(3)} \sin(\phi_j^{(3)}) / D_j^{(2)} \sin(\phi_j^{(2)}) = 5 \]
and corresponding values calculated from the experimental data of Fig. 15 are given in Fig. 23. The theoretical value is smaller than the experimental ratio, which equals approximately 13. We assume that the discrepancy is related to the limited accuracy of the presentation of the hysteresis by the cubic polynomials in \( E \) (Eq. (25)), which gives the hysteretic part represented by quadratic terms in Eq. (27). Analysis shows that if the hysteretic part of the dependence \( D_j(E) \) is described by the term including \( E^4 \) only, the ratio
\[ D_j^{(3)} \sin(\phi_j^{(3)}) / D_j^{(2)} \sin(\phi_j^{(2)}) = 5 \]
and corresponding values calculated from the experimental data of Fig. 15 are given in Fig. 23. The theoretical value is smaller than the experimental ratio, which equals approximately 13. We assume that the discrepancy is related to the limited accuracy of the presentation of the hysteresis by the cubic polynomials in \( E \) (Eq. (25)), which gives the hysteretic part represented by quadratic terms in Eq. (27). Analysis shows that if the hysteretic part of the dependence \( D_j(E) \) is described by the term including \( E^4 \) only, the ratio
\[ D_j^{(3)} \sin(\phi_j^{(3)}) / D_j^{(2)} \sin(\phi_j^{(2)}) = 5 \]
and corresponding values calculated from the experimental data of Fig. 15 are given in Fig. 23. The theoretical value is smaller than the experimental ratio, which equals approximately 13. We assume that the discrepancy is related to the limited accuracy of the presentation of the hysteresis by the cubic polynomials in \( E \) (Eq. (25)), which gives the hysteretic part represented by quadratic terms in Eq. (27). Analysis shows that if the hysteretic part of the dependence \( D_j(E) \) is described by the term including \( E^4 \) only, the ratio
\[ D_j^{(3)} \sin(\phi_j^{(3)}) / D_j^{(2)} \sin(\phi_j^{(2)}) = 5 \]
and corresponding values calculated from the experimental data of Fig. 15 are given in Fig. 23. The theoretical value is smaller than the experimental ratio, which equals approximately 13. We assume that the discrepancy is related to the limited accuracy of the presentation of the hysteresis by the cubic polynomials in \( E \) (Eq. (25)), which gives the hysteretic part represented by quadratic terms in Eq. (27). Analysis shows that if the hysteretic part of the dependence \( D_j(E) \) is described by the term including \( E^4 \) only, the ratio
\[ D_j^{(3)} \sin(\phi_j^{(3)}) / D_j^{(2)} \sin(\phi_j^{(2)}) = 5 \]
and corresponding values calculated from the experimental data of Fig. 15 are given in Fig. 23. The theoretical value is smaller than the experimental ratio, which equals approximately 13. We assume that the discrepancy is related to the limited accuracy of the presentation of the hysteresis by the cubic polynomials in \( E \) (Eq. (25)), which gives the hysteretic part represented by quadratic terms in Eq. (27). Analysis shows that if the hysteretic part of the dependence \( D_j(E) \) is described by the term including \( E^4 \) only, the ratio
\[ D_j^{(3)} \sin(\phi_j^{(3)}) / D_j^{(2)} \sin(\phi_j^{(2)}) = 5 \]
and corresponding values calculated from the experimental data of Fig. 15 are given in Fig. 23. The theoretical value is smaller than the experimental ratio, which equals approximately 13. We assume that the discrepancy is related to the limited accuracy of the presentation of the hysteresis by the cubic polynomials in \( E \) (Eq. (25)), which gives the hysteretic part represented by quadratic terms in Eq. (27). Analysis shows that if the hysteretic part of the dependence \( D_j(E) \) is described by the term including \( E^4 \) only, the ratio
\[ D_j^{(3)} \sin(\phi_j^{(3)}) / D_j^{(2)} \sin(\phi_j^{(2)}) = 5 \]
and corresponding values calculated from the experimental data of Fig. 15 are given in Fig. 23. The theoretical value is smaller than the experimental ratio, which equals approximately 13. We assume that the discrepancy is related to the limited accuracy of the presentation of the hysteresis by the cubic polynomials in \( E \) (Eq. (25)), which gives the hysteretic part represented by quadratic terms in Eq. (27). Analysis shows that if the hysteretic part of the dependence \( D_j(E) \) is described by the term including \( E^4 \) only, the ratio
\[ D_j^{(3)} \sin(\phi_j^{(3)}) / D_j^{(2)} \sin(\phi_j^{(2)}) = 5 \]
measured results using the sinusoidal-pulse technique (Fig. 17): the discrepancy does not exceed 10%-25%.

Thus, analysis of the hysteric model of $D_3(E)$ conducted above demonstrates that the model is a very good approximation for describing the behavior of the electrical displacement $D_3$ under the high electric field and stress-free conditions. A simplification of the hysteresis loop $D_3(E)$ can be made since despite a significant change in the dc component and first harmonic with increasing electric field, the relative level of the second and third harmonics does not exceed several percent (see Figs. 15-17). Therefore, Eq. (30) reduces to

$$D_3(E) = D_3^{0}(E_m) - D_3^{11}(E_m). \quad E = E_m \sin(\omega t). \quad (35)$$

Thus, the dielectric response of the soft piezoelectric PZT ceramics to the applied sinusoidal electric field, which is smaller than the coercive one, basically consists of two components: the dc part and ac part at the same frequency $\omega$. Usually, since only the first harmonic is of importance, changes in the remnant polarization level can be omitted and complex-function formalism can be used for describing the dielectric response:

$$D_{m}^* = \varepsilon_{ij}^{0}(E_m)E_m,$$

where $D_{m}^*$ is the complex amplitude of the ac component of the electrical displacement along the polar axis, and $\varepsilon_{ij}^{0}(E_m)$ is the complex dielectric permittivity in the sinusoidal electric field, which is approximately given by [see Eq. (34)]

$$\varepsilon_{ij}^{0}(E_m) = \varepsilon_0 - \varepsilon(E_m)e^{-i\phi(E_m)}.$$  

(37)

where $\varepsilon_0$ practically coincides with small-signal dielectric permittivity, $\varepsilon_0 = \varepsilon_0'$. It should be noted that Eq. (37) is valid only for the case when no dc bias is applied (The electric field changes from $-E_m$ to $-E_m$). With the dc bias applied, function $\varepsilon_{ij}^{0}(E_m)$ can differ from that described by Eq. (37).

As seen from Figs. 7 and 9 there is a strong correlation between pairs $D_3^{11}(E_m)$ and $S_{11}^{1}(E_m)$ for all ceramics studied in electric fields 0-1.4 kV/cm rms. Therefore, the suggested hysteretic model should be also valid for function $S_{11}^{1}(E)$ in this range. Corresponding to Eq. (36) the strain-field relation under stress-free conditions is

$$S_{11}^{m} = d_{33}(E_m)E_m.$$  

(38)

where $S_{11}^{m}$ is the complex amplitude of the ac component of the mechanical strain in the direction perpendicular to the polar axis and $d_{33}(E_m)$ is the complex piezoelectric coefficient in the sinusoidal electric field: the coefficient is approximately described by the function directly proportional to the right part of Eq. (37). Thus, as follows from Eqs. (36) and (38), experimentally measured complex functions $D_3^{11}(E_m)$ and $S_{11}^{1}(E_m)$ (see Figs. 7-9 and 11-14) are equal to complex dielectric $\varepsilon_{ij}^{0}(E_m)$ and piezoelectric $d_{33}(E_m)$ coefficients, respectively.

It is also reasonable to assume that the behavior of elastic properties of soft piezoelectric ceramics is of the hysteretic type and mechanical compliance $s_{11}^{m}(E_m)$ is described by a formula similar to Eq. (37). Therefore, the first harmonic should be dominant in the ac component of the mechanical response and should increase linearly with increasing amplitude of the electric field. Therefore, the fundamental resonant frequency of bending vibrations under stress-free conditions instead of Eq. (12) is described by

$$\nu = \frac{1.375 f_0}{\pi} \left( \frac{1}{\varepsilon_{ij}^{0}(E_m)} \right)$$

(39)

where $|s_{11}^{m}(E_m)|$ is the amplitude of the elastic coefficient. As follows from Fig. 10, $|s_{11}^{m}(E_m)|$ increases almost linearly with increasing electric field that is consistent with the suggested hysteretic model. The field dependence of the amplitude of $s_{11}^{m}(E_m)$ is more gradual than that of $\varepsilon_{ij}^{0}(E_m)$ and $d_{33}(E_m)$. For example, for PK550 ceramics, as follows from Fig. 7 and Eq. (37), $|s_{11}^{m}(E_m)/s_{11}^{m}(0)| \approx 0.981 - 0.251 E_m$ and as follows from Fig. 10 and Eq. (39), $|s_{11}^{m}(E_m)/s_{11}^{m}(0)| \approx 0.998 - 0.097 E_m$, where $E_m$ is in kV/cm rms. Therefore the slope of the function $s_{11}^{m}(E_m)$ is about 40% of that of $\varepsilon_{ij}^{0}(E_m)$ for these ceramics.

Thus, the analysis conducted above strongly supports the hypothesis that hysteresis is the cause of the high field behavior of the soft piezoelectric ceramics. Previous works [13] suggested that the dependence of piezoelectric coefficients on applied mechanical stress is also of hysteretic type. As logical sequel to this, it can be assumed that under simultaneous action of the electric field and mechanical stress, the electromechanical response of the ceramics will be essentially of the hysteretic type. If sinusoidal electric field is applied along the $Z$ axis, and sinusoidal mechanical stress along the $X$ axis, the following set of equations may approximately describe the electromechanical response:
\[ S_{1m}^+ = s_{15}^2 (E_m - T_{1m})T_{1m} + d_3 (E_m - T_{1m})E_m, \]
\[ D_{3m}^- = d_3^1 (E_m - T_{1m})T_{1m} - e_{53}^m E_m, \]
where \( T_{1m} \) is the amplitude of the mechanical stress. dc components are not included in set (40). It should be noted that there is a principal difference between Eqs. (5) and (6) corresponding to the nonlinear model and Eq. (40) describing the hysteretic model. First, \( D_{3m}^- \) describes relations between complex amplitudes while Eqs. (5) and (6) are related to instant values. Second, in the nonlinear model, a significant increase in the first harmonic of the electromechanical response implies a corresponding significant increase in the amplitude of higher odd harmonics [see, for example, Eq. (24)]. This is not true for the hysteretic model (40) in which only first harmonics are taken into account since experiments show that higher harmonics are not significant at electric fields lower than the coercive one.

Before proceeding with the analysis of the physical reasons for the observed hysteretic behavior of soft piezoelectric ceramics, several important remarks regarding limitations of the developed hysteretic model should be made. The suggested hysteretic model is of a static type, i.e., it does not take into account the frequency dispersion of electromechanical characteristics of the ceramics. Experimental data (Figs. 11-14), demonstrate that there is a small frequency dependence in the measured electrical displacement \( D_1 \) and mechanical strain \( S_1 \); the frequency dispersion is also a function of the electric field. Experimental data also show (Fig. 12) that there are peculiarities in the behavior of \( \varepsilon_{33}^T (E) \) at low electric fields that are not described by Eqs. (37) and (47). Finally, according to Eq. (37) the phase of the dielectric permittivity \( \varepsilon_{33}^T (E) \) approaches zero at low electric fields, while corresponding experimental values fall within the range \(-0.2^\circ-1^\circ\) (see Fig. 24). Thus, there are dielectric losses having nonhysteretic origin that are not included in the suggested hysteretic model. Nevertheless, the hysteretic model describes the main peculiarities of studied electromechanical properties of soft piezoelectric ceramics at high electric fields.

C. Physical models

In this section, possible physical mechanisms of the observed behavior of soft piezoelectric ceramics under the electric field will be analyzed. As is seen from Figs. 7-9 and the corresponding hysteretic model [Eqs. (26) and (38)] there is a strong correlation between complex dielectric \( \varepsilon_{33}^T (E) \) and piezoelectric \( d_{31} (E) \) coefficients for all ceramics studied in the electric field of 0-1.4 kV/cm rms. The data of Fig. 11 demonstrate that frequency dependencies of \( \varepsilon_{33}^T (E) \) and \( d_{31} (E) \) are also similar. As is seen from a comparison of Figs. 7 and 10 and Eqs. (36), (38), and (19), the field dependence of the amplitude of \( s_{15}^2 (E) \) is more gradual than that of \( \varepsilon_{33}^T (E) \) and \( d_{31} (E) \). Comparison of \( D_{31} (E) \), \( S_{1} (E) \), and \( S_{1} (E) \) hysteresis loops (Figs. 18 and 19) also shows that there is a linear relation, without phase shift, between \( D_1 \) and \( S_1 \) in the electric field of 1 kV/cm rms. In electric fields higher than the coercive one the correlation breaks up: the electrical displacement \( D_1 \) and strain \( S_1 \) behave in a very different manner (Fig. 19).

One possible explanation of the observed behavior of the electromechanical properties of the ceramics is based on the model of the domain wall and interphase boundary motion. According to this approach, the reorientation of the spontaneous polarization of microscopic domains under applied electric field contributes to the dielectric, piezoelectric, and elastic response of the media.\textsuperscript{11,22} This extrinsic contribution to the electromechanical properties was estimated to be especially significant for soft piezoelectric ceramics.\textsuperscript{11} XRD data (Fig. 20) show that the ceramics studied consist of a mixture of tetragonal and rhombohedral phases. 180° polarization switching in both phases does not produce a change in the domain dimensions along certain directions. Therefore, the above discussed domain processes can contribute mainly to the dielectric response in PZT. 90° polarization reorientation in the tetragonal state and the tetragonal/rhombohedral phase boundary motion can contribute significantly to both dielectric and piezoelectric responses since the polarization change is accompanied by a significant (up to several percent) change in the domain dimensions along certain directions. This domain reorientation also causes mechanical stress in the media that finally produces an apparent change in the elastic properties. Therefore, according to the model based on the polarization reorientation, an increase in the 90° and tetragonal/rhombohedral polarization reorientation should be responsible for an observed increase in \( \varepsilon_{33}^T (E) \), \( d_{31} (E) \), and \( s_{15}^2 (E) \) with increasing electric field. Clearly, hysteretic behavior of the electromechanical characteristics is also a result of the polarization reorientation. When the electric field is sufficiently high, 180°, 71°, and 109° polarization switching probably becomes dominant. Therefore, \( D_{31} (E) \) and \( S_{1} (E) \), unlike for fields smaller than the coercive one (Fig. 18), behave in a different manner (Fig. 19). It is rather difficult to develop a corresponding qualitative mathematical model since it should take into account the polarization reorientation in numerous domains with corresponding complex microscopic boundary conditions. It should be noted that it is unclear from the model of the domain wall motion, why functions \( \varepsilon_{33}^T (E) \) and \( d_{31} (E) \) under zero bias behave in a similar manner as follows from the experimental data. In general, 90° and tetragonal/rhombohedral polarization reorientation should not necessarily contribute equally to the change in the dielectric and piezoelectric functions.

Soft piezoelectric PZT ceramics consist of numerous grains with extremely complicated domain structure and possess very high electromechanical coupling. Thus, the application of electric field to the ceramics can cause a change in the internal mechanical stress acting on a microdomain since this microdomain and its surrounding matrix may have a different orientation of their polar axes. This change in the mechanical stress, in turn, can affect macroscopic electromechanical characteristics of the ceramics. Therefore, a model can be suggested in which the observed changes in the electromechanical properties of the ceramics are attributed to the change in the microscopic mechanical stress acting on tetrag-
onal and rhombohedral microdomains because of the applied electric field. The stress is a result of the interaction of a microdomain with surrounding grains and domains. Hysteresis behavior of the electromechanical characteristics is probably caused by the hysteresis nature of this interaction since mechanical properties of the surface of grains differ from those of the microdomains. Calculations based on the thermodynamic theory of the ferroelectric state of PZT ceramics with composition on the morphotropic phase boundary show that compressive and tensile stress on the order of 10–20 MPA can cause a corresponding change in electromechanical properties of ceramics on the order of 10–40%.

This value of the mechanical stress as well as corresponding changes in piezoelectric, dielectric, and elastic properties are quite reasonable for the soft PZT ceramics. This model is applicable only if the amplitude of the electric field is smaller than the coercive field. For fields greater than the coercive field, polarization switching should dominate (Fig. 19). It should be noted that the effect of the internal static mechanical stress was suggested as being responsible for the high dielectric permittivity of fine-grained BaTiO₃ ceramics and the theory was proved correct experimentally. Nevertheless, at the present level of knowledge of the physical properties of soft PZT ceramics, it is impossible to determine which model—that of the domain wall motion, that of the internal stress, or that of their combination—is valid.

VI. SUMMARY AND CONCLUSIONS

In this work, experimental and theoretical studies of the behavior of dielectric, piezoelectric, and elastic characteristics of commercial soft piezoelectric PZT ceramics in the sinusoidal electric field applied along the poling direction under mechanical stress-free conditions in the frequency range 100 Hz–10 kHz were conducted. Experimental results demonstrate that amplitudes and phases of the harmonics of relative strain $S_1^{(1)}/E_{10}$ and relative electrical displacement $D_3^{(1)}/E_{10}$ increase similarly with increasing amplitude $E_{10}$, if the amplitude is less than coercive field. An increase in the square of the resonant frequency of electromechanically decoupled bending vibrations is more moderate. These amplitude dependencies are very well described by linear functions except for small electric fields. Relative amplitude of the second harmonic of the electrical displacement $D_3^{(2)}/D_3^{(1)}$, which is polar, also increases almost linearly with increasing $E_{10}$. The relative third harmonic $D_3^{(3)}/D_3^{(1)}$ demonstrates saturation-like behavior. Based on these experimental data a novel model in which the ceramics are considered to behave like a hysteretic transducer, is developed. This approach makes it possible to model hysteretic dependencies of the electromechanical characteristics and to relate directly electric field dependencies of complex piezoelectric coefficient $d_{11}(E_{10})$ and $S_1^{(1)}/E_{10}$, of complex dielectric permittivity $e_{33}(E_{10})$ and $D_3^{(1)}/E_{10}$, and of the amplitude of elastic compliance $s_0^{(1)}(E_{10})$ and $D_3^{(1)}/E_{10}$ to $E_{10}$. The model effectively describes field dependencies of the first three harmonics of $D_i(E_{10})$, as well as the irreversible change in the remnant polarization under pulsed electric field. Several physical models for the observed behavior were analyzed. In one approach, the explanation is based on the 90° ferroelectric domain and tetragonal/rhombohedral phase boundary motion. In order to better explain the experimental results, a new physical model was developed. According to the model, the observed change in electromechanical properties under the electric field is caused by the field dependence of the mechanical stress acting at interdomain boundaries. It is assumed that this stress is caused by the electromechanical interaction of a microdomain with surrounding microdomains and grains. This surrounding media acts as a viscoelastic structure that causes hysteretic behavior in the electromechanical properties.

ACKNOWLEDGMENT

This work was supported by the Office of Naval Research under Contract No. N00014-94-1-1140.

APPENDIX 14
A piezoelectric pseudoshear multilayer actuator

Qing-Ming Wanga) and L. E. Cross
Intercollege Materials Research Laboratory, The Pennsylvania State University, University Park, Pennsylvania 16802-4800

(Received 9 February 1998; accepted for publication 2 March 1998)

A type of high strain piezoelectric ceramic actuator, namely, pseudoshear multilayer actuator, is described. In this structure, a stack of prepoled rectangular piezoelectric transducer ceramic sheets are conductively bonded at alternate ends, while the bottom layer bonded on a fixed base. When driven, alternate layers elongate or shrink in the same direction through converse piezoelectric effect. which results in the actuator structure developing a strong shear motion about the face perpendicular to the bonding direction. Experimental results indicate that more than 50 μm displacement can be achieved from the top layer for an actuator consisting of 18 layers with dimensions of 25.57 mm × 4.02 mm × 0.51 mm (1 × w × t). By reducing ceramic sheet thickness and using more layers, even large displacement can be obtained, and driving voltage can also be reduced while keeping the same field level. Nonlinear piezoelectric response under high driving field further enhances the displacement level. © 1998 American Institute of Physics. [S0003-6951(98)00418-5]

Piezoelectric actuators are usually classified into three categories: multilayer actuators, bimorph or unimorph bend- ers, and flexensional composite actuators. Multilayer actuators, in which about 100 thin piezoelectric/electrostrictive ce- ramic sheets are stacked together with internal electrodes utilizing the longitudinal (d31) mode, are characterized by large generated force, high electromechanical coupling, high resonant frequency, low driving voltage, and quick response, but small displacement level (<10 μm).1 On the other hand, bimorph or unimorph actuators consist of two thin ceramic sheets or one ceramic and one metal sheet bonded together with the poling and driving directions normal to the interface. When driven, the alternative extension and shrinkage of ceramic sheets due to transverse (d33) mode result in a bending vibration. Bimorph and unimorph actuators can gen- erate a large displacement level but low electromechanical coupling, low resonant frequency, and low driving force. Flexensional composite actuator, "moonie",2,3 consists of a piezoelectric or electrostrictive ceramic disk and metal end caps which act as a mechanical transformer converting and amplifying the lateral displacement of ceramic disk into axial motion. Medium displacement and force level can be ob- tained in moonie actuators.

Most recently developed cantilever monomorph, RAINBOW, and shear mode actuators4-7 are basically modified unimorph actuators with advantage of simple monolithic structure by which bonding problems usually observed in bimorph-type actuators can be avoided. However, large dis- placement is achieved with sacrificing the generative force. In this letter, we report a new type of pseudoshear mode actuator which makes use of a composite structure assembled from thin sheet soft piezoelectric transducers (PZT) piezo- electric ceramics, poled in the usual direction normal to the major face of sheet and driven by electrodes upon the major faces.

In pseudoshear multilayer actuator structure, N layer

fully electroded piezoelectric ceramic plates are bonded to- gether by a stiff conductive epoxy at alternate ends of the plate structure as depicted in Fig. 1(a). The bottom layer is bonded on fixed base with one end. The gaps between the nonbonded faces of the sheets are filled by thin plastic sheets or teflon coatings to stabilize the structure. All the ceramic plates have the same normal poling direction and they are connected electrically in parallel through the conductive epoxy bond line with the wires contacting alternate pairs of plates from each side. For PZTs, the piezoelectric d31 coeffi- cient is negative. Tracing now the effect of a voltage applied between the two contact wires of sign shown in Fig. 1(b). The bottom plate is subjected to a field against the original poling direction, i.e., a negative field thus, since d31 is negative the resulting linear piezoelectric strain is

\[ x_1 = -d_{31}(-E_3) = d_{31}E_3. \tag{1} \]

If L is the length dimension of ceramic stack, the elongation or shrinkage of each layer is

\[ d_1 = d_{31}E_3L. \tag{2} \]

![FIG. 1. Schematic drawing of pseudoshear actuator structure.](image-url)
Because ceramic plate are bonded at its right end, the left end is free and then moves to the left by a distance \(d_1\).

For the second plate in the stack, however, a positive field is applied along the poling direction, thus plate 2 shrinks so that

\[
d_2 = -d_3 E_3 L.
\]

(3)

Now the right hand end of plate is free to move which will move a distance \(d_2\) also to the left, thus its total displacement, combining with plate 1, is \(d_1 + d_2\) to the left.

Obviously, each layer is carried by its following layer thus the direction of elongation and shrinkage of each layer are the same. Therefore the displacement of the upper layer is cumulative from its lower layers. For a stack of \(N\) plates, the total displacement of its top layer, with respect to base, is

\[
d = d_1 - d_2 + \ldots + d_N = N d_3 E_3 L
\]

(4)

at the free end. or

\[
d = d_1 - d_2 + \ldots + d_{N-1} = (N-1) d_3 E_3 L
\]

(5)
at bonded end.

If the voltage is reversed to the stack as shown in Fig. 1 (c), the actuator will now sheave over upper surface moving to the right. Therefore under the ac field, this multilayer actuator vibrates similar to pure shear vibration can thus be called pseudoshear multilayer actuator.

From Eqs. (4) and (5), it can be found that the total displacement of the top layer depends on the number of ceramic layer \(N\), the value of \(d_3\), the magnitude of electric field \(E_3\), and also the length of actuator \(L\). It is interested to note that, due to \(E_3 = V/t\), where \(V\) is the applied voltage and \(t\) is thickness of ceramic layer, the driving voltage can be reduced by reducing the thickness of ceramic layer while keeping the same field strength.

Assuming the space between each layers in the actuator stack is \(t_1\), the total stack thickness will be \((t + t_1)N\) and the shear angle or shear strain \(\delta\) will be

\[
\delta = \frac{N d_3 E_3 L}{t + t_1} = \frac{d_3 E_3 L}{t + t_1}.
\]

(6)

since \(\delta \ll L\), the stack can be designed to generate large shear angles.

From Eq. (6), we may introduce an effective shear coefficient \(d_{15}^*\) by comparing with pure shear strain, \(\delta = d_{15}^* E_3\).

\[
d_{15}^* = \frac{d_3 L}{t + t_1},
\]

(7)

which is a dimension-dependent coefficient.

To examine the performance of a pseudoshear actuator, an 18 layer stacked prototype actuator with dimensions of 25.57 mm \(\times\) 4.02 mm \(\times\) 0.51 mm \((1 \times w \times t)\) of each layer is prepared using commercial Motorola soft PZT (3203HD) ceramics. In the multilayer structure, all layers are aligned in the same poling direction. Silver epoxy (E-Solder\textsuperscript{9} No. 3021 Conductive Adhesives, IMI Insulation Materials, Inc., New Haven, CT) was used to bond the alternative ends of ceramic sheets. Very fine copper wires were used to connect the electrodes in such a way that each layer was connected electrically in parallel. Thin plastic sheets of about 0.1 mm thickness or silicon grease were used to fill the gaps between each layer to stabilize the structure. A load of about 1 kg was applied on the top of multilayer structure when hardening the silver epoxy.

To characterize the actuator performance, the displacement of the top layer of pseudoshear actuator was measured as a function of driving electric field. The measurement setup and procedure were described in a previous letter.\textsuperscript{7} To check the various vibration modes, impedance spectrum of actuator was measured by using an impedance/gain-phase analyzer (HP4194A, Hewlett Packard Co.).

It should be noted that for nonlinear piezoelectric materials such as PZTs, piezoelectric coefficient is electric field dependent. The data provided by material manufacturers are usually measured under weak field by resonance-antiresonance method, and thus are not applicable for the actuator performance evaluation because strong electric fields are usually applied to drive the actuators. Therefore, the piezoelectric \(d_{31}\) coefficient of Motorola PZT 3203 HD ceramics was measured as a function of electric field. A rectangular sample with dimensions of 38.0 mm \(\times\) 12.5 mm...
A sinusoidal driving field with frequency of 50 Hz is used to excite the sample. The effective $d_{31}'$ coefficient was obtained by $d_{31}'' = x_1/E_1$, here $x_1$ is transverse strain under electric field $E_1$. The effective piezoelectric $d_{31}'$ coefficient as a function of electric field $E_1$ is depicted in Fig. 2. It can be seen that due to its nonlinear ferroelectric nature, $d_{31}$ is no longer a constant over the used electric field range for soft PZT ceramics. As the magnitude of applied electric field increases, $d_{31}'$ coefficient increases. The inset in Fig. 2 shows the transverse strain under weak electric field. It is clear that linear piezoelectric relation between strain and electric field, $x_1 = d_{31}' E_1$, is obeyed only at a rather low field level. The measured low field level $d_{31}'$ coefficient is $-318 \times 10^{-12}$ m/V, which is in good agreement with the data provided by the manufacturer's data sheet. $-320 \times 10^{-12}$ m/V, measured by resonance-antis resonance method.

It has been shown\textsuperscript{8,9} that the nonlinear piezoelectric response in ferroelectric PZT ceramics under high driving field originates mainly from the motion of non-180° domain walls. Experimental results also showed that dielectric losses drastically increase with the amplitude of driving field, since the non-180° domain wall motion is inherently a lossy process. But no significant temperature increase was observed when frequency is far off the resonant frequency of the ceramic sample. So for off-resonant frequency driving, soft PZT can be driven under quite a high electric field level. However, the nonlinear piezoelectric response of PZT ceramics has to be considered in the actuator design.

Figure 3 shows the displacement as a function of electric field for the prototype pseudoshear actuator composed of 18 layer PZT ceramic sheets. More than 50 μm shear displacement was obtained under a driving field of 270 V/mm. A nonlinear piezoelectric response under high electric field contributes significantly to the shear displacement. Considering the thickness of ceramic sheet and the space between layers ($l = l_s = 0.61$ mm), we can calculate the effective piezoelectric $d_{15}'$ coefficient by using Eq. (7). Under weak field, since $d_{31}' = -318 \times 10^{-12}$ m/V, we have $d_{15}' = 13,380 \times 10^{-12}$ m/V, which is about 13 times higher than the shear mode piezoelectric coefficient $d_{15} (=1045 \times 10^{-12}$ m/V for soft PZT 3203HD) of the ceramic material. It is interesting to note that $d_{15}'$ is a dimension-dependent parameter. By using an even thinner ceramic sheet, $d_{15}'$ coefficient can be further increased. Since $d_{15}'$ increases with a driving field due to nonlinear piezoelectric response, $d_{15}'$ coefficient will also increase as the amplitude of applied ac field increases.

frequency characteristics of pseudoshear actuator may be deduced from the impedance spectrum shown in Fig. 4. There is no resonance observed below 1 kHz, indicating the displacement of pseudoshear actuator may have flat frequency responses in the low frequency range. Two weak resonances are evident near 1.33 and 3.0 kHz which are probably due to spurious bending modes. It was found that if a load was placed on the top of actuator, these two resonant modes were greatly reduced or even disappeared. By making more compact structure through better selection of the polymer filler material and better bonding processing, the two peaks could be eliminated. The resonance at 10 kHz is the first extended longitudinal mode which in pseudoshear con-figuration will now represent the first effective shear mode. A second strong mode at 26.8 kHz is probably a shifted third harmonic. Weaker higher order modes are evident at 53 and 90.6 kHz.

In summary, a new type of piezoelectric actuator, namely, pseudoshear actuator is developed. Large displacement was obtained through this multilayer configuration. A nonlinear piezoelectric response further enhances the vibrational level of actuator. The frequency characteristics are also studied which indicated the resonant frequency of effective shear mode is much higher than that of the bending vibration mode of bimorph-type actuators. The pseudoshear actuator has the following advantages: (1) The actuator is assembled from thin sheet PZT already electroded and pre-poled: shear vibration is realized through $d_{31}'$ mode by applying ac voltage through the thickness direction so that high fields may be realized at low driving voltage; (3) the generative force is controlled by stiffness $c_{44}$, which is larger than the shear stiffness $c_{66}$; (4) the magnitude of shear can be controlled by the ratio of $L/t-1$. and can be very large for thin sheets.

The pseudoshear mode actuator may have a very wide range of applications. The systems where it could replace conventional actuators would be linear and rotary motors, step and repeat (inchworm) type actuators, flow sensing and flow control, valving and pumping system, and many others.
APPENDIX 15
Pseudo-shear universal actuator driving flextensional panel diaphragm low frequency acoustic source

Ruibin Liu, Qingming Wang, Qiming Zhang, Valery D. Kugel, L.E. Cross

Materials Research Laboratory
The Pennsylvania State University, University Park, PA 16802

ABSTRACT

Acoustic source used for Active Noise Control (ANC) at low frequency (80-250 Hz) is designed and developed by using a piezoelectric ceramic actuator and a flextensional panel diaphragm. In order to reach the vibration magnitude and radiation area needed for high and flat sound pressure level in the low frequency range, Pseudo-Shear Universal (PSU) actuator has been used as the driving part which is a new type of multilayer piezoelectric actuator originated from MRL offering the advantages of large displacement and high blocking force; on the other hand, Carbon Fiber Reinforced Composite has been used as the diaphragm material which provides a more rigid structure than conventional loudspeaker paper. A prototype device was fabricated which has the following characterizations: 40 layers PSU actuator with a compact dimension: 38 mm x 50 mm x 23.6 mm. Two of them are needed for a device. Diaphragm area is 126 mm x 152 mm. At quasistatic condition (5 Hz) and at the 0.84 kV/cm electric field, 344 μm displacement could be achieved at the apex of the diaphragm resulted from the flextensional amplifying mechanism with an amplification factor more than 11. The sound pressure level in the frequency range 100-250 Hz shows better flat behaviour than the acoustic sources studied earlier such as Double Amplifier and PANEL air transducers which exhibit a significant reduction of sound pressure level in the low frequency range. By a slight modification, it is likely to make this device in a total thickness of 10 -15 mm range. High and stable sound pressure level as well as thin flat structure make it much more competitive in the whole area of applications for low frequency active noise control.

Keywords: piezoelectric actuator, pseudo-shear universal actuator, air acoustic transducer, active noise control.

1. INTRODUCTION

Recently, piezoelectric integrated composite structure has been developed as the air transducer for the purpose of Active Noise Control in the frequency below 1 kHz. Those transducers consist of two amplifying mechanisms to achieve high vibration magnitude. One amplifying mechanism is to make use of the piezoelectric bending actuator. Another amplifying mechanism is utilization of the flextensional amplifying mechanism in the triangle shape diaphragm. Fig. 1 gives a schematic configuration for this type of piezoelectric integrated composite structure used for an air transducer.
There are many bending piezoelectric actuators: such as bimorph, unimorph, RAINBOW, CRESCENT etc. All those bending actuator can in principle be used as the first amplifying driving part. For the diaphragm, low density and high stiffness material is desired. The loudspeaker paper is the most popular diaphragm material. This kind of composite structure composed of those parts possesses the advantages of light weight and panel structure.

Two generations of air transducer prototype devices have been developed since 1994. The first generation was Double Amplifier air transducer. Piezoelectric bimorph was used as driving part and loudspeaker paper was used as diaphragm. Multilayer bimorph was also used to replace the traditional two layers bimorph in this structure to decrease the driving voltage. The displacement of the diaphragm apex can easily reach the millimeter scale. A far field (1m) sound pressure level can be more than 80 dB in the frequency range 300-1000 Hz at a driving electric field 0.7 kV/cm.

The second generation air transducer called PANEL (Piezoelectric Acoustic Noise ELiminator) used the unimorph as the leg part. Between the two legs, a bimorph was bonded to serve as both the driving and the substrate functions. In this second generation, instead of loudspeaker paper, Carbon Fiber Reinforced Composite was used as the diaphragm material. The second generation prototype device is much more rigid than the first generation. PANEL can also produce more than 80 dB sound pressure level in the 300-1000 Hz frequency range. This type of device has been already used for Active Noise Control experiments. An average 10 dB sound pressure level reduction could be achieved.

However, both generation devices have a common drawback: there is a significantly sound pressure level reduction from 250 Hz to 100 Hz. For Double Amplifier air transducer, the falling off in magnitude is 18 dB (driven by a 0.7 kV/cm electric field). For PANEL, the difference is 22 dB (calculated sound pressure level, driven by a 1 kV/cm electric field). This inadequacy will restrict the application of these acoustic sources. In daily life and industry area, there are a lot of low frequency noise sources, such as car engine noise (120 Hz), transformer noise (120 Hz and 240 Hz), turbo prop (70-100 Hz) and MD80/DC-9 (100 Hz). Usually the lower frequency, the more difficult to cancel the noise by conventional methods like absorption and barrier treatments. Thus applying Active Noise Control technology to those low frequency noise sources needs a specific acoustic source which can provide high and stable sound pressure level in the 100-250 Hz frequency range. Theoretically speaking, there are two reasons for the reduction of sound pressure level in low frequency: 1. diaphragm vibration magnitude is not high enough to maintain the sound pressure level; 2. vibrating area is not big enough to achieve the better matching with the air medium.

In this paper, we present a new acoustic source particularly designed for the low frequency (80-300 Hz) application. Piezoelectric PSU actuator is used to replace the piezoelectric bending actuator. Thus big blocking force and large displacement from PSU actuator make it possible to reach higher vibration magnitude and larger area diaphragm operation. Prototype device with a diaphragm area 126mm x 152 mm exhibits a very flat and high sound pressure level in the 100-250 Hz frequency. The mass production of multilayer PSU actuator could be easier and less cost than other multilayer piezoelectric actuator. A small modification will make this device with very thin total thickness, which makes it very promising to be the desired acoustic source for the active noise control in ultra low frequency.

2. PSU ACTUATOR, AIR TRANSDUCER AND EXPERIMENTAL PROCEDURE

2.1 PSU multilayer actuator and air transducer

Conventional multilayer actuator uses a compact volume to realize a large displacement. The principle for its operation is to enable high electric field established through moderate applied voltage by inserting thin layers into the stack.

PSU multilayer actuator uses a compact volume to achieve a high displacement. The mechanism for its operation is to utilize specific design to get large displacement from an expanded total length. This device was originated from the MRL. The design idea and operation principle are demonstrated in Fig. 2.

PSU actuator is made by stacking the rectangular ceramic pieces with the electrode in the surface perpendicular to the thickness. The ceramic pieces are poled along the thickness direction, then bond the neighbouring piece at the side edge in an alternative opposite cyclic manner. The ceramic pieces are connected electrically parallel but for the neighbouring pieces the electric polarity is opposite.
When voltage is applied to the ceramic plate, a transverse strain will happen through the $d_{31}$ piezoelectric effect to every piece of ceramic. If the bottom piece expands, the second piece to the bottom will contract. The displacement of the second layer free end will be the sum of secondary layer volume strain plus the first piece volume strain. The shear displacement will direct to the right. The rest layers behaviour will be deduced by analogy. The displacement of top layer will be the cumulation of all layers. When the voltage polarity changed, top layer displacement will direct to left. The free end displacement of top layer will be determined by the equation:

$$\delta = N \times d_{31} \times E \times l$$

Where $\delta$ is the top layer displacement, $N$ is the number of layers, $d_{31}$ is the piezoelectric coefficient, $E$ is applied electric field, $l$ is length of ceramic plate.

When the AC voltage is applied to the device, left and right horizontal vibration will happen, which can be used as the driving element to drive the flexensional diaphragm. Because the displacement is resulted from volume deformation effect, the blocking force of PSU actuator should be bigger than that of bending actuator. In order to make the diaphragm more rigid and improve the frequency behaviour, Carbon Fiber Reinforced Composite was chosen as the diaphragm material. The Carbon Fiber Reinforced Composite has a Young's modulus 190 GPa and the loudspeaker paper only has a Young's modulus 0.55 GPa.

### 2.2 Experimental procedure

Piezoelectric ceramics used in this device were provided by Material System Inc., which is PZT-5H composition. Ceramic plate with a dimension 38 mm x 50 mm x 0.5 mm was silver electroded on the 38 mm x 50 mm surface. There is a 3.8 mm width margin on the right side of upper surface and same width margin on the left side of the bottom surface. The samples were poled before shipment. Measured $d_{31}$ for the 10 mm x 10 mm small cut piece is around $600 \times 10^{-12}$ C/N. In order to avoid the depolarization, all the curing processing procedures were carried out at the room temperature. J-B weld epoxy (JB WELD CO.) was used for the side edge bonding among ceramics and diaphragm hinges as well as mechanical connecting bonding between diaphragm and actuator. Silver epoxy (E-solder No 3021 conductive adhesives, IMI Insulation Materials
Incorporated, New Haven, CT) was used for electric connecting among the ceramic pieces. Fabrication procedure is shown in Fig. 3.

Fig. 3 Fabrication procedure for air transducer

To characterize the device properties, the displacements of diaphragm apex and of the top layer of actuator were measured by MTI 2000 Fotonic Sensor System (MTI Instrument, Inc.). MTI-2000 Fotonic sensor can perform the displacement and vibration measurement with the advantages of no contact to the target, high accuracy, wide frequency range and insensitive to the electric and magnetic fields by using the fiber-optic technology. A high reflectivity mirror is needed to stick to target with the face perpendicular to the optical probe. The sensor probe was placed at a three dimensional movable micropositioner (Ealing Electro-optic, Inc.). Function generator is DS-345 (Stanford Research System, Inc.), which generates the required signal shape and basic magnitude. Then a power amplifier (790 Series, PCB piezotronics) was used to amplify the driving signal. A lock-in amplifier SR-830 DSP (Standford Research System Inc.) was used to measure signal output from Fotonic Sensor. The reference frequency is synchronlized to the function generator. Sound pressure level was measured in an
anechoic room at far field (1 m) by a ND 21 Sound Level Meter (Rion Company, Japan). Fig. 4. shows the schematic diagram for displacement measurement. Fig. 5 shows the schematic view for the sound pressure level measurement.

Fig. 5 Schematic view for the sound pressure level measurement

### 3. EXPERIMENTAL RESULTS OF PROTOTYPE DEVICE

#### 3.1 Properties of unloaded PSU actuator

Each PSU actuator consists of 40 piezoelectric ceramic plates with a dimension as 38 mm x 50 mm x 23.6 mm. Two actuators have a total capacitance of 4.59 μF. Dielectric loss is 3.8% (measured at 1000 Hz). Displacements of top layer at quasistatic and dynamic conditions are given in Fig. 6.

#### 3.2 Quasistatic and dynamic properties of diaphragm loaded device

Carbon Fiber Reinforced Composite diaphragm with a dimension: width: 126 mm, length: 152 mm, height: 6 mm is for the finished device. Vibration behaviours of diaphragm at quasistatic and as a function of frequency were obtained and shown in Fig. 7.

![Fig. 6 Quasistatic (a) and dynamic (b) properties of unloaded 40 layers PSU actuator](image-url)
In order to investigate the relationship between the diaphragm height, different diaphragm material and the resonant behaviour of the device. We measured the dynamic properties of the device with the diaphragm height 6 mm 10mm. We also measured the the dynamic properties of loudspeaker paper diaphragm with a height 8 mm. The results are given in Fig.8.

Fig. 7 Quasistatic (a) and dynamic (b) properties of diaphragm loaded device

Fig. 8 Comparison of dynamic behaviour for different diaphragm height and different diaphragm material

3.3 Sound pressure level at low frequency range
Sound pressure level were measured at the driving electric field 0.24 kV/cm (12 V voltage) on the final product (height 6 mm, carbon fiber reinforced composite). Fig. 9 gives the measured result and calculated sound pressure level at the 1kV/cm driving electric field.

Fig. 9 Sound pressure level for 6 mm height carbon fiber diaphragm device

4. DISCUSSIONS AND SUMMARY

4.1 Comparison of PSU actuator with other types of actuator

In bimorph and unimorph driving air transducer, the actuator wall has the same length with the diaphragm, even though the width of the diaphragm for a cell is only 50 mm, the amplification factor is about 7. In PSU actuator driving air transducer, the length of driving wall is only one third of the diaphragm length. The width of the diaphragm can be 126 mm and amplification factor can be more than 11. This fact demonstrates the PSU actuator has bigger generative force than the bimorph and unimorph. The value of blocking force for the PSU actuator was investigated and will be reported in another paper. We can still increase the blocking force by modifying PSU actuator structure. It is possible to design the monolithic structure PSU actuator whose blocking force would be much bigger than the current one. There are several ways to increase the displacement of PSU actuator: increasing the layers, decreasing the layer thickness in a limited volume, increasing the driving voltage. We can tailor it in an easy way according to different requirements.

Bending actuator such as bimorph and unimorph can reach big displacement, but the blocking force is small, response is slow and the durability is poor because of delamination. In contrast to the bending actuator, the multilayer actuator possesses the advantages of low driving voltage (100 V), quick response (10 m sec), high generative force (100 kgf) and high electromechanical coupling, but the displacement is only in the range of 10 μm. For a long time, scientists were searching for a new type actuator with the properties located in intermediate range between bending and multilayer actuator. A lot of efforts have been done to improve the properties of the piezoelectric bending actuator. Several novel bending actuator, like RAINBOW, CERAMBOW, CRESCENT, d33 bimorph and THUNDER came out. Kugel has defined the figure of merit for bending actuator, which is equal to the displacement times the blocking force and divided by the admittance. From his
comparative analysis, there are some improvements in properties through these efforts, but no breakthrough happened. The gap between the multilayer actuator and the bending actuator is still existing.

The appearance of "Moonie" and "Cymbal" were a great progress of the efforts. Intermediate properties can be provided by such metal and ceramic composite structures. The invention of PSU actuator provides another kind of actuator offering versatile intermediate properties. The blocking force of which can reach to 100 µm very easily and the blocking force of the pseudo-shear structure is at least several times that of bending actuator. Another advantage for PSU actuator is without the bonding problem between metal and ceramic. The different thermal expansion coefficient between the metal and ceramic in bonding will give rise to very large internal stress, even close the fracture strength of the ceramic. In PSU actuator, we can even design monolithic structure to eliminate the bonding part.

4.2. Sound pressure level in low frequency

According to the theory of acoustics, for the direct-radiator diaphragm at the low frequency (low frequencies is when \( k\alpha < 1.0 \), where \( k = \omega / c \) is the wave number, \( \alpha \) is the diaphragm dimension), a piston whose diameter is less than one-third wavelength (\( k\alpha < 1.0 \)) is essentially nondirectional at low frequency. Hence, we can approximate it by a hemisphere whose rms volume velocity equals \( U_p = \frac{5}{6} U_{rms} \), where \( S_D \) is the projected area of the loudspeaker cone, \( U_r \) is the diaphragm surface velocity.

\[
Pr = \frac{U_c}{f\rho_0} \frac{f\rho_0}{r}
\]

Where \( \rho_0 \) is the average density of the medium, \( f \) is the frequency.

From the equation, we can see the sound pressure level is proportional to the vibration displacement, to the projected area, and to the square of the frequency. So if we want to keep the sound pressure level at the same magnitude when the frequency decreases from 200 Hz to 100 Hz, we need four times displacement magnitude or we have to increase the projected area 4 times. So it is important to have large displacement and large vibration area to keep high sound pressure in low frequency.

Sound is produced by vibrating surface such as the diaphragm. In addition to the energy required to move the vibrating surface itself, energy is radiated into the air by the diaphragm. Part of this radiated energy is useful and represents the power output of the acoustic source. The remainder is stored (reactive) energy that is returned to the generator. Consequently, the acoustic impedance has a real part, accounting for the radiated power, and an imaginary part, accounting for the reactive power. We need the \( k\alpha \) larger than some value for example, 0.5, the real part of the impedance can have enough value to radiate the sound power. So increasing the radiation area is very important for increasing the volume velocity and improving the impedance matching.

The resonant frequency of 6 mm height diaphragm is 55 Hz for first mode and the 140 Hz for second mode. Those two resonant peaks provide high displacement in the operating frequency range (Fig. 8).

So we can conclude that three factors are mainly attributed to the high and stable sound pressure level in low frequency for the current device: the first is the large displacement, the second is the large vibration area and the third is the resonant behaviour during this frequency range.

Further experiment will be emphasized on the following respects:

Understanding the resonant behaviour is very important for the device design. The resonant frequency decrease with the decreasing of diaphragm height for the same diaphragm materials. Although the material properties are very different between the Carbon Fiber Reinforced Composite and loudspeaker paper, the resonant behaviours are almost similar in the investigation (Fig 8). There is further work under way to get more understanding of the resonant mechanism.
In the prototype device, the polarization direction of the ceramic plate is vertical to the diaphragm face, which limits the way of increasing the displacement by adding layer number since we do not want the device being too thick. With a small modification, we can make the polarization of the ceramic piece parallel to the diaphragm face. It will dramatically reduce the total thickness of the device and more room can be used to increase the displacement by increasing the layer number. It is highly possible to reduce the total thickness of the device to the thickness 10-15 mm, which is much useful for application where only a limited space of locating secondary acoustic source is available.

In summary, the acoustic source presented in this paper can produce high and stable sound pressure level in the frequency range 100-250 Hz. The integrated composite of PSU actuator and flexensional diaphragm is a useful way to get high and stable sound pressure level in low frequency because large displacement and large radiation area can be obtained by PSU actuator driving. This kind of acoustic source also offers the advantages of panel structure and thin dimension which make it very attractive in the application of ultra low frequency Active Noise Control.

5. ACKNOWLEDGMENT

This work was supported by the Office of Naval Research under the contract N00014-94-1-1140.

6. REFERENCES

6. L. E. Cross and Qingming Wang, U.S. patent has been applied for.
12. Leo. L. Beranek, Acoustics, p183, the American Institute of Physics, Inc., 1986
13. Leo. L. Beranek, Acoustics, p116, the American Institute of Physics, Inc., 1986
APPENDIX 16

Comparison of two structural sensing approaches for
Active Structural Acoustic Control.

Julien P. Maillard
Chris R. Fuller

Vibration and Acoustics Laboratories
Mechanical Engineering Department
Virginia Polytechnic Institute and State University
Blacksburg, Virginia 24061-0238

Received
Abstract

A numerical study comparing the use of two structural sensing approaches for sound radiation control is performed on a baffled rectangular plate. The first sensing approach implements an array of accelerometers whose outputs are filtered to construct an estimate of the sound pressure radiated at given angles in the far field. The second method uses the same array of point sensors to estimate the net volume acceleration of the plate. Results show the improved performances of the sensor observing far-field sound radiation over a volume acceleration based sensor.

PACS Numbers: 43.40.At, 43.40.Rj, 43.60.Gk
Introduction

An important issue in the active control of structurally radiated noise remains the design and implementation of error sensors. The early works on Active Structural Acoustic Control (ASAC) [1] were concentrated on the design of structural actuators [2] while the error signals to minimize were provided by microphones located in the far field. For the last couple of years, the research in ASAC has made some progress in replacing error microphones by error sensors integrated in the structure [3]. Unlike microphones which directly measure the quantity to be minimized, i.e., acoustic pressure, structural sensors only measure information related to the structural vibrations. The goal is then to design a structural sensor whose output is strongly related to radiated sound pressure so that minimizing the magnitude of the sensor output signal will also minimize the total radiated sound power. In other words, an efficient structural sensing technique should take into consideration the structure-fluid interaction such that only the radiating part of the structural vibrations is observed by the sensor. Indeed, minimizing both radiating and non-radiating components of the vibrations typically requires more control effort.

Spatially distributed structural sensors such as polyvinylidene fluoride (PVDF) materials have shown potential for observing the radiating part of the structural vibrations. In modal sensing, the PVDF film is designed to observe the dominant radiating modes among those found in the structural response [4–6]. The concept of “radiation modes” has also been used in designing error sensors [7, 8]. In particular, several authors have demonstrated the use of PVDF sensors designed to observe the first and dominant radiating of the radiation modes of a rectangular plate, i.e., the piston mode [8–11]. This mode can be detected by measuring the net volume displacement over the surface of the structure and specially designed
PVDF sensors have been constructed for this task [8–11].

Recently, a sensing technique referred to as Discrete Structural Acoustic Sensing (DSAS) was demonstrated both analytically and experimentally on baffled planar radiators [12–14]. The technique implements an array of structural point sensors (usually accelerometers) whose outputs are passed through digital filters to estimate in real time the far-field radiated pressure in a given direction, or equivalently, a given wave-number component, over a broad frequency range. The digital filters are usually Finite Impulse Response (FIR) filters designed to model the appropriate Green's function associated with each point sensor and far-field locations. It was shown that only a low number of point sensors is needed to provide accurate sound radiation estimates over a bandwidth encompassing the first few modes of a rectangular plate. Moreover, the sensor's output can be shifted in time such that the acoustic path delay present in the Green's functions is removed. This allows the use of only a few coefficients in the sensor FIR filter array and makes its implementation on a Digital Signal Processor (DSP) very efficient in terms of computational load.

A simplified version of the above approach that has been suggested consists in replacing the radiation Green's functions by a unity transfer function. The sensor output then simply becomes the sum of the acceleration signals. The resulting error information thus represents an estimate of the net volume acceleration of the structure. It is therefore somewhat equivalent to the PVDF volume velocity sensor referred above except it is implemented in discrete form. This alternative sensing approach will be referred to as Discrete Structural Volume Acceleration Sensing (DSVAS). For planar radiators, this error information is also equivalent to measuring the far-field pressure in the direction normal to the plane of the radiator [15].

This letter presents a short numerical study comparing the performance of
the two sensing approaches introduced above, i.e., DSAS and DSVAS methods, in Active Structural Acoustic Control (ASAC) applied to a rectangular baffled plate. The plate is excited by a normal point force over a bandwidth encompassing its first 9 flexural modes. The control is achieved through two single-sided piezoelectric patches. The system modeling is briefly described in the first two sections along with its physical characteristics. The third section then presents the control performance in terms of mean-square velocity and radiated power attenuation obtained for various sensing configurations.

I. System modeling

The numerical model describing the structural response of the plate excited by point forces and piezo-electric patches is briefly presented. In this model, the mechanical displacements and electrical fields generated in the piezo-electric inclusions are fully coupled. In order to solve for the dynamics of this coupled electro-mechanical system, a Rayleigh-Ritz formulation is derived based on the generalized form of Hamilton’s principle for coupled electro-mechanical systems [16]. This energy-based method also allows to easily model arbitrary boundary conditions applied along the edges of the plate. The plate mid-plane displacements along the two longitudinal and transverse directions are approximated by polynomial series where the unknown polynomial amplitudes are solution of a linear system. After solving this system at each frequency of interest, various structural quantities, such as mean-square velocity, or point acceleration, can be computed. For a complete derivation of this approach applied to rectangular plates, the reader is referred to the work of Berry [17].

The far-field sound pressure radiated from the plate is obtained from the dis-
placement amplitudes by computing the continuous wave-number transform of
the acceleration distribution over the frequency bandwidth. The radiated power
is estimated from the radiation impedance matrix associated with the polynomial
basis used to approximate the structural response. This approach is more efficient
than simply integrating the far-field pressure over the hemisphere surrounding the
plate. More details can be found in [18].

The output from the discrete structural acoustic sensor is computed from the
structural acceleration evaluated at nine point sensor locations on the plate and
multiplied by the appropriate sensor transfer functions as described in [14]. Note
that the sensor transfer functions used in this numerical study correspond to the
ideal case of perfect filter modeling. It was shown in previous work however that
these transfer functions are easily modeled to a high level of accuracy with only a
few coefficients per FIR filter [12].

The controlled structural response is obtained by adding to the uncontrolled
response due to the point force excitation, the control field response due to the
two one-sided piezo-electric patches. The optimal control voltage to each PZT
is computed using standard Linear Quadratic Optimal Control theory (see for
example [19]), where the cost function to be minimized is a quadratic function
of the control voltage amplitudes. It should be noted that this frequency domain
optimal controller does not take into account the causality constraint inherent
to a real time implementation on a DSP. In other words, the optimal controller
transfer functions in the frequency domain might be acausal and therefore cannot
be accurately reproduced by a digital filter. This results in overestimating the
attenuation achieved by the controller however it does represent the maximum
achievable attenuation if the system is causal. Also, the limitation due to the
finite dynamic range of the analog to digital converters is not included in this
model.
As mentioned above, the plate is excited by a point force while two PZT patches are implemented to achieve control. Preliminary tests showed that a single piezo-electric actuator was not sufficient to achieve global sound radiation attenuation over the entire frequency bandwidth. The sensor based on unit transfer functions, i.e., estimating the structural net volume acceleration, provides one single error signal. To facilitate the comparison with the sensor based on far-field pressure estimates, only one far-field direction is considered. This control layout, i.e., two control outputs and one error input, results in an under-determined system which is solved by including in the cost function the weighted control effort (sum of the square modulus of each control voltage) as an additional quantity to minimize, effectively constraining the solution. This second variable is scaled down (by varying its weighting) so that the main forces of the control are directing towards minimizing the error information rather than the control voltages.

II. System characteristics

The system studied here is a model based on the plate tested experimentally in a previous paper [14]. The translation and rotation stiffness factors along each side of the plate were numerically adjusted in an effort to match the experimental natural frequencies. The resulting modeled boundary conditions were close to the simply-supported case. The plate geometry is shown in Figure 1 along with the location of the normal point force disturbance, the two PZT control inputs, and the nine accelerometers implemented in the discrete structural acoustic sensor. The physical parameters and dimensions for the plate and piezo-electric patches are shown in Table 1. The point force and PZT center locations are presented in Table 2. The control actuator locations in Table 2 were optimally chosen for
minimizing the (3,1) and (1,3) flexural modes, while the disturbance force location ensures all nine modes present in the bandwidth of interest are excited. Finally, the model's natural frequencies are shown in Table 3.

The structure's response is computed over the frequency bandwidth, 50 to 650 Hz, with a one Hertz increment. Note that due to the small level of damping included in the system (see Table 1), the peak levels of the response at those frequencies is likely to be significantly below the actual resonance level reached at the resonance frequency.

III. Control performances

Figure 2 shows the sound radiated power plotted versus frequency for the uncontrolled plate (solid line) and the controlled plate using the two sensing approaches described earlier. The dashed line corresponds to the DSAS approach (pressure estimate based sensor) and the dotted line to the DSVAS approach (net volume acceleration based sensor). The direction of minimization for the DSAS approach was set to \((\theta = 36^\circ, \phi = 30^\circ)\). This direction was chosen so that it does not coincide with any of the nodal lines of the radiation directivity of the modes present in the bandwidth. In other words, the sensor based on far-field pressure estimate observes the radiation of all the flexural modes of the plate.

The first sensing approach of DSAS achieved good control over the entire bandwidth with very small amount of spillover. The largest level of attenuation is obtained for mode (1,1) around 86 Hz. Very good attenuation is also seen for the other modes present in the frequency bandwidth except for the resonance at 333 Hz corresponding to the (2,2) mode of the plate. Around this frequency, the control system only reduces the radiated power level by about 10 dB. Look-
ing at the pressure spatial directivity of the uncontrolled and controlled plate at
this frequency reveals that the pressure distribution is slightly shifted along the
\( \phi \) direction so that the (2,2) mode radiation nodal lines go through the direc-
tion of minimization \( (\theta = 36^\circ, \phi = 30^\circ) \). In other words, the amplitude of this
mode is not reduced. Indeed, the structural velocity distribution at this frequency
shows hardly any changes between the uncontrolled and controlled responses. This
phenomenon is commonly referred to as modal restructuring. This is a direct con-
sequence of minimizing the sound pressure field in a single direction. To obtain a
larger level of sound power reduction, additional far-field pressure error estimates
in other directions could be included in the control system. Or alternatively, a
single error signal directly related to the radiated power would also provide better
overall sound attenuation.

This second solution motivates the use of a sensor based on net structural
volume acceleration. The resulting error information directly relates to the volu-
metric modes of the radiators, i.e., the odd-odd modes for a plate. These modes
are the most efficient radiators, and the sensor output is therefore directly related
to sound radiated power. This approach corresponds to the dotted line where the
sensor output is the sum of the nine acceleration signals (unit transfer function).
As seen in Figure 2, this second sensing approach performs well near the resonance
frequencies of the odd-odd modes of the plate. On the other hand, the radiated
power around the resonances of odd-even or even-even modes is not significantly
reduced. The global radiation level of the (2,1) mode, for example, even increases
after control. By definition, the net volume acceleration based sensor does not
observe any of the odd-even or even-even modes of the plate. Therefore the sys-
tem is unable to effectively control these modes. However, they still contribute to
the total radiated power, despite their lower radiation efficiency compared to the
odd-odd modes. In fact for some frequencies they can radiate more power than

Another interesting feature of the control performance of the volume acceleration based sensor is the large control spillover occurring around 290 Hz. In this frequency range, the controlled structural response exhibits a velocity distribution which perfectly cancels the summation of the nine acceleration amplitudes. However, this velocity distribution radiates significantly in the far field which explains the control spillover. On examining the radiated power level at the resonance frequency of mode (4,1), the dotted line shows significant reduction even though the symmetry of the velocity distribution associated with the (4,1) mode should cancel the sensor output. This phenomenon is explained by the presence of the two PZT patches in the model resulting in slightly asymmetric modes. These slight differences in magnitude and phase of the acceleration amplitudes measured at the nine location shown in Figure 1 result in a non-zero error signal amplitude. Note that these differences are averaged out when using a higher level of discretization. To illustrate this point, another simulation was performed where the sensor uses a 20 by 20 grid of acceleration point measurements to estimate the net volume acceleration. This level of discretization ensures almost perfect estimate of the net volume acceleration in the frequency bandwidth of interest. Figure 3 shows the radiated power before (solid line) and after control using the previous 3 by 3 sensor (dashed line) and the 20 by 20 sensor (dotted line). The 20 by 20 sensor yields a much more accurate estimate of the net volume displacement. This is apparent around 558 Hz where the (4,1) mode is not observed by the sensor and results in no radiated power attenuation. Also, the system's behavior noticed in the case of the 3 by 3 sensor around 290 Hz no longer appears with the 20 by 20 sensor. In conclusions, even though the 3 by 3 sensor might yield larger global attenuation in some cases such as at a frequency around 558 Hz, a higher level of discretization ensures a better estimate of the net volume acceleration and thus...
enable the controller to observe all volumetric velocity distribution. Note that a PVDF continuous sensor designed to observe the same information is therefore likely to perform better compared to a 3 by 3 sensor. However, several authors have shown the difficulties of designing such a sensor with the required accuracy, especially for two-dimensional radiators [9, 11].

To summarize the above results, the reduction levels for the total mean-square velocity and sound radiated power computed over the frequencies of interest are shown in Table 4. The first line corresponds to an additional simulation case where the far-field pressure error estimate (sensor output) is replaced by the actual far-field pressure evaluated in the same direction. Note that this configuration nearly yields the same level of reductions as the pressure estimate structural sensor (DSAS). This demonstrates the ability of the sensor to replace far-field measurements as it was shown in previous work. On the other hand, the sensors based on the net volume acceleration (DSVAS) yield small level of overall reduction (10 dB for the 3 by 3 sensor and 8 dB for the 20 by 20 sensor) when compared to the performance of the DSAS sensor. The higher levels of attenuation achieved by the 3 by 3 sensor is consistent with the curves of Figure 3 discussed earlier. Despite strong attenuation levels at the resonance frequencies of the odd-odd modes, the overall reduction level remains small due to the remaining peaks in the controlled response of the odd-even and even-even modes.

IV. Conclusions

The Discrete Structural Volume Acceleration Sensing (DSVAS) approach, i.e., based on unit transfer functions, performs well in the low frequency range below the resonance of the first non-volumetric mode of the plate. Due to its simplicity,
it is the recommended method in this frequency range. At higher frequencies, the 
Discrete Structural Acoustic Sensing (DSAS) approach, i.e., based on radiation 
transfer functions, provides much increased performance due to the fact that it 
observer radiation from all modes. Thus near and above the resonance frequency 
of the first non-volumetric mode, DSAS is the recommended sensing approach for 
ASAC.

Acknowledgments

The financial support of the Office of Naval Research, Dr Kam Ng, Technical 
Monitor for this work is gratefully acknowledged.

References

[1] C. R. Fuller. Active control of sound transmission/radiation from elastic 
plates by vibration inputs. I. Analysis. *Journal of Sound and Vibration*, 

radiated noise using piezoceramic actuator. In *Proceedings of Inter-Noise 89*, 


List of Figures

1  Plate geometry and transducer locations ............................................. 16
2  Radiated power ..................................................................................... 17
3  Radiated power ..................................................................................... 18
Figure 1: Plate geometry and transducer locations
Figure 2: Radiated power
Before control
3 by 3 DSAS after control
20 by 20 DSVAS after control

Figure 3: Radiated power
List of Tables

1. Plate and PZT's dimensions and material properties ........................................ 20
2. Transducer location ................................................................................................. 21
3. Plate natural frequencies .......................................................................................... 22
4. Total mean-square velocity and radiated power attenuation .................................... 23
Table 1: Plate and PZT's dimensions and material properties

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Plate</th>
<th>PZT</th>
</tr>
</thead>
<tbody>
<tr>
<td>length in x axis (mm)</td>
<td>380</td>
<td>38</td>
</tr>
<tr>
<td>length in y axis (mm)</td>
<td>298</td>
<td>32</td>
</tr>
<tr>
<td>thickness, $h$ (mm)</td>
<td>1.93</td>
<td>0.1905</td>
</tr>
<tr>
<td>Young’s modulus (N/m²)</td>
<td>$2.04 \times 10^{11}$</td>
<td>$6.1 \times 10^{10}$</td>
</tr>
<tr>
<td>Poisson ratio</td>
<td>0.28</td>
<td>0.33</td>
</tr>
<tr>
<td>mass density (kg/m³)</td>
<td>7700</td>
<td>7750</td>
</tr>
<tr>
<td>hysteretic damping factor</td>
<td>0.0005</td>
<td>0</td>
</tr>
<tr>
<td>$d_{31}$ constant (m/V)</td>
<td>-----------</td>
<td>$171 \times 10^{-12}$</td>
</tr>
</tbody>
</table>
Table 2: Transducer location

<table>
<thead>
<tr>
<th>transducer type</th>
<th>center coordinate, $x/L_x$</th>
<th>center coordinate, $y/L_y$</th>
</tr>
</thead>
<tbody>
<tr>
<td>disturbance point force</td>
<td>-0.60526</td>
<td>-0.56376</td>
</tr>
<tr>
<td>control PZT 1</td>
<td>0.41579</td>
<td>-0.45302</td>
</tr>
<tr>
<td>control PZT 2</td>
<td>0.41579</td>
<td>0.38926</td>
</tr>
</tbody>
</table>
Table 3: Plate natural frequencies

<table>
<thead>
<tr>
<th>Mode (m, n)</th>
<th>Natural frequency (Hz)</th>
</tr>
</thead>
<tbody>
<tr>
<td>(1,1)</td>
<td>86.0</td>
</tr>
<tr>
<td>(2,1)</td>
<td>181.2</td>
</tr>
<tr>
<td>(1,2)</td>
<td>241.4</td>
</tr>
<tr>
<td>(2,2)</td>
<td>332.7</td>
</tr>
<tr>
<td>(3,1)</td>
<td>339.1</td>
</tr>
<tr>
<td>(3,2)</td>
<td>484.6</td>
</tr>
<tr>
<td>(1,3)</td>
<td>494.5</td>
</tr>
<tr>
<td>(4,1)</td>
<td>557.7</td>
</tr>
<tr>
<td>(2,3)</td>
<td>580.7</td>
</tr>
<tr>
<td>(4,2)</td>
<td>695.5</td>
</tr>
</tbody>
</table>
Table 4: Total mean-square velocity and radiated power attenuation

<table>
<thead>
<tr>
<th>Attenuation (dB)</th>
<th>Mean-square velocity</th>
<th>Radiated power</th>
</tr>
</thead>
<tbody>
<tr>
<td>Error mic.</td>
<td>17.7</td>
<td>23.8</td>
</tr>
<tr>
<td>DSAS</td>
<td>17.1</td>
<td>23.8</td>
</tr>
<tr>
<td>DSVAS (3 by 3)</td>
<td>5.5</td>
<td>10.1</td>
</tr>
<tr>
<td>DSVAS (20 by 20)</td>
<td>3.9</td>
<td>8.2</td>
</tr>
</tbody>
</table>
APPENDIX 17
Advanced time domain wave-number sensing for structural acoustic systems. Part III. Experiments on active broadband radiation control of a simply supported plate

J. P. Maillard and C. R. Fuller

Vibration and Acoustic Laboratories, Mechanical Engineering Department, Virginia Polytechnic Institute and State University, Blacksburg, Virginia 24061-0238

(Received 23 November 1994; revised 12 May 1995; accepted 21 June 1995)

The present work gives further developments and experimental testing of a new time domain structural sensing technique for predicting wave-number information and acoustic radiation from vibrating structures. Most structure-borne active sound control approaches now tend to eliminate the use of microphones located in the far field by developing sensors directly mounted on the structure. In order to reduce the control authority and complexity required to minimize sound radiation, these sensors should be designed to provide error information that is solely related to the radiating part of the structural vibrations, e.g., the supersonic wave-number components in the case of planar radiators. The approach discussed in this paper is based on estimating supersonic wave-number components coupled to acoustic radiation in prescribed directions. The spatial wave-number transform is performed in real time using a set of point structural sensors with an array of filters and associated signal processing. The use of the sensing approach is experimentally demonstrated in the time domain LMS active control of broadband sound radiated from a vibrating plate. Comparisons of the control performances obtained with the wave-number sensor and error microphones in the far field show that only a few point sensors are required to provide accurate radiation information over a broad frequency range. The approach demonstrates good broadband global control of sound radiation. © 1995 Acoustical Society of America.

PACS numbers: 43.40.Dx, 43.40.Vn

INTRODUCTION

Over the past few years, much research has been conducted in the area of active control applied to structure-borne sound. In active structural acoustic control (ASAC), the minimization of sound radiation is achieved by applying oscillating force inputs directly to the structure rather than by exciting the acoustic medium with loudspeakers (active noise control). The technique can often produce global far-field attenuations with relatively few actuators as compared to active noise control. In early studies, Fuller has shown both theoretically and experimentally the use of point forces as control actuators. More recent works have demonstrated the potential of multiple piezoelectric actuators to replace shakers and realize a more compact or “smart” structure.

One of the primary concerns in active control of sound is choosing the appropriate sensor in order to provide the control system with “error” information. Error microphones located in the far field have yielded good results, since the quantity to minimize, i.e., acoustic power radiated from the structure, is directly related to the far-field pressure. However, the microphone solution is often impractical in real applications and, in an attempt to make the system more compact, the current research tends to develop radiation sensors that are mounted on the structure. One possible approach is to design structural error sensors in order to minimize the vibration: over the entire radiating surface (active vibration control). Such an approach would obviously yield sound attenuation. However it will require, in many cases, a great number of control inputs. A more practical sensing technique should take into consideration the structure/fluid interaction so that only the radiating part of the structural vibrations is observed by the sensor. The principal advantage of this approach is that it allows “modal restructuring” in the control mechanism; in some cases, the residual response is not attenuated but rather the structure is forced to behave like an inefficient radiator. As a result, the control authority and number of channels required to achieve sound attenuation is reduced. Clark and Fuller have discussed the use of polyvinylidene fluoride (PVDF) thin film modal sensors in sound radiation control for rectangular radiators below the critical frequency. With appropriate shapes and locations, PVDF sensors observe only those structural modes that efficiently radiate to the far field.

For planar radiators, the radiating part of the structural vibrations corresponds to the supersonic region of the wave-number spectrum where the structural wave number is smaller than the acoustic wave number in the surrounding medium. Therefore, a possible approach is to build a wave-number sensor that selects and estimates those supersonic wave-number components. Fuller and Burdisso previously showed the potential of a wave-number domain controller based on this type of error information. For implementation with broadband control algorithms, the wave-number estimate must be performed in the time domain. Following these ideas, Maillard and Fuller introduced a new sensing technique that predicts, in the time domain, supersonic wave-number components coupled to acoustic radiation in prescribed directions. In the approach of Maillard and Fuller,
the wave-number transform of the structural acceleration distribution is performed in real time using point sensors with an array of finite impulse response (FIR) filters and associated signal processing. The resulting error signal can be directly used as the error information in time domain control algorithms. The present work gives further development and experimental testing of the method. In particular, this paper experimentally demonstrates the use of such an approach in the active control of sound radiation.

After recalling the theoretical basis of wave-number sensing applied to acoustic radiation previously discussed in Refs. 7 and 8, the sensor is briefly described for practical implementation. Experimental results from wave-number sensing of a simply supported plate are first discussed. The sensor output is compared to the actual wave-number component of the plate in order to evaluate the prediction accuracy. The radiation from the simply supported plate is then controlled using a three-channel feedforward control approach. The system is excited by a bandlimited white noise including the first five bending modes of the plate. Wave-number sensing is compared to the use of error microphones. Results show the ability of the technique to replace far-field acoustic measurements and provide accurate error information over a broadband frequency range. Its use in active structural acoustic control is successfully demonstrated.

I. THEORY

This section briefly reviews the theoretical concepts associated with the proposed sensing technique. For more details, the reader is referred to the two previous companion papers.7,8

A. Far-field pressure and structural wave-number component

Figure 1 shows a two-dimensional planar structure of length \( L_x \) and width \( L_y \), with out-of-plane displacement along the \( z \) axis, \( W(x,y,z) = W(x,y) e^{i\omega t} \). Time harmonic excitation is assumed here. The following results are easily extended to far-field excitations by means of Fourier transforms. The structure is mounted in an infinite baffle, i.e., \( W(x,y) = 0 \) for \( |x| > L_x/2 \) and \( |y| > L_y/2 \). The symbol \( \omega \) represents the angular frequency and \( t \) is the continuous time. The acceleration field is given by \( \ddot{W}(x,y,z) = -\omega^2 W(x,y) e^{i\omega t} \), where \( \ddot{W} \) represents the second time derivative of \( W \). The pressure radiated by the structure, \( P(x,y,z) = P(x,y) e^{i\omega t} \), must satisfy the three-dimensional Helmholtz equation,

\[
\left( \frac{\partial^2}{\partial x^2} + \frac{\partial^2}{\partial y^2} + \frac{\partial^2}{\partial z^2} + k_0^2 \right) P(x,y,z) = 0. \tag{1}
\]

along with the boundary condition that defines the interaction between the structure and fluid,

\[
\rho \ddot{W}(x,y,z) = -\frac{\partial P}{\partial z} (x,y,z) \quad \text{at} \quad z = 0, \tag{2}
\]

where \( k_0 = \omega/c \) denotes the acoustic wave number and \( c \) is the speed of sound in the surrounding medium. The quantity \( W(x,y) \) represents the complex amplitude of the out-of-plane acceleration distribution, i.e., \( \ddot{W}(x,y) = -\omega^2 W(x,y) \).

Defining the two-dimensional wave-number transform of the spatial distribution \( f(x,y) \) as

\[
\tilde{f}(k_x,k_y) = \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} f(x,y) e^{i(k_x x + k_y y)} \, dx \, dy, \tag{3}
\]

Eqs. (1) and (2) become, respectively, in the wave-number domain,

\[
\left( k_0^2 - k_x^2 - k_y^2 + \frac{\partial^2}{\partial x^2} + \frac{\partial^2}{\partial y^2} \right) \tilde{P}(k_x,k_y,z) = 0, \tag{4}
\]

and

\[
\rho \tilde{W}(k_x,k_y) = -\frac{\partial \tilde{P}}{\partial z} (k_x,k_y,z) \quad \text{at} \quad z = 0. \tag{5}
\]

Note that the sign of the forward transform in Eq. (3) is appropriately chosen to obey the Sommerfeld radiation condition. The above second-order homogeneous differential equation is solved for \( \tilde{P}(k_x,k_y,z) \)

\[
\tilde{P}(k_x,k_y,z) = -\frac{j \rho \tilde{W}(k_x,k_y)}{(k_0^2 - k_x^2 - k_y^2)^{1/2}} \times \exp[-j(k_0^2 - k_x^2 - k_y^2)^{1/2} z]. \tag{6}
\]

The term \( \exp[ -j(k_x^2 + k_y^2) z] \) that represents waves going toward the structure is omitted since the acoustic medium is boundless. The constant that remains in the solution is then found using the transformed boundary condition in Eq. (5). From Eq. (6), the transformed pressure field is seen to decrease exponentially as waves travel away from the structure for values of \( k_x \) and \( k_y \) such that \( k_x^2 + k_y^2 > k_0^2 \), i.e., in the subsonic wave-number region. The structure radiates in the far field only when \( k_x^2 + k_y^2 < k_0^2 \), i.e., in the supersonic wave-number region. Now, the pressure field is expressed in the spatial domain by applying the inverse wave-number transform:

\[
P(x,y,z) = \frac{1}{(2\pi)^2} \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} \tilde{P}(k_x,k_y,z) e^{-j(k_x x + k_y y)} \times e^{-j k_0 z} \, dk_x \, dk_y. \tag{7}
\]

Jungre and Feit9 showed that the above integral can be evaluated in the far field using the stationary phase approximation. Using spherical coordinates, \( x = r \sin \theta \cos \phi \), \( y = r \sin \theta \sin \phi \), and \( z = r \cos \theta \), the points of stationary phase are found to be

\[
k_x = k_0 \sin \theta \cos \phi \quad \text{and} \quad k_y = k_0 \sin \theta \sin \phi. \tag{8}
\]

The pressure field at a particular spherical coordinate now becomes a function of the acceleration wave-number transform evaluated at \( (k_x, k_y) \):

\[
p(r, \theta, \phi, t) = \frac{\rho e^{-j k_0 r}}{2\pi r} \times \tilde{W}(k_0 \sin \theta \cos \phi, k_0 \sin \theta \sin \phi) e^{i\omega t}. \tag{9}
\]

The main significance of the above expression can be stated as follow: For a baffled planar structure, the far-field radiation in a given direction is solely a function of the wave number transform, \( \tilde{W}(k_0 \sin \theta \cos \phi, k_0 \sin \theta \sin \phi) \), of the plane wave of wave number \( k_0 \) evaluated at a set of points on a sphere of radius \( r \).

number component evaluated at the stationary-phase wave numbers. This result motivates the design of a structural wave-number sensor.

### Estimate of the acceleration wave-number transform

A time domain estimate of the structural out-of-plane acceleration wave-number component coupled to acoustic radiation in direction $(\theta, \phi)$ can be constructed by discretizing the spatial structural response over a grid of $N_x$ by $N_y$ measurement points equally spaced along the $x$ and $y$ axis. The continuous representation of the acceleration wave-number transform is approximated by a finite summation defined as

$$w_d(k_x, k_y, t) = \Delta x \Delta y \sum_{n_x=1}^{N_x} \sum_{n_y=1}^{N_y} w(x_{n_x}, y_{n_y}, t) \times \exp(j k_x x_{n_x}) \exp(j k_y y_{n_y}),$$

where $\Delta x = L_x/N_x$ and $\Delta y = L_y/N_y$. Writing the stationary-phase wave numbers, $k_x$ and $k_y$, in terms of frequency [Eq. (3)], the discrete wave-number transform in Eq. (10) can be rewritten in terms of the frequency $\omega$ and radiation angles $(\theta, \phi)$ as

$$w_d(\omega, \theta, \phi, t) = \Delta x \Delta y \sum_{n_x=1}^{N_x} \sum_{n_y=1}^{N_y} w(x_{n_x}, y_{n_y}, t) \exp(j \omega \tau_{n_x, n_y}).$$

In the above expression, $w(x_{n_x}, y_{n_y}, t)$ represents the time domain acceleration measured at location $(x_{n_x}, y_{n_y})$ while the exponential term is a constant magnitude and linear phase $\exp(j \omega \tau_{n_x, n_y})$. It should be noted that the discrete wave-number transform allows evaluation of $w_d(k_x, k_y, t)$ at any value of $(k_x, k_y)$ and thus radiation at any particular angle. The more conventional discrete Fourier transform is usually defined only for a discrete set of values, $n_0 k_x$ and $n_0 k_y$, where $\Delta k_x$ and $\Delta k_y$ represent the spatial frequency resolution along the $x$ and $y$ axes. This definition makes possible the use of fast Fourier transform algorithms in the off-line evaluation of the wave-number transform. However, the FFT approach cannot be applied to real time wave-number sensing.

### EXPERIMENTAL SETUP

An experimental investigation of the sensing technique and ASAC implementation was performed on a simply supported plate mounted in a rigid baffle inside an anechoic chamber with a cutoff frequency of 250 Hz. The plate is made of plain carbon steel and has dimensions $380 \times 298 \times 1.93$ mm. The arrangement is shown schematically in Fig. 1. Thin, flexible metal shims connect the edges of the plate to a heavy support stand to provide the simply supported boundary conditions. Table I gives the experimentally measured resonance frequencies of the first ten bending modes of the plate.

To validate the sensing method, a first set of tests was conducted to compare the wave-number components predicted by the sensing technique with the actual wave-number components of the plate over a broadband frequency range. The disturbance signal is a bandlimited random noise from 0 to 630 Hz which is generated by a Bruel & Kjær 2032 spectrum analyzer and a Frequency Device 9002 low-pass filter with cutoff frequency 630 Hz. This signal is amplified and fed to a point force shaker actuator. The shaker location is given in Table II.

The predicted wave-number component is constructed

<table>
<thead>
<tr>
<th>Mode $(m, n)$</th>
<th>Measured frequency (Hz)</th>
</tr>
</thead>
<tbody>
<tr>
<td>(1,1)</td>
<td>87</td>
</tr>
<tr>
<td>(2,1)</td>
<td>183</td>
</tr>
<tr>
<td>(1,2)</td>
<td>244</td>
</tr>
<tr>
<td>(2,2)</td>
<td>330</td>
</tr>
<tr>
<td>(3,1)</td>
<td>343</td>
</tr>
<tr>
<td>(2,3)</td>
<td>474</td>
</tr>
<tr>
<td>(1,3)</td>
<td>491</td>
</tr>
<tr>
<td>(3,2)</td>
<td>556</td>
</tr>
<tr>
<td>(4,1)</td>
<td>571</td>
</tr>
<tr>
<td>(2,4)</td>
<td>657</td>
</tr>
</tbody>
</table>

FIG. 1. Plate geometry and control arrangement.
according to the discrete formulation in Eq. (11). Figures 1 and 2 present the configuration of the wave-number sensor. Nine accelerometers are mounted on the structure to provide the structural information, i.e., $N_x = N_y = 3$. Referring to Fig. 2, the acceleration signals are sampled and passed through digital filters whose impulse responses model the transfer functions $\exp(j\omega \tau_{\eta \eta}^\nu)$ as described in Refs. 7 and 8. The sum of the filtered signals gives the time domain estimate of the wave-number component. The wave-number sensor uses three mini B&K accelerometers and six BBN 501 accelerometers. Six in-house developed charge amplifiers and three B&K charge amplifiers type 2635 are used to condition the accelerometer output signals. The signal processing part of the sensing technique is implemented on a TMS320C30 digital signal processor (DSP) board installed into a host 80486-based personal computer. A C code has been previously written to process up to 12 input signals and generate three output signals corresponding, respectively, to 12 structural measurements points and three wave-number components. In the following tests, the three outputs are used, thus allowing to test the sensor accuracy in three different directions simultaneously. The three estimated wave-number components were chosen to be coupled to far-field radiation at angles $\theta = -36^\circ$, $\theta = 0^\circ$, and $\theta = 36^\circ$ in the $x-z$ plane ($\phi = 0^\circ$). These three angles were found to adequately sample the pressure field in the radiation midplane. The filter impulse responses associated with each point sensor are computed analytically as described in Ref. 8. The sampling frequency was set to 2000 Hz and each filter had three coefficients to model the transfer functions. In order to compensate for the slight magnitude and phase differences between the dynamic response of each accelerometer and charge amplifier path, each "sensing" discrete impulse response is convolved with a "calibration" impulse response. The calibration impulse response is implemented by a FIR filter designed to model the magnitude and phase differences relative to a reference path. To this purpose, the accelerometers are initially mounted on a piston shaker providing constant acceleration over its surface. With the shaker being exciting over the frequency range of interest, i.e., $0-630$ Hz, the outputs of the charge amplifiers are fed to the DSP board and an on-line system identification code using the LMS algorithm is implemented to update the FIR filter coefficients. Five coefficients were used for the calibration impulse response. Thus the resulting filters had seven coefficients.

The actual wave-number component of the plate is computed using the plate out-of-plane response data measured by a laser vibrometer. The laser vibrometer is mounted on a computer-controlled linear traverse that allows the measurements of the transfer function between the disturbance signal and the plate out-of-plane velocity on an angle by seven grid of points equally spaced on the plate. The plate modal amplitudes and its wave-number transform are then computed off-line as described in the Appendix. Preliminary studies on an analytical model showed that the above discretization gave almost perfect matching between continuous, i.e., the closed-form expression of the wave-number transform, and discrete representation, on the frequency range of interest. Here, the discrete representation refers to the off-line estimate of wave-number components using Eq. (10) rather than the output of the wave-number sensor.

The second set of experiments was conducted to demonstrate the use of the sensing technique in a feedforward sound radiation control approach with broadband disturbance. Three single-sided G1195 PZT piezoelectric actuators with dimension $38 \times 32 \times 0.19$ mm are mounted on the plate as control inputs. Their center coordinates are given in Table II and they are shown schematically in Fig. 1. The disturbance input is applied through the same shaker as for the sensing tests. The low-pass filter cutoff frequency is now set to 400 Hz in order to include the first five bending modes of the plate. The wave-number sensor also uses the same configuration as for the sensing tests, i.e., same sampling frequency and filter coefficients. To compare the performances of the wave-number sensor with far-field pressure sensing, three error microphones are set up in the radiation farfield. The three microphones are located in the horizontal plane ($\phi = 0^\circ$) at a radius $r = 1.85$ m along the three directions of pressure estimate given above. The filtered-X version of the LMS algorithm is implemented on a second TMS320C30 DSP board in order to drive the three control inputs. The control system adapts the coefficients of three FIR compensators based upon the error signals, i.e., outputs of wave-number sensor or far-field error microphones, and the reference signal filtered by an estimate of the transfer functions between each one of the three control input signals and three error output signals. These nine fixed transfer functions can be modeled by infinite impulse response (IIR) filters in the case of broadband disturbances as discussed in Refs. 13 and 14. The reference signal is taken from the output of the low-pass filter, thus providing a signal coherent with the disturbance. The sampling frequency of the controller board is set to 1500 Hz. A set of nine 35th-order IIR filters are used to model the filtered-X paths, while the three adaptive FIR compensators have a 35 coefficient impulse response. Note that a single board could implement both sens-

### Table II. Transducer locations.

<table>
<thead>
<tr>
<th>Transducer type</th>
<th>$x$ coordinate (mm)</th>
<th>$y$ coordinate (mm)</th>
</tr>
</thead>
<tbody>
<tr>
<td>disturbance (shaker)</td>
<td>-115</td>
<td>-84</td>
</tr>
<tr>
<td>control 1 (pzt)</td>
<td>79</td>
<td>-67.5</td>
</tr>
<tr>
<td>control 2 (pzt)</td>
<td>79</td>
<td>58</td>
</tr>
<tr>
<td>control 3 (pzt)</td>
<td>-35</td>
<td>-6</td>
</tr>
</tbody>
</table>

![Wave-number sensor block diagram](image-url)

J. P. Maillard and C. R. Fuller: Time domain sensing 2616

TABLE III. Extra microphone locations.

<table>
<thead>
<tr>
<th>Location</th>
<th>r (m)</th>
<th>( \theta ) (deg)</th>
<th>( \phi ) (deg)</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>1.19</td>
<td>48</td>
<td>8</td>
</tr>
<tr>
<td>2</td>
<td>1.32</td>
<td>10</td>
<td>0</td>
</tr>
<tr>
<td>3</td>
<td>1.96</td>
<td>53</td>
<td>-148</td>
</tr>
</tbody>
</table>

and control part, due to the relatively low computational load required by the wave-number sensor. However, for these laboratory tests, it was more convenient to program the sensing and control on separate DSP boards. To monitor the radiated sound from the uncontrolled and controlled system, a microphone traverse located in the far-field measures the pressure field in the horizontal plane at 21 angles with a 9-deg increment. Three extra fixed microphones provide pressure information out of the horizontal plane in order to confirm the global nature of the control. Their location appears in Table III. The laser vibrometer data is also used to compute the plate modal amplitudes before and after control as described in the Appendix. All the measured time domain signals are transformed into the frequency domain using a B&K analyzer type 2032.

III. RESULTS

A. Wave-number sensing

Analytical results and simulations from Refs. 7 and 8 show that only a small number of point sensors are required to estimate wave-number components for the low-order modes of a simply supported beam. The following section discusses the experimental results obtained with the simply supported plate and sensor presented earlier.

Figure 3 shows the magnitude of the actual (solid line) and predicted (dashed line) wave-number components associated with direction of radiation \( \theta = 36^\circ \). Both transfer functions are relative to the disturbance signal (output of the signal generator) and the peaks observed in the response do not always match the resonance frequencies given in Table I. As in standard modal analysis practice, these resonance frequencies are based on transfer functions relative to the output signal of a force transducer monitoring the input force applied by the disturbance shaker. The plot shows excellent matching between actual and predicted wave-number components up to 200 Hz. Slight magnitude increases occur above this frequency but the variations remain within 3 dB over the first five modes of the plate. Above 400 Hz, the error between actual and predicted wave-number component further increases reaching a maximum of 8 dB at the resonance frequency of mode (4, 2). The phase angle exhibits the same trends and is not shown here. Also note that similar performances were obtained in the two other directions of radiation, and control on separate DSP boards. To monitor the radiated sound from the uncontrolled and controlled system, a microphone traverse located in the far-field measures the pressure field in the horizontal plane at 21 angles with a 9-deg increment. Three extra fixed microphones provide pressure information out of the horizontal plane in order to confirm the global nature of the control. Their location appears in Table III. The laser vibrometer data is also used to compute the plate modal amplitudes before and after control as described in the Appendix. All the measured time domain signals are transformed into the frequency domain using a B&K analyzer type 2032.

The variations between actual and predicted wave-number components can be further discussed in terms of aliasing due to the discrete integration scheme of the wave-number transform. A one-dimensional structure is considered here for simplicity. As discussed by Faby,\(^1\) the \( p \)th bending mode presents a main peak in the wave-number domain at \( k_x = \pm p \pi / L \). Hence significant aliasing will occur at the corresponding resonance frequency if the Nyquist wave number, \( K_s = \pi / \Delta x = L / N_s \) (being the spatial sampling period), is smaller than \( p \pi / L \) or \( N_s > p \). Moreover, since the structure is finite, the main peak obtained at a resonance frequency will be scattered into a continuum of smaller wave-number components extending up to infinity. Therefore, some aliasing will also occur for modes where \( N_s > p \). In order to reduce aliasing errors, it is necessary to filter out the wave-number components that are higher than the Nyquist wave number. No sensor design has been proposed at this time to achieve spatial wave-number filtering. Such a sensor should convolve the spatial distribution of the structural response with the appropriate impulse response, whose wave-number transform has the characteristics of a low-pass filter. It should be noted that distributed sensors, such as PVDF films, perform spatial weighting of the structural response, i.e., a multiplication rather than a convolution product. Discussion of aliasing and its effect on the discrete wave-number transform and aliasing are also given in Ref. 7.

To further validate the wave-number sensing technique, it is of interest to study how the wave-number information relates to the actual radiated pressure. In theory, wave-number components coupled to a given direction of radiation are related to the far-field pressure by a constant magnitude and linear phase factor as expressed in Eq. (9). Figure 4 shows the magnitude of the transfer function between the signal generator and the measured actual wave-number component coupled to direction \( \theta = 36^\circ \) along with the magnitude of the measured transfer function between the signal generator and the traverse microphone output located along the same direction. Note that a scaling factor has been introduced to facilitate the comparison. Very good agreement between wave-number information and measured radiated pressure is obtained at practically all frequencies. The important variation noticed at the resonance frequency of mode (1, 2) of
the plate, 244 Hz, is mainly due to 60-Hz noise in the microphone charge amplifier used for the measurements. Very poor coherence (below 0.4) was noticed at this frequency. It should also be noted that the even modes along the y direction, i.e., modes (1,2), (2,2), and (3,2), do not exhibit the expected zero sound-pressure level in the horizontal x-z plane. Due to imperfections in the boundary conditions as well as the discontinuities introduced by the piezoelectric patches and the disturbance shaker, the mode shapes of the plate are not perfectly symmetric with respect to the x and y axes. Hence, the velocity distribution along the y direction does not perfectly cancel to give zero far-field pressure in the horizontal plane.

B. Radiation control

The second set of experiments described in this paper implements wave-number sensing in the radiation control of the plate presented earlier. As described in part II, the control approach uses a three channel filtered-X LMS algorithm and the wave-number sensor provides an estimate of the acceleration wave-number components coupled to radiation in directions θ = 36°, θ = 0°, and θ = 36°. Three microphones located in the far field at the same angles in the horizontal plane are also used as error signals in order to compare the relative performances of both types of sensing.

Figure 5 shows the experimentally measured auto-spectrum of the third error signal (wave-number component estimate coupled to radiation at angle θ = 36°) for the uncontrolled (solid line) and controlled (dashed line) system using wave-number sensing as described above. Significant reduction is achieved across the entire bandwidth. The radiation for each one of the five modes is attenuated and the residual response does not show any clear resonance behavior, resulting in a rather flat frequency content. The same trends were measured for the first (θ = 36°) and second (θ = 0°) error signals. The total sound-pressure level attenuation of the three error signals was calculated by integrating the three auto-spectra over 10–600 Hz. Reduction levels of 12, 12.8, and 13.5 dB were achieved, respectively. When replacing the wave-number sensor outputs by the three error microphones, similar performances are obtained in terms of error signals. Figure 6 shows the auto-spectrum of the third microphone error signal (θ = 36°) before and after control. Again, attenuation is achieved over the entire bandwidth. In this case, the

![Graph](image)

FIG. 5. Auto-spectrum of the third wave-number sensor error signal (θ = 36°, φ = 0°) before (---) and after (-----) control.

![Graph](image)

FIG. 6. Auto-spectrum of the third microphone error signal (θ = 36°, φ = 0°) before (---) and after (-----) control.

FIG. 7. Total sound-pressure level reduction integrated over 10–600 Hz using wave-number sensor (----) and error microphones (-----).
TABLE IV. Total sound-pressure level reduction integrated over 10-40 Hz.

<table>
<thead>
<tr>
<th>SPL reduction in dB</th>
<th>Location 1</th>
<th>Location 2</th>
<th>Location 3</th>
</tr>
</thead>
<tbody>
<tr>
<td>wave-number sensing</td>
<td>11.8</td>
<td>9.6</td>
<td>10.3</td>
</tr>
<tr>
<td>error microphones</td>
<td>12.4</td>
<td>10.6</td>
<td>10.6</td>
</tr>
</tbody>
</table>

The total reduction levels for the three microphones are 9.3, 10.4, and 12.6 dB, respectively.

In order to estimate the overall performances of both control systems using wave-number sensing or pressure measurements, far-field pressure auto-spectra are measured in the horizontal plane for -90° ≤ θ ≤ 90° and at three "random" locations out of the horizontal plane. Figure 7 shows the reduction in decibels integrated over the 10- to 600-Hz bandwidth and measured in the horizontal plane. The solid line corresponds to the use of the wave-number sensor and the dashed line represents the performances obtained with error microphones. The three dashed-dotted straight lines show the three direction of minimization. The reduction obtained at the three extra locations shown in Table IV and confirm the global nature of the control. The two sensing methods yield almost the same performances. Between 10- and 12-dB reduction is obtained at all angles and similar results can be found at the three extra locations which were out of the traverse plane. Thus the wave-number sensor appears to be fully able to replace the three error microphones in the far field.

To further compare the performances of wave-number sensing versus the use of error microphones, the system structural and acoustic response is now studied at three distinct frequencies. The plate modal amplitudes are computed at 244, 305, and 110 Hz from the measured broadband data (laser vibrometer), as outlined in the Appendix. Results are shown in Figs. 8, 10, and 12, respectively. For each of the above frequencies, the corresponding far-field directivity pattern obtained from the traverse microphone measurements is also presented in Figs. 9, 11, and 13, respectively.

At 244 Hz, the plate is on resonance of the mode (1,2). On-resonance cases yield better accuracy for the wave-number sensor than off-resonance cases as shown in Fig. 3. As a result, the plate modal response after control for 244 Hz (Fig. 8) shows almost the same characteristics for both wave-number and microphone sensing methods. As expected, the far-field pressure in the horizontal plane also exhibits the same behavior whether wave-number or pressure sensing is used.

Off-resonance, the sensor accuracy deteriorates and from the increased modal complexity, the control system becomes more sensitive to small variations in the error signal. As shown in Fig. 10, the modal amplitudes after control using the wave-number sensor differ slightly from the response using pressure sensing. However the variations in the error signals remain small and both set of modal amplitudes follow the same tendency. Moreover the variations do not significantly affect the far-field pressure as seen in Fig. 11. For the second off-resonance case presented here (110 Hz), both methods reduced mode (1,1) to about the same level. However the amplitude of mode (2,1) increases much more with the error microphones. In addition to the reasons given above, the far-field error measurements also differ from the estimated wave-number components due to the presence of reflected sound waves. At this frequency, the chamber is no
longer perfectly anechoic. As a result, the measured sound pressure is perturbed by reflections from the chamber walls while the output of the wave-number sensor remains unaffected by the surrounding sound field. As expected, the error microphones do a better job in reducing the sound pressure in the far field. The microphones measure the free field as well as the reflected field. On the other hand, the wave-number sensor only observed the free field, thus resulting in poorer performances. The above results present a good example of modal restructuring. The controlled modal amplitude of mode (2, 1) increases but the total structural response still has a lower radiation efficiency resulting in far field pressure reduction.

IV. CONCLUSIONS

Broadband structural wave-number sensing has been experimentally demonstrated in the case of a baffled simply supported vibrating plate. The main significance of the approach is its ability to estimate, in the time domain, supersonic wave-number components coupled to prescribed directions of radiation over a broadband frequency range. Provided the sensor is accurate enough, it can completely replace the use of error microphones in the case of baffled planar structures. The available bandwidth is directly related to the sensor complexity, i.e., number of measurement points and order of the FIR filters. When observing the low-order structural modes, comparison of predicted and actual wave-number information shows that only a few point sensors are required to build a fairly accurate integration scheme of the wave-number transform. In order to improve the accuracy of the sensor, the use of spatially convolving sensors should be considered to provide antialiasing filters in the wave-number domain.

When applied to radiation control, wave-number sensing yields the same levels of attenuation as error microphones in the far field. By observing the radiating part of the structural vibrations (supersonic wave-number components), wave-number sensing provides a more selective error information compared to other structural sensing methods. As a result, the control authority needed to reach a given level of sound reduction is decreased and thus, the wave-number approach results in improved performances and efficiency. Moreover, it can be of special interest when directional rather than global control is needed. Sensing radiation in prescribed directions rather than over the entire surrounding medium also reduces control authority.

The present approach is only valid in the case of baffled planar radiators. For more complex geometries and non-baffled structures, a different type of integration scheme must be derived to account for the diffraction terms. This is the topic of future investigations.

ACKNOWLEDGMENTS

The author gratefully acknowledges the support of this work by the Office of Naval Research under Grant No. ONR-N00014-92-J-1170, Dr. Kam Ng, Technical Monitor.

APPENDIX

The evaluation of the plate modal amplitudes and wave-number transform from the out-of-plane velocity measurements is briefly outlined. The plate out-of-plane velocity $w(x,y,t)$ is expressed using the assumed modes method as
\[ w(x,y,t) = \sum_{m=1}^{n} \sum_{n=1}^{N} \tilde{W}_{mn}(x) \phi_m(x) \phi_n(y) e^{i\omega t} \]

where \( \tilde{W}_{mn} \) is the complex modal amplitude of mode \((m,n)\).

Assuming the plate is effectively subject to simply supported boundary conditions, the mode-shapes along the \( x \) and \( y \) directions are expressed in the coordinate system of Eq. (1) as

\[ \phi_m(x) = \sin(\gamma_m(x+L_x/2)), \]

\[ \phi_n(y) = \sin(\gamma_n(y+L_y/2)), \]

respectively, where \( \gamma_m = m\pi L_x \) and \( \gamma_n = n\pi L_y \). Equation (A1) can be written for \( N \) different locations \((x_i,y_i)\), \( i=1,2,...,N \). Truncating the modal summation to include the first \( N \) modal amplitudes, the resulting set of equations is expressed as a linear system of the form

\[ S(\tilde{W}_{mn}) = \langle \tilde{w} \rangle. \]

The column vector \( \tilde{W}_{mn} \) contains the first \( N \) velocity modal amplitudes \( W_{mn} \), \( j=1,2,...,N \), included in the summation of Eq. (A1). The column vector \( \tilde{w} \) contains the \( N \) measured transfer functions between a reference signal, e.g., signal generator, and the plate out-of-plane velocity at locations \((x_i,y_i)\). The square matrix \( S \) is given by

\[ S_{ij} = \phi_{m_j}(x_i) \phi_{n_j}(y_j), \quad i=1,2,...,N, \quad j=1,2,...,N. \]

Since there is only a finite number of measurement points, spatial aliasing occurs in the modal decomposition. Using \( N \) measurement points, only \( N \) modal amplitudes can be resolved. Therefore, the response of the higher-order modes must be assumed negligible. The 8 by 7 generator and the plate out-of-plane velocity at locations \((x_i,y_i)\) on the measurement points, only spatial aliasing occurs in the modal decomposition.

After solving the linear system in Eq. (A3) for the modal amplitudes \( W_{mn} \), the wave-number transform of the out-of-plane acceleration is obtained as

\[ \tilde{W}(k_x,k_y) = j\omega \sum_{j=1}^{N} W_{mn} \xi_{m_j,n_j}(k_x,k_y). \]

where \( \xi_{m_j,n_j}(k_x,k_y) \) represents the wave-number transform of the eigenfunction of mode \((m_j,n_j)\). A closed-form expression is given as

\[ \xi_{m_j,n_j}(k_x,k_y) = \tilde{\phi}_{m_j}(k_x) \tilde{\phi}_{n_j}(k_y). \]
ACTIVE CONTROL OF SOUND RADIATION FROM CYLINDERS WITH PIEZOELECTRIC ACTUATORS AND STRUCTURAL ACOUSTIC SENSING.

J. P. Maillard, C. R. Fuller

Vibration and Acoustics Laboratories
Mechanical Engineering Department
Virginia Polytechnic Institute and State University
Blacksburg, Virginia 24061-0238 (USA)

ABSTRACT

In this paper, analytical and experimental results of an investigation of active control of sound radiated from cylinders are presented. The aluminum cylinder is 1 m in length, 25 cm in diameter and 2.4 mm in thickness with two rigid end-caps at both ends. The excitation is a band-limited random noise encompassing the first five modes of the cylinder and the control actuators are surface mounted piezoelectric transducers. Since it is desired to integrate the error sensors into the structure, the recently developed Discrete Structural Acoustic Sensing (DSAS) approach is extended to cylindrical coordinates and implemented using twelve accelerometers mounted on the cylinder. The structural acoustic sensor provides time domain estimates of far-field radiated sound at predetermined radiation angles. The controller is a 3 by 3 Filtered-X LMS paradigm implemented on a TMS320C30 DSP. The results show good global control of the radiated sound over the frequency bandwidth of excitation. Most important, the proposed discrete structural acoustic sensor yields similar performances as error microphones located in the far field. [Work supported by the Office of Naval Research]

Keywords: active radiation control, cylinder, experiment, piezoelectric transducer, transducer array

INTRODUCTION

Much research has been conducted in the active control of low-frequency structure-borne sound. When compared to passive methods, active control presents significant advantages in the low-frequency range where passive control becomes often impractical due to prohibitive volume and/or mass requirements. For the past decade, Active Structural Acoustic Control (ASAC) has received much attention as it presents a practical alternative to the control of low-frequency radiated noise [1, 2]. In this technique, the radiated sound pressure is attenuated by applying mechanical inputs directly to the structure rather than by exciting the surrounding medium with acoustic sources (Active Noise Control). Piezoelectric devices have been applied extensively to Active Structural Acoustic Control systems as structural actuators [3–5] thus yielding a compact or “smart” structure. In an attempt to further reduce the size of the overall control arrangement, the microphones located in the far field to provide radiation error information are also being replaced by structural sensors such that all transducers are integrated in the structure.

As most ASAC applications involve noise control below the coincidence frequency of the radiating structure, appropriate structural sensors for ASAC should only observe the radiating part of the structural vibrations. This gives more flexibility to the controller which in some situations modifies the structural vibrations such as to attenuate far-field radiation with no net reduction in the overall vibration levels. Sound attenuation in the far field can then be achieved with a reduced control authority compared to cases where
all structural motion is canceled (Active Vibration Control) [6].

With the emergence of polyvinylidene fluoride (PVDF) as a sensor material, several structural sensors for ASAC have been proposed to observe the radiating part of the structural vibrations. Most of these sensing techniques are based on modal sensing [7]: the sensor effectively observes a specific set of modes of vibration (natural or radiation modes of the uncontrolled response) which couples well to far-field radiation [8-11]. An accurate knowledge of the structure’s dynamic properties is therefore required. Recently, an alternative sensing technique referred to as Discrete Structural Acoustic Sensing (DSAS) was demonstrated both analytically [12,13] and experimentally [14,15] on baffled planar radiators. The technique implements an array of structural point sensors whose outputs are passed through digital FIR filters to estimate in real time the far-field radiated pressure in a given direction, or equivalently, a given wave-number component, over a broad frequency range. It uses the relation between the structural out-of-plane vibrations and the far-field sound pressure as defined by the Helmholtz integral. One of the significant advantages of this strategy lies in its low modeling requirements compared to modal sensing approaches. In particular, the sensor design does not require the knowledge of the structural mode shapes and thus remains largely independent of the boundary conditions. Consequently, it is particularly well adapted for feedforward control approaches commonly used in ASAC systems where no analytical system modeling is necessary. This paper presents analytical and experimental results on the extension of Discrete Structural Acoustic Sensing to baffled cylindrical radiators.

Most of the work on ASAC systems deals with planar geometries or systems than can be decomposed in a set of planar radiators and few reports of experiments on cylindrical structures can be found in the literature. Previous work by Clark and Fuller [16] demonstrated experimentally the harmonic control of sound radiation from a finite enclosed cylinder using PVDF error sensors and piezo-electric actuators. The present study extends the above work by considering broadband radiation control over the first five flexural modes of the structure. After briefly introducing the theoretical formulation, analytical and experimental results are presented. In both cases, the discrete structural acoustic sensor is first studied in terms of its accuracy to predict radiated pressure. Broadband radiation control results are then discussed by comparing the performances of the sensor to those of error microphones located in the far field.

THEORETICAL BASIS

This section presents the analytical formulation of Discrete Structural Acoustic Sensing. A relation between discrete structural acceleration and far-field pressure estimate is derived for the case of baffled cylindrical geometries.

Far-field sound pressure For the general case of arbitrary geometries, the sound pressure radiated from a vibrating structure into an unbounded medium can be expressed using the Kirchhoff-Helmholtz integral formulation [17]. Assuming a harmonic solution for the pressure, \( p(r) e^{i\omega t} \), where \( \omega \) is the angular frequency, this is expressed as

\[
p(r) = \int_{S_0} \left[ \rho G(r|r_0) \hat{w}(r_0) + p(r_0) \frac{\partial G}{\partial r_0}(r|r_0) \right] dS(r_0), \quad r \in V
\]

In the above equation, \( S_0 \) denotes the radiating surface and \( V \) the surrounding volume. The sound pressure \( p(r) \) at field point \( r \) is expressed as a surface integral involving the out-of-plane structural acceleration, \( \hat{w}(r_0) \), the surface pressure \( p(r_0) \), the Green's function, \( G(r|r_0) \), and its normal gradient, and the fluid density, \( \rho \). Note that this formulation assumes the radiator has solid boundaries such that the fluid velocity on the boundary is equal to the structural out-of-plane velocity. The normal pressure gradient then becomes equal to \( \rho \hat{w}(r_0) \). Discrete Structural Acoustic Sensing is based on the existence of a Green's function satisfying the Neumann boundary condition,

\[
\frac{\partial G}{\partial n_0}(r|r_0) = 0, \quad r \in S_0
\]

such that the radiated pressure field becomes solely dependent on the structural acceleration and geometry.

For baffled cylindrical geometries (see Figure 1), a closed-form solution exists for the Green's function
Active control of sound radiation from cylinders

verifying equation (2). The radiated pressure is then expressed as

\[ p(R, \theta, \phi) = \rho \int_0^{2\pi} \int_{-L}^{+L} G(R, \theta, \phi | r = a, \phi', z') \bar{\omega}(\phi', z') a d\phi' dz' \] (3)

The Green's function in equation (3) can be obtained from the radiated pressure due to a point acceleration distribution located on an infinite cylindrical baffle. The resulting expression is approximated in the far field as [17]

\[ G(R, \theta, \phi | a, \phi_0, z_0) = \frac{\exp \left[ -jk(R - z_0 \cos \theta) \right]}{\pi^2 ak R \sin \theta} \sum_{n=0}^{\infty} \frac{j^{n+1} \cos [n(\phi - \phi_0)]}{\epsilon_n H^{(2)'}_n (ka \sin \theta)} \] (4)

where \( \epsilon_n = 2 \) for \( n = 0 \) and \( \epsilon_n = 1 \) for \( n > 0 \) (n integer). The function \( H^{(2)'}_n (z) \) denotes the first derivative of the \( n^{th} \) Hankel function of the second kind. The acoustic wave-number is denoted as \( k = \omega/c \) where \( c \) is the speed of sound. The various coordinates and dimensions involved in equation (4) are shown in Figure 1.

Sensor estimate An estimate of the radiated pressure in equation (3) is now constructed. The integral over the radiating surface \( S_0 \) is approximated using a \( N \) point zero-order interpolation of the acceleration distribution [15], i.e., the acceleration is assumed constant over \( N \) small elemental surfaces, \( S_m, m = 1, 2, \ldots, N \), such that \( S = \bigcup_{m=1}^{N} S_m \). The resulting pressure estimate takes the general form

\[ p_d(R, \theta, \phi, t) = \sum_{m=1}^{N} \bar{\omega}(\phi_m, z_m, t) H_m(R, \theta, \phi) \] (5)

where \((\phi_m, z_m)\) represents the coordinates of the \( m^{th} \) node, and \( H_m(R, \theta, \phi) \), the \( m^{th} \) sensor transfer function. Defining \( S_m = a \Delta z_m \Delta \phi_m \) as the \( m^{th} \) elemental surface aligned with the axial and circumferential directions such that its center coincides with \((\phi_m, z_m)\), the sensor transfer functions can be expressed as

\[ H_m(R, \theta, \phi) = \rho \int_{\phi_m - \Delta \phi_m/2}^{\phi_m + \Delta \phi_m/2} \int_{z_m - \Delta z_m/2}^{z_m + \Delta z_m/2} G(R, \theta, \phi | a, \phi_m, z_m) a d\phi dz \] (6)

The transfer function \( H_m(R, \theta, \phi) \) can be interpreted physically as the sound pressure radiated at \((R, \theta, \phi)\)
from the \(m^{th}\) elemental surface vibrating along the normal to its center \((\phi_m, z_m)\) with a unit acceleration. In other words, the pressure estimate is constructed by summing the radiation contribution of \(N\) cylindrical pistons weighted by the measured acceleration amplitudes. It is thus referred to as the piston approximation. Assuming the Green's function remains almost constant over each surface \(S_m\), the transfer function in equation (6) can be replaced by \(p(a, \Delta \phi_m, \Delta G(R, \theta, \phi, \phi_m, z_m))\). In this case, the far-field pressure is estimated from the contribution of \(N\) monopole sources (monopole approximation). As expected, both approximations become equivalent as \(k_{\text{max}}(\Delta z_m, \Delta \phi_m) \ll 1\).

It should be stressed that the sensor transfer functions solely depend on the geometry of the problem and the properties of the fluid medium. No accurate knowledge of the structure's dynamics is thus required for their design. Note however that some information is still needed in order to determine an appropriate discretization level for accurate estimates. Furthermore, the sensing approach can be extended to geometries for which no Green's function is available analytically. The far-field pressure radiated from each elemental surface vibrating independently on the structure's boundary must then be solved numerically using the Boundary Element Method.

PRACTICAL IMPLEMENTATION

This section briefly recalls some of the important issues associated with practical implementation. The pressure estimate presented in the previous section is implemented on a real system using a set of accelerometers mounted on the structure and arrays of digital filters. More precisely, each measured acceleration signal is passed through a digital filter modeling the associated sensor transfer function. All filter outputs are then summed to provide the sound pressure estimate. Several arrays of filters can be implemented in order to provide pressure estimates at different locations.

As explained above, each transfer function represents the far-field radiation into an unbounded medium from a cylindrical piston source (or monopole) with unit acceleration and located on a cylindrical baffle. The associated characteristics directly motivate the use of Finite Impulse Response (FIR) filters to model the sensor transfer function. In particular, no resonance behavior occurs due to the assumption of an unbounded medium and notches in the transfer functions magnitude associated with zero pressure angles are easily modeled by appropriate zeros in the filter's impulse response.

Another important issue is related to the time delay of the sensor transfer functions, which is directly related to the acoustic path propagation time, \(R/c\). As the pressure estimate is only valid in the far field, this delay can become significant compared to the sampling period of the digital filter, thus increasing its complexity. It can be shown however that error signals based on far-field pressure at a given location can be shifted in time without loss of performance of the control system (the time shift is equivalent to moving the minimization point along a constant radiation angle) [15]. Removing the above time delay yield transfer functions with a minimum phase delay which significantly reduces the number of FIR filter coefficients required for accurate modeling.

SYSTEM CHARACTERISTICS AND EXPERIMENTAL SETUP

Testing of the structural acoustic sensor described above was performed on a finite aluminum cylinder. This section presents the main characteristics of the system, the control and measurement setups implemented in the experiments as well as the numerical model of the structure.

System characteristics Due to the limited number of accelerometers available for implementing Discrete Structural Acoustic Sensing, the choice of the cylinder's dimensions and material was made such that the first few flexural modes of the structure have low modal order in both the axial and circumferential directions. The cylinder characteristics are given in Table 1. The dimensions were measured on the actual structure while the material properties are based on standard values for aluminum. In order to allow structural vibration measurements over the entire surface of the cylinder as well as acoustic measurements over the sphere surrounding the structure, the cylinder's attachment to its support stand is designed to allow full rotation along its main axis. The cylinder is closed at both ends by aluminum end-caps 12.7 mm in thickness. Each end-cap is attached to the cylinder with a set of 12 small screws equally spaced along the circumference.
A steel rod 3.18 mm in diameter is threaded into each end-cap and mounted on a nylon ring. The ring is fixed in a 19.05 mm aluminum section which is bolted to a heavy steel support stand. To allow acoustic measurements of the baffled structure, two sections of "rigid" PVC pipe are installed on each side of the cylinder along its main axis. The two pipes extend in length up to the walls of the anechoic chamber. A picture of the complete rig including the baffle is shown in Figure 2.

All structural vibration inputs, i.e., both disturbance and control inputs, are applied through single-sided piezo-electric actuators. No curved actuators were available for the experimental testing and flat actuators [18] were mounted on the cylinder by cutting them across their width into a set of eight strips of same dimensions. The original actuators are 63.5 mm in length and 38.1 mm in width, which results in eight 7.9 mm by 38.1 mm strips. They are mounted on the cylinder's outside surface side by side with their length along the cylinder's axis and wired in phase. A gap of approximately 1 mm remains between each actuator strip to avoid short circuits between the electrodes of two adjacent actuators. The total surface area covered by the set of eight actuators is 69.9 mm by 38.1 mm. Table 1 presents the dimensions and material properties of the piezo-electric actuators. Four sets of the actuator arrangement described above are mounted on the cylinder according to the center locations given in Table 2. The disturbance actuator center location serves as the origin of the circumferential direction, \( \phi = 0^\circ \). Its axial location, \( z/L = -0.328 \), ensures that all flexural modes present in the 0-1000 Hz bandwidth are excited. The three other actuator sets are implemented as control inputs. Their location was determined such as to allow various control configurations.

The discrete structural acoustic sensor implemented on the cylinder uses 12 accelerometers (PCB Piezoelectric ICP accelerometers - Model 352A10). The point sensors are arranged as 2 rings of 6 accelerometers equally spaced according to

\[
\begin{align*}
\phi_m &= m\Delta\phi, \\
z_n &= -L + \Delta z/2 + n\Delta z, \\
&= n = 0, 1, \ldots, N_z - 1, \\
&\Delta\phi = 2\pi/N_\phi \\
&\Delta z = 2L/N_z
\end{align*}
\]  

(7)

where \( N_z = 2 \) and \( N_\phi = 6 \) are the number of points along the axial and circumferential direction, respectively. Note that the accelerometers are aligned with the disturbance actuator such that the modes excited by the disturbance input have antinodes along the circumferential directions aligned with the point sensor locations. As it will be discussed later, the sensor accuracy is independent of the point sensor circumferential locations when the number of sensors along the circumferential direction is greater than the highest circumferential order of the modes present in the bandwidth. When this condition is not satisfied however, care must be taken so that the point sensors do not coincide with the nodal lines of a given mode (mode \( n = 3 \) in this case). Furthermore, placing the sensors on anti-nodal lines ensures better signal to noise ratio thus improving the accuracy of the sensor estimate.

Control and measurement setups For all experimental testing, the cylinder is excited through the disturbance actuator with a band-limited random noise. The sensor accuracy tests are performed over a 200-630 Hz bandwidth while the control tests use a 200-500 Hz bandwidth. The reduced bandwidth associated with the control tests ensures that the 2 by 6 sensor yields relatively accurate estimates over the frequency range. A three channel Filtered-X LMS algorithm [19] is implemented on a Texas Instrument TMS320C30 digital signal processor (DSP) to provide up to three control signals. The results presented in this paper correspond to cases where only the first and second control actuators were excited however (see

<table>
<thead>
<tr>
<th>Actuator</th>
<th>( z/L )</th>
<th>( \phi ) (degrees)</th>
</tr>
</thead>
<tbody>
<tr>
<td>disturbance PZT</td>
<td>-0.328</td>
<td>0</td>
</tr>
<tr>
<td>control PZT # 1</td>
<td>0.370</td>
<td>180</td>
</tr>
<tr>
<td>control PZT # 2</td>
<td>0.220</td>
<td>60</td>
</tr>
<tr>
<td>control PZT # 3</td>
<td>-0.395</td>
<td>250</td>
</tr>
</tbody>
</table>

Table 2: Actuators center location.
The controller's reference is taken from the signal fed to the disturbance actuator. Most of the control tests include an artificial delay in the disturbance path so as to improve system causality. A few cases were also run with a zero delay in order to evaluate the influence of system causality on the control performance. All control tests use the following settings: the sampling frequency is set to $F_s = 2000$ Hz, the FIR compensators have 50 coefficients, and the IIR filters modeling the filtered-$x$ path transfer functions have 60 coefficients in both numerator and denominator. All tests use three error signals based on the structural acoustic sensor and far-field error microphones (B&K 1/2 in microphone, Model 4166), respectively. The sensor's array of filters is implemented on a second TMS320C30 digital DSP. Note that both controller and sensing code could be implemented on a single DSP. The three error microphones are located along $\theta = 70^\circ$, $\theta = 90^\circ$, and $\theta = 110^\circ$, in the $\phi = 0^\circ$ plane at $R = 1.85$ m while several sets of FIR filters were designed to provide pressure estimates for various radiation angles. All FIR filters have 22 coefficients with a sampling frequency, $F_s = 6000$ Hz. The sensor transfer functions are accurately modeled up to about 2500 Hz. This wide bandwidth relative to the actual bandwidth of excitation was found necessary in order to ensure stability for the control system. Cases where the response of the sensor filters is not constrained at higher frequencies can lead to unwanted amplification of high frequency content remaining in the system due to the finite roll-offs of the low-pass filters.

All tests were conducted in a 4.2 m by 2.2 m by 4.5 m anechoic chamber at the Vibration and Acoustics Laboratories (VAL), Virginia Tech. The chamber has an approximate cut-off frequency of 250 Hz. Out-of-plane structural vibrations are measured with a Politec laser vibrometer (Model OFV-2600/OFV-501). To allow measurements over the entire radiating surface, the laser head is mounted on a one-dimensional linear traverse driven by a stepper motor while a second stepper motor mounted on the end-cap assembly rotates the cylinder around its main axis. The structural velocity measurements use a grid of 13 points along the axial direction and 18 point along the circumferential direction. The measurement point locations are defined by equation (7) with $N_x = 13$ and $N_\phi = 18$. This discretization level proved to be sufficient to accurately measure the structural response over the frequency bandwidth of excitation. The sound pressure radiated from the cylinder is measured inside the anechoic chamber with a B&K 1/2 in microphone (Model 4166) mounted on a circular traverse. The traverse microphone is located at a radius of 1.85 m from the center of the cylinder. Due to the legs of the support stand, the traverse can only move in the $x$-$z$ plane from $\theta = 10^\circ$ to $\theta = 170^\circ$. The rotation of the cylinder also allows measurements along the circumferential direction from $\phi = 0^\circ$ to $\phi = 360^\circ$. All far-field measurements use a grid of $N_\theta = 13$ points along the azimuthal direction, $\theta$, and $N_\phi = 18$ points along the circumferential direction, $\phi$.

Numerical simulations In order to study various sensor configurations, numerical simulations were performed prior to the experiments described above. The cylinder structure is modeled under steady-state harmonic excitation of point forces and piezo-electric actuators with a variational approach implementing the Rayleigh-Ritz formulation [15]. In this model, the mechanical displacements and electrical fields within the piezo-electric actuators are fully coupled thus including the mass and stiffness loading of the actuators. This energy based formulation also allows modeling of arbitrary boundary conditions applied along the edges of the cylinder. The reader is referred to [15] for a complete description of the model. The optimal control voltage to each actuator is computed using standard Linear Quadratic Optimal Control theory, where the cost function to be minimized is a quadratic function of the control voltage amplitudes.
Table 3 presents the natural frequencies of the first few modes of the cylinder as obtained from the numerical model and experimental modal analysis, respectively. Examination of the associated mode shapes reveals the structure's attachment creates approximately simply-supported boundary conditions. The first modal index, m, is associated with the axial direction, and the second index, n, with the circumferential direction. Note that each mode of vibration is associated with two distinct natural frequencies and mode shapes rotated along the circumferential direction by \( \pi/(2n) \) relative to one another. This behavior is expected due to the asymmetry introduced by the added mass and stiffness of the piezo-electric actuators. Disregarding the mismatch of mode (1,1), good agreement between numerical and experimental natural frequencies can be observed. Including the circumferential dependence of the stiffness factors used to model the boundary conditions would possibly improve the match especially for the (1,1) "beam" mode.

<table>
<thead>
<tr>
<th>mode</th>
<th>modeled (Hz)</th>
<th>measured (Hz)</th>
</tr>
</thead>
<tbody>
<tr>
<td>((1,2))</td>
<td>240.8</td>
<td>241.2</td>
</tr>
<tr>
<td>((1,2)^*)</td>
<td>241.0</td>
<td>244.2</td>
</tr>
<tr>
<td>((1,3))</td>
<td>304.9</td>
<td>302.9</td>
</tr>
<tr>
<td>((1,3)^*)</td>
<td>305.5</td>
<td>303.1</td>
</tr>
<tr>
<td>((2,2))</td>
<td>498.8</td>
<td>497.0</td>
</tr>
<tr>
<td>((2,3)^*)</td>
<td>499.2</td>
<td>500.3</td>
</tr>
<tr>
<td>((1,4))</td>
<td>547.6</td>
<td>540.6</td>
</tr>
<tr>
<td>((1,4)^*)</td>
<td>547.8</td>
<td>541.1</td>
</tr>
<tr>
<td>((1,1))</td>
<td>565.3</td>
<td>708.0</td>
</tr>
<tr>
<td>((1,1)^*)</td>
<td>565.5</td>
<td>-</td>
</tr>
<tr>
<td>((2,4))</td>
<td>609.1</td>
<td>601.7</td>
</tr>
<tr>
<td>((2,4)^*)</td>
<td>609.2</td>
<td>604.6</td>
</tr>
</tbody>
</table>

Table 3: Comparison of the numerical and experimental natural frequencies.

**SENSOR ACCURACY**

This section presents successively analytical and experimental results showing the accuracy of the structural acoustic sensor. The sensor estimate is compared to the actual sound pressure radiated in the far field over the 200–680 Hz bandwidth.

**Analytical results** Figure 3 shows the magnitude of the far-field radiated pressure in direction \( (\theta = 70^\circ, \phi = 240^\circ) \) (solid line) along with the sensor estimate based on the piston approximation and two different point sensor configurations. The dashed line corresponds to a 8 by 7 sensor, i.e., the acceleration measurement points are located according to equation (7) with \( N_x = 8 \) and \( N_\phi = 7 \), and the dotted line to a 2 by 12 sensor. Recalling the natural frequencies given in Table 3, the resonance peaks correspond, as frequency increases, to modes \((1,2), (1,3), (2,3), (1,4), (1,1)\) and \((2,4)\) where the first and second index refers to the axial and circumferential order, respectively. As seen on the plot, the 8 by 7 sensor yields excellent accuracy at the resonance frequency of all modes included in the bandwidth except for modes \((1,4)\) and \((2,4)\) where large errors can be observed. Considering the 2 by 12 sensor, excellent accuracy is obtained at the resonance frequency of all modes except for a small error near resonance of modes \((2,3)\) and \((2,4)\). These results illustrate two fundamental properties of the sensor estimate for cylindrical geometries.

Analogous to the case of planar radiators, the far-field radiated pressure in equation (3) can be expressed in terms of the two-dimensional wave-number transform of the structural out-of-plane acceleration [15]. The wave-number transform along the circumferential direction maps a periodic spatial distribution with period \( 2\pi \) into two sets of wave-number components (or Fourier coefficients) defined over a discrete set of...
ACTIVE CONTROL OF SOUND RADIATION AND TRANSMISSION

wave-numbers, \( n = 0, 1, \ldots, +\infty \), while the transform along the axial direction maps a finite length axial distribution into a wave-number distribution extending from \(-\infty \) to \(+\infty \). Now considering a sensor based on the monopole approximation and a set of equally spaced measurement points, the resulting estimate can be expressed in terms of the associated discrete wave-number transforms as [15]

\[
p_d(R, \theta, \phi) = \frac{pe^{-kR}}{\pi k R \sin \theta} \sum_{n=0}^{+\infty} H_n^{(2)}(k \alpha \sin \theta) \left\{ \tilde{\omega}_{d,n}^s(k \cos \theta \cos(n \phi) + \tilde{\omega}_{d,n}^s(k \cos \theta \sin(n \phi)) \right\}
\]

where

\[
\begin{align*}
\tilde{\omega}_{d,n}^s(\gamma) &= \frac{\Delta \phi}{\pi} \sum_{m=1}^{N_z-1} \tilde{\omega}_d(\phi_m, \gamma) \cos(n \phi_m) \quad \text{and} \quad \tilde{\omega}_d(\phi, \gamma) = \Delta z \sum_{m=1}^{N_z-1} \tilde{\omega}(\phi, z_m) e^{i\gamma z_m} \\
\tilde{\omega}_{d,n}^s(\gamma) &= \frac{\Delta \phi}{\pi} \sum_{m=1}^{N_z-1} \tilde{\omega}_d(\phi_m, \gamma) \sin(n \phi_m)
\end{align*}
\]

Note that the actual radiated pressure is obtained by replacing \( \tilde{\omega}_{d,n}^s(\gamma) \) and \( \tilde{\omega}_{d,n}^s(\gamma) \) in equation (8) by their continuous equivalent

\[
\begin{align*}
\tilde{\omega}_{n}^s(\gamma) &= \frac{1}{\pi} \int_{-\infty}^{+\infty} \int_0^{2\pi} \tilde{\omega}(\phi, z) \cos(n \phi) e^{i\gamma z} d\phi \, dz \\
\tilde{\omega}_{n}^s(\gamma) &= \frac{1}{\pi} \int_{-\infty}^{+\infty} \int_0^{2\pi} \tilde{\omega}(\phi, z) \sin(n \phi) e^{i\gamma z} d\phi \, dz
\end{align*}
\]

The pressure radiated in the far field at a particular angle is associated with a single axial wave-number component, \( k \cos(\theta) \), within the supersonic region, \([-k, +k]\), and an infinite number of circumferential wave-numbers, \( n = 0, 1, \ldots, +\infty \). In practice, the infinite summation can be truncated based on the highest order of the circumferential modes included in the response and the range of the non-dimensional parameter, \( ka \), as the magnitude of the Hankel derivative tends towards infinity as \( n \) increases.

From equation (9), the accuracy of the sensor estimate is dictated by the levels of aliasing occurring in the axial and circumferential discrete wave-number transforms. The number of point measurements along the axial direction, \( N_z \), should be such that the Nyquist axial wave-number, \( K_a/2 = N_z \pi/(2L) \), remains above the main axial wave-number components of the acceleration distribution. Likewise, the circumferential Nyquist wave-number, \( K_f/2 = N_f/2 \), should be greater than the highest circumferential order of the modes found in the structural response. This requirement is a direct consequence of the sampling theorem commonly applied to the sampling of time domain signals. It should be pointed out that unlike the axial wave-number transform which extends up to infinity regardless of the spatial distribution due to the finite cylinder length, the circumferential wave-number transform only contains the components associated with modes found in the distribution. Therefore, while the axial discrete wave-number transform always result in some level of aliasing, the discrete circumferential wave-number transform will yield no aliasing, i.e., perfect estimates, provided all components above the Nyquist wave-number have zero amplitude. In addition, all or part of the wave-number components associated with circumferential wave-numbers above the Nyquist wave-number will yield large errors due to the periodicity of the discrete wave-number transform. Thus, the Nyquist circumferential wave-number should be high enough such that errors associated with higher wave-numbers are canceled by the large magnitude of the Hankel derivative term. Finally, it can be shown that the circumferential wave-number component estimate is independent of the origin of the point sensors locations along \( \phi \) provided the number of measurement points satisfies the sampling theorem along this direction. Note that similar trends are expected for the piston approximation.

Returning to Figure 3, the axial wave-number transform on resonance features a main peak around \( \gamma = m \pi/(2L) \) where \( m \) is the axial modal index. Consequently, the number of measurement points along the axial direction should be greater than the modal index of the mode dominating the response, i.e., \( N_z > m \). Similarly, the number of point measurements along the circumferential direction should be greater than twice the modal index of the associated mode, i.e., \( N_f > 2n \). Cases where at least one of the above conditions is not satisfied yield aliasing errors as shown in Figure 3. At off-resonance frequencies, the response includes higher order modes which deteriorate the accuracy of the estimate. Note that the discretization level is the critical parameter affecting the sensor estimate. In other words, good sensor accuracy is ensured over the entire radiating field as long as the number of measurement points is high enough relative to the dominant modes of the response. Finally, it should be mentioned that the above results assume perfect modeling of the
length axial sensor based simile can
sensor transfer functions. However, due to the relative simple characteristics of the sensor transfer functions, excellent modeling accuracy can be achieved with only a few FIR coefficients.

Experimental results This section briefly presents the accuracy of the sensor which is implemented experimentally on the cylinder. The cylinder is excited through the disturbance piezo-electric actuator with a band-limited random noise over 200–630 Hz. As true far-field conditions do not exist in the anechoic chamber especially at low frequencies due its limited size relative to the dimensions of the cylinder and the acoustic wave-length within the frequency bandwidth of excitation, the pressure radiated in the far field is “reconstructed” from the laser out-of-plane velocity measurements. Analogous to the real time pressure estimate implemented in the sensor, the pressure field is computed off-line from the structural laser measurements and the associated Green’s function replacing the continuous integral in equation (3) by its discrete approximation [15]. The magnitude of the reconstructed pressure at angle (76.8°, 0°) is compared to the associate sensor estimate in Figure 4. Recalling the system’s natural frequencies presented in Table 3, five flexural modes have their natural frequencies within the 200–630 Hz bandwidth. With increasing frequency, the five main resonance peaks noticed on the plot correspond to modes (1, 2), (1, 3), (2, 3), (1, 4), and (2, 4), respectively. Notice that the response also exhibits small contribution from “double” modes (1, 2)*, (2, 3)*, and (2, 4)*.

Examining the sensor output (dotted line), the pressure estimate shows good accuracy around the resonance frequency of the (1, 2) mode. A small variation of about 2 dB is observed at the resonance frequency of mode (1, 3). The reconstructed pressure is also relatively well estimated at off-resonance frequencies around the (1, 2) mode. The sensor accuracy then deteriorates as the frequency increases. A 6 dB variation between reconstructed and estimated pressure is noticed at the resonance frequency of mode (2, 3) while the estimated pressure at the resonance frequencies of the last two modes in the bandwidth (modes (1, 4) and (2, 4)) is off by more than 20 dB.

The above tendencies confirm the analytical results discussed in the previous section and agree well with the properties of the sensor estimate. As mentioned earlier, $N_z = 2$ measurement points along the axial direction yield good estimates for modes of axial order 1 or less. In the circumferential direction, $N_\phi = 6$ measurement points ensure accurate estimates of modes up to $n = 2$. That is to say, spatial distributions that are dominated by modes with axial and circumferential order larger than one and two, respectively, do not yield accurate pressure estimates.

![Figure 4: Reconstructed and estimated far-field pressure in direction (76.8°, 0°).](image)

RADIATION CONTROL

The following two sections discuss the analytical and experimental control results. In both cases, the cylinder is under broadband excitation through the disturbance actuator. In order to evaluate the performance of the structural acoustic sensor, results compare its control performances to those obtained with error microphones located in the far field. Note that the following analysis focuses on the influence of the “error” information on the control performances. In particular, the system is not optimized in terms of control input and/or error measurement locations to achieve the best possible attenuation in radiated power over the frequency bandwidth of excitation.
Analytical results This section presents two control cases based on the minimization of the estimated and actual radiated pressure in directions (70°, 0°), (110°, 40°), (90°, 160°), (70°, 240°), and (110°, 320°). These five radiation angles, (θ, φ), were chosen such as to obtain good global sound attenuation, i.e., none of the modes included in the frequency bandwidth have radiation nodal lines along all five angles. Both cases use the same control input configuration: the actuators locations are those given in Table 2 except for the circumferential angle of the third control actuator which was set to φ = 225°. All three control actuators are included in the following results. Note that this configuration yields an over-determined system such that the optimal control amplitudes are solved in the least square sense. The cost function thus does not include the control effort usually required to condition the solution of under-determined systems [19].

Figure 5 shows the radiated sound power before control (solid line) and after control based on the two cost functions. The dashed line corresponds to the minimization of the five pressure estimates using a 3 by 9 structural acoustic sensor (Nz = 3, Nφ = 9) while the dotted line corresponds to the minimization of the actual radiated pressure in the same directions. As seen on the controlled response, both sensing approaches yield very close control performances. The radiated sound power is attenuated over the entire bandwidth with very small control spillover. The total attenuation across the bandwidth is 20.3 dB for the structural acoustic sensor and 18.6 dB for the error microphones. These results are expected since the discretization level of the sensor ensures reduced aliasing errors for acceleration distributions including modes with axial and circumferential orders up to 2 and 4, respectively. In other words, all modes included in the bandwidth yield accurate pressure estimates. Note that minimizing the actual radiated pressure yields a slight decrease in overall sound attenuation thus suggesting the small errors introduced in the pressure estimates result in a slightly more global error information. These results show that the structural acoustic sensor implemented in this case can effectively replace the error microphones. Comparatively, a reduced number of acceleration measurement points would increase aliasing errors in the sensor estimate and in turn reduce the global sound attenuation levels.

It should be mentioned that the above results are based on an optimal control solution in the frequency domain which presents a number of limitations when used to predict the performance of a real time domain control system under broadband frequency disturbances. Specifically, the optimal control transfer functions are not constrained to yield realizable FIR filters. This often leads to over-estimating the controller performance and, in some cases, to control spillover not observed experimentally due to the finite number of coefficients in the control compensator among other factors. Consequently, the analytical results presented above do not accurately model the performance of the control system investigated experimentally. However, they still provide insight into the control performance of Discrete Structural Acoustic Sensing compared to other sensing approaches.

Experimental results This section discusses some of the control results obtained experimentally on the system described previously. The first two cases use the 2 by 6 structural acoustic sensor to provide error signals associated with pressure estimates in three directions. A third case uses three error microphones located along the same directions for comparison purpose. All three cases use the first two control actuators, leaving the third actuator unexcited. The radiated sound power presented below is estimated from the traverse microphone measurements over the sphere surrounding the cylinder.
The first control case corresponds to the minimization of the three sensor outputs associated with pressure estimates in directions (76.8°, 340°), (63.5°, 0°), and (103.2°, 0°), respectively. Figure 6 shows the radiated sound power before and after control. The dashed line corresponds to the standard configuration where a 20 sample delay is included in the disturbance path. The dotted line corresponds to an additional case based on the same control and error configuration with no delay included in the disturbance path. In both cases, the radiated sound power is attenuated by 20 dB or more near all resonance frequencies of the uncontrolled response. Small increases in sound power can be observed at off-resonance frequencies. Note that the dotted line associated with a possible acausal system does not show significant loss in performance compared to the dashed line. In other words, system causality does not appear to be a critical factor in this case. Due to the small levels of damping present in the system, the response is largely dominated by the five resonance frequencies of the modes included in the bandwidth. It is therefore highly predictable allowing control regardless of the system’s causality. The total reduction level in radiated sound power achieved across the frequency bandwidth of excitation is 15.4 dB with the 20 sample delay and 15.0 dB with no delay. Note that the three directions of minimization ensure good global control, i.e., no spillover is observed on the radiated sound power of the controlled response. In other words, the controller is forced to attenuate the amplitude of all modes in order to minimize all three error signals. Modal reduction is therefore the main control mechanism involved in this case.

The second control case based on the structural acoustic sensor uses three pressure estimates in directions (70°, 0°), (90°, 0°), and (110°, 0°). These three directions of minimization correspond to the locations of the error microphones implemented in the next control case. Figure 7(a) shows the radiated sound power before and after control. Excellent attenuation levels can be observed near resonance of modes (1,2), (1,3), (2,3) and (2,4) while significant control spillover is noticed near resonance of modes (1,2)* and (1,4)*. The total attenuation in radiated sound power is 5.9 dB. In this case, all directions of minimization are in the plane which coincides with radiation nodal line of modes (1,2)*, (1,4)* and (2,4)*. In other words these modes are not well observed by the three error signals. This explains the increase in radiated sound power noticed near resonance of mode (1,2)* and (1,4)*: rather than canceling the associated modal amplitudes, the controller recombines the modal amplitudes of the “double” modes and rotates the acceleration distribution such that the resulting nodal lines are aligned with the minimization angles. This modal restructuring mechanism is illustrated in Figure 7(b) where the out-of-plane velocity distribution measured by the laser vibrometer is shown before and after control at 541 Hz, i.e., near resonance of mode (1,4). The acceleration distribution after control is clearly rotated. Note that modal restructuring does not occur near resonance of mode (1,3) and (2,3). The first two control actuators are aligned with the anti-nodal lines of the associated “double” modes (1,3)* and (2,3)* therefore preventing their excitation. It should be pointed out that the previous case does not allow the above modal restructuring as the error signals do not correspond to pressure estimates along circumferential angles multiple of 45°, i.e., one error signal at least observes the “double” modes thereby forcing the controller to reduce the associated vibrations. These results therefore confirm the relatively good accuracy of the structural acoustic sensor implemented on the cylinder. Despite the small level of discretization ($N_z = 2$, $N_\phi = 6$) which only provides accurate estimates near resonance of mode (1,2), the sensor still yields error information somewhat related to the radiated sound pressure at higher frequencies. This is further illustrated by comparing these results with the next control case based on far-field error microphones.
The third control case replaces the outputs of the structural acoustic sensor by three error microphones located in the far field. To facilitate the comparison of the two sensing approaches, this case uses the same control configuration as previously. In particular, the three error microphones are located along the same directions of pressure estimates implemented in the previous case. The radiated sound power estimated from the traverse microphone measurements is presented before and after control in Figure 8. As for the case of pressure estimates (see Figure 7(a)), excellent global sound attenuation is achieved near resonance of modes (1,2), (1,3) and (2,3). Modal reduction occurs at these frequencies therefore canceling the sound pressure radiated over the entire field. Similarly, the sound power near resonance of mode (1,2)* increases as in the previous control case based on the structural acoustic sensor. Again, this mode has a radiation nodal line along $\phi = 0^\circ$ and thus remains unobserved by any of the three error microphones. Note that the control spillover at this frequency is larger than in the previous control case. As a result, the total sound radiated power across the frequency bandwidth of excitation increases by 0.7 dB. Similar behavior can be observed near resonance of modes (1,4)* and (2,4)*. This case clearly illustrate the importance of the control actuator and error sensor configurations. In particular, global sound attenuation for lightly damped cylindrical radiators can only be achieved through modal reduction. As a final remark, it should be mentioned that the controlled response based on pressure estimates in the same three directions and the

Figure 7: Minimization of pressure estimates in directions $(70^\circ, 0^\circ)$, $(90^\circ, 0^\circ)$, and $(110^\circ, 0^\circ)$. 

Figure 8: Radiated sound power - Minimization of pressure at error microphones in directions $(70^\circ, 0^\circ)$, $(90^\circ, 0^\circ)$, and $(110^\circ, 0^\circ)$. 

same two control actuators exhibits very similar trends as the ones obtained with error microphones. In particular, the structural acoustic sensor yields nearly the same response near resonance of modes (1, 2) and (1, 2)* which confirms the accuracy of the pressure estimates at these frequencies.

**CONCLUSIONS**

Broadband radiation control from a finite cylinder has been demonstrated both analytically and experimentally. The ASAC system implements a multi-channel Filtered-x LMS control algorithm. Structural control inputs are applied through piezo-electric actuators while error information is provided by a discrete structural acoustic sensor. Results show attenuation in total radiated sound power of up to 15 dB is achieved across a frequency bandwidth encompassing the first five flexural modes of the cylinder.

These results also validate the extension of Discrete Structural Acoustic Sensing to baffled cylindrical geometries. In this technique, the sensor constructs real time estimates of the pressure radiated in the far field at prescribed angles from discrete structural acceleration measurements and associated signal processing. Sensor accuracy and comparisons of control performances based on Discrete Structural Acoustic Sensing and the use of far-field error microphones, respectively, demonstrate the ability of the structural acoustic sensor to replace direct far-field pressure measurements. In particular, the analytical results showed that minimizing a cost function based on sensor estimates yields similar control performance to that obtained with a cost function based on actual far-field pressure, provided accuracy is ensured for the sensor. The experimental results on broadband radiation control confirms this result. Similar control performances are obtained for both types of error information. Examination of the system's response at single frequencies within the bandwidth of excitation also reveals the controller behaves similarly in each case. This includes frequencies where modal restructuring is the main control mechanism which confirms the ability of the sensor to provide accurate radiation information.

The accuracy of the sensor estimate over a given frequency bandwidth is primarily related to the number of structural measurements implemented in the sensor. In particular, for equally spaced measurement points, accuracy of the sensor estimates is ensured provided the following two conditions are satisfied. First, the number of measurement points along the cylinder main axis should be such that the associated Nyquist wave-number (half the spatial sampling frequency) is located on the wave-number axis above the main content of the wave-number transform of the structural out-of-plane motion along this axis. In terms of simply-supported mode shapes, this first requirement is satisfied if the number of measurement points along the main axis is greater than the highest order along the same axis of the modes which dominate the response. Second, the number of measurement points along the circumferential direction of the cylinder should be greater than twice the order of the highest circumferential order of the modes dominating the response. These conditions define the frequency bandwidth of accuracy for the structural acoustic sensor.

More generally, the design and implementation of the sensor do not require precise knowledge of the vibration characteristics of the structure. While some knowledge of the structural velocity distributions allows for predicting the accuracy of the pressure estimate and the appropriate discretization level, the sensor transfer functions strictly depend on the geometry of the structure and the surrounding fluid properties. This differs from a number of alternative structural sensing techniques whose design and implementation is directly based on dynamic properties of the structure, such as mode shapes. In comparison, Discrete Structural Acoustic Sensing provides a more robust error information. The modeling of the sensor transfer functions involves simple Finite Impulse Response (FIR) filters. This type of filter is well suited to represent the radiation of monopole and piston sources into an unbounded medium. In particular, the sensor transfer functions do not feature resonances, which would otherwise require the use of the more complex Infinite Impulse Response (IIR) filter model. Finally, when implemented in a far-field radiation control system, the time delay associated with the acoustic path can be removed from the sensor transfer functions without affecting the controller performance. This property greatly simplifies the sensor digital filters by reducing the number of coefficients required for accurate modeling of the sensor transfer functions. The sensor DSP implementation then becomes very efficient in terms of computations, allowing higher sampling frequencies or the combination of the sensor and controller code on a single DSP.
ACTIVE CONTROL OF SOUND RADIATION AND TRANSMISSION

ACKNOWLEDGMENTS

The authors gratefully acknowledge the Office of Naval Research, Dr. Kam Ng, Technical Monitor. The authors also wish to thank Kent Schuster and Scott Kelly at Reynold’s Metal Company in the Extrusion Division for donating the aluminum cylinder used in the experiments.

REFERENCES


ACTIVE CONTROL OF SOUND RADIATION FROM CYLINDERS
WITH PIEZOELECTRIC ACTUATORS AND
STRUCTURAL ACOUSTIC SENSING

J. P. Maillard and C. R. Fuller
Vibration and Acoustics Laboratories
Virginia Polytechnic Institute and State University
Blacksburg, Virginia, USA 24061-0238

45 pages
10 figures
4 tables
4 copies submitted
Running Headline: Active control of sound radiation from cylinders

Address for correspondence:

Dr. C. R. Fuller
Vibration and Acoustics Laboratories
Virginia Polytechnic Institute and State University
Blacksburg, Virginia, USA 24061-0238
In this paper, analytical and experimental results of an investigation of active control of sound radiated from cylinders are presented. The aluminum cylinder is 1 m in length, 25 cm in diameter and 2.4 mm in thickness with two rigid end-caps at both ends. The excitation is a band-limited random noise encompassing the first five modes of the cylinder and the control actuators are surface mounted piezoelectric transducers. Since it is desired to integrate the error sensors into the structure, the recently developed Discrete Structural Acoustic Sensing (DSAS) approach is extended to cylindrical coordinates and implemented using 12 accelerometers mounted on the cylinder. The structural acoustic sensor provides time domain estimates of far-field radiated sound at predetermined radiation angles. The controller is a 3 by 3 Filtered-x LMS paradigm implemented on a TMS320C30 DSP. The results show good global control of the radiated sound over the frequency bandwidth of excitation. Most important, the proposed discrete structural acoustic sensor yields similar performances as error microphones located in the far field. The sensor is also shown to improve far-field attenuation over minimization of normal acceleration at discrete locations on the cylinder structure. [Work supported by the Office of Naval Research]
1 INTRODUCTION

Much research has been conducted in the active control of low-frequency structure-borne sound. When compared to passive methods, active control presents significant advantages in the low-frequency range where passive control becomes often impractical due to prohibitive volume and/or mass requirements. For the past decade, Active Structural Acoustic Control (ASAC) has received much attention as it presents a practical alternative to the control of low-frequency structurally radiated noise [1,2]. In this technique, the radiated sound pressure is attenuated by applying mechanical inputs directly to the structure rather than by exciting the surrounding medium with acoustic sources (Active Noise Control). Piezoelectric devices have been applied extensively to Active Structural Acoustic Control systems as structural actuators [3–5] thus yielding a compact or “smart” structure. In an attempt to further reduce the size of the overall control arrangement, the microphones traditionally located in the far field to provide radiation error information are also being replaced by structural sensors such that all transducers are integrated in the structure.

As most ASAC applications involve noise control below the coincidence frequency of the radiating structure, appropriate structural sensors for ASAC should only observe the radiating part of the structural vibrations. This gives more flexibility to the controller which in some situations modifies the structural vibrations such as to attenuate far-field radiation with no net reduction in the overall vibration levels. Sound attenuation in the far field can then be achieved with a reduced control authority compared to cases where all structural motion is canceled (Active Vibration Control) [6].

With the emergence of polyvinylidene fluoride (PVDF) as a sensor material, several structural sensors for ASAC have been proposed to observe the radiating part of the structural vibrations. Most of these sensing techniques are based on modal sensing [7]: the sensor effectively observes a specific set of modes of vibration (natural or radiation modes of the uncontrolled response) which couples well to far-field radiation [8–11]. An accurate knowledge of the structure’s dynamic properties is therefore required. Recently, an alternative sensing technique referred to as Discrete Structural Acoustic Sensing (DSAS) was demonstrated both analytically [12, 13] and experimentally [14, 15] on baffled planar radiators. The technique implements an array of structural point sensors whose outputs are passed through digital
FIR filters to estimate in real time the far-field radiated pressure in a given direction, or equivalently, a given wave-number component, over a broad frequency range. It uses the relation between the structural out-of-plane vibrations and the far-field sound pressure as defined by the Helmholtz integral. One of the significant advantages of this strategy lies in its low modeling requirements compared to modal sensing approaches. In particular, the sensor design does not require the knowledge of the structural mode shapes and thus remains largely independent of the boundary conditions. Consequently, it is particularly well adapted for feedforward control approaches commonly used in ASAC systems where no analytical system modeling is necessary. It also provides time domain information which is required by the Filtered-x LMS algorithm commonly used in feedforward control. This paper presents analytical and experimental results on the extension of Discrete Structural Acoustic Sensing to baffled cylindrical radiators.

Most of the work on ASAC systems deals with planar geometries or systems that can be decomposed in a set of planar radiators and few reports of experiments on cylindrical structures can be found in the literature. Ruckman and Fuller [16] reported numerical simulations of ASAC applied to a finite cylinder system. Previous work by Clark and Fuller [17] studied experimentally the harmonic control of sound radiation from a finite enclosed cylinder using PVDF error sensors and piezoelectric actuators. Results showed the PVDF sensor was effective in observing the longitudinal extensional waves of the cylinder ("accordion" modes) and thus, good sound attenuation was obtained for this type of excitation. However, very little attenuation was achieved for radial excitations (normal bending modes) due to the relatively high modal density of the cylinder. The present study extends the above work by considering broadband radiation control over the first five bending modes of the structure. After briefly introducing the theoretical formulation, analytical and experimental results are presented. In both cases, the discrete structural acoustic sensor is first studied in terms of its accuracy to predict radiated pressure. Broadband radiation control results are then discussed by comparing the performances of the sensor to those of error microphones located in the far field.
This section presents the analytical formulation of Discrete Structural Acoustic Sensing. A relation between discrete structural acceleration and far-field pressure estimate is derived for the case of baffled cylindrical geometries.

2.1 FAR-FIELD SOUND PRESSURE

For the general case of arbitrary geometries, the sound pressure radiated from a vibrating structure into an unbounded medium can be expressed using the Kirchhoff-Helmholtz integral formulation [18]. Assuming a harmonic solution for the pressure, \( p(r)e^{\omega t} \), where \( \omega \) is the angular frequency, this is expressed as

\[
p(r) = \iint_{S_0} \left[ \rho G(r|r_0)\ddot{w}(r_0) + p(r_0)\frac{\partial G}{\partial \eta_0}(r|r_0) \right] dS(r_0), \quad r \in V
\]

(1)

In the above equation, \( S_0 \) denotes the radiating surface and \( V \) the surrounding volume. The sound pressure \( p(r) \) at field point \( r \) is expressed as a surface integral involving the out-of-plane structural acceleration, \( \ddot{w}(r_0) \), measured at location \( r_0 \) on the radiating surface, the surface pressure \( p(r_0) \), the Green's function, \( G(r|r_0) \), and its normal gradient (\( \partial G/\partial \eta_0 \) represents the component of the gradient of \( G \) along the unit vector \( \eta \) normal to \( S_0 \)), and the fluid density, \( \rho \). Note that this formulation assumes the radiator has solid boundaries such that the fluid velocity on the boundary is equal to the structural out-of-plane velocity. The normal pressure gradient then becomes equal to \( \rho \ddot{w}(r_0) \). Discrete Structural Acoustic Sensing is based on the existence of a Green's function satisfying the Neumann boundary condition,

\[
\frac{\partial G}{\partial \eta_0}(r|r_0) = 0, \quad r \in S_0
\]

(2)

such that the radiated pressure field becomes solely dependent on the structural acceleration and geometry.

For baffled cylindrical geometries (see Figure 1), a closed-form solution exists for the Green's function verifying equation (2). The radiated pressure is then expressed in the
cylindrical coordinate system \((R, \theta, \phi)\) as

\[
p(R, \theta, \phi) = \rho \int_{L}^{2\pi} \int_{-L}^{+L} G(R, \theta, \phi | r = a, \phi', z') \tilde{w}(\phi', z') a \, d\phi' \, dz' 
\]  

(3)

The Green's function in equation (3) can be obtained from the radiated pressure due to a point acceleration distribution located on an infinite cylindrical baffle. The resulting expression is approximated in the far field as [18]

\[
G(R, \theta, \phi | a, \phi_0, z_0) = \frac{\exp[-jk(R - z_0 \cos \theta)]}{\pi^2 akR \sin \theta} \sum_{n=0}^{\infty} \frac{j^{n+1} \cos [n(\phi - \phi_0)]}{\varepsilon_n H^{(2)}_n(ka \sin \theta)}
\]

(4)

where \(\varepsilon_n = 2\) for \(n = 0\) and \(\varepsilon_n = 1\) for \(n > 0\) (\(n\) integer). The function \(H^{(2)}_n(x)\) denotes the first derivative of the \(n^{th}\) Hankel function of the second kind [19]. Here \(n\) represents the circumferential modal order. The acoustic wave-number is denoted as \(k = \omega/c\) where \(c\) is the speed of sound. The various coordinates and dimensions involved in equation (4) are shown in Figure 1.

2.2 SENSOR ESTIMATE

An estimate of the radiated pressure in equation (3) is now constructed. The integral over the radiating surface \(S_0\) is approximated using a \(Q\) point zero-order interpolation of the acceleration distribution [15], i.e., the acceleration is assumed constant over \(Q\) small elemental surfaces, \(S_q, q = 1, 2, \ldots, Q\), such that \(S = \bigcup_{q=1}^{Q} S_q\). The resulting pressure estimate takes the general form

\[
p_d(R, \theta, \phi, t) = \sum_{q=1}^{Q} \tilde{w}(\phi_q, z_q, t) H_q(R, \theta, \phi)
\]

(5)

where \((\phi_q, z_q)\) represents the coordinates of the \(q^{th}\) node, and \(H_q(R, \theta, \phi)\), the associated sensor transfer function. Defining \(S_q = a \Delta z_q \Delta \phi_q\) as the \(q^{th}\) elemental surface aligned with the axial and circumferential directions such that its center coincides with \((\phi_q, z_q)\), the sensor
transfer functions can be expressed as

$$H_q(R, \theta, \phi) = \rho \int_{\phi_q - \Delta \phi_q/2}^{\phi_q + \Delta \phi_q/2} \int_{z_q - \Delta z_q/2}^{z_q + \Delta z_q/2} G(R, \theta, \phi|a, \phi_q, z_q) a d\phi dz$$  \hspace{1cm} (6)$$

The transfer function \(H_q(R, \theta, \phi)\) can be interpreted physically as the sound pressure radiated at \((R, \theta, \phi)\) from the \(q^{th}\) elemental surface vibrating along the normal to its center \((\phi_q, z_q)\) with a unit acceleration. In other words, the pressure estimate is constructed by summing the radiation contribution of \(Q\) cylindrical pistons weighted by the measured acceleration amplitudes. It is thus referred to as the piston approximation. Assuming the Green's function remains almost constant over each surface \(S_q\), the transfer function in equation (6) can be replaced by \(\rho a \Delta z_q \Delta \phi_q G(R, \theta, \phi|a, \phi_q, z_q)\). In this case, the far-field pressure is estimated from the contribution of \(Q\) monopole sources (monopole approximation). As expected, both approximations become equivalent as \(k \max(\Delta z_q, \Delta \phi_q) \ll 1\).

It should be stressed that the sensor transfer functions solely depend on the geometry of the problem and the properties of the fluid medium. No accurate knowledge of the structure's dynamics (e.g., natural mode shapes) is thus required for their design. Note however that some information is still needed in order to determine an appropriate discretization level for accurate estimates. Furthermore, the sensing approach can be extended to geometries for which no Green's function is available analytically. The far-field pressure radiated from each elemental surface vibrating independently on the structure's boundary must then be solved numerically using a technique such as the Boundary Element Method [20].

3 PRACTICAL IMPLEMENTATION

This section briefly recalls some of the important issues associated with the practical implementation of Discrete Structural Acoustic Sensing. The pressure estimate presented in the previous section is implemented on a real system using a set of accelerometers mounted on the structure and arrays of digital filters. More precisely, each measured acceleration signal is passed through a digital filter modeling the associated sensor transfer function. All filter outputs are then summed to provide the sound pressure estimate. Several arrays of filters
can be implemented in order to provide pressure estimates at different locations. This ar-
angement is shown in Figure 2 along with a schematic of the controller based on the three
channel Filtered-x LMS algorithm.

As explained above, each transfer function represents the far-field radiation into an un-
bounded medium from a cylindrical piston source (or monopole) with unit acceleration and
located on a cylindrical baffle. The associated characteristics directly motivate the use of
Finite Impulse Response (FIR) filters to model the sensor transfer functions. In particular,
no resonance behavior occurs due to the assumption of an unbounded medium at infinity and
notches in the transfer functions magnitude associated with zero pressure angles are easily
modeled by appropriate zeros in the filter’s impulse response.

Another important issue is related to the time delay of the sensor transfer functions, which
is directly related to the acoustic path propagation time, $R/c$. As the pressure estimate is
only valid in the far field, this delay can become significant compared to the sampling period
of the digital filter, thus increasing its complexity. The authors have shown in previous work
however that error signals based on far-field pressure at a given location can be shifted in time
without loss of performance of the control system (the time shift is equivalent to moving the
minimization point along a constant radiation angle) [15]. Removing the above time delay
yields transfer functions with a minimum phase delay which significantly reduces the number
of FIR filter coefficients required for accurate modeling.

4 SYSTEM CHARACTERISTICS AND EXPERIMENTAL SETUP

Testing of the structural acoustic sensor described above was performed on a finite alu-
minum cylinder. This section presents the main characteristics of the system, the control
and measurement setups implemented in the experiments as well as the numerical model of
the structure.
4.1 SYSTEM CHARACTERISTICS

Due to a limited number of accelerometers available for implementing Discrete Structural Acoustic Sensing, the choice of the cylinder’s dimensions and material was made such that the first few flexural modes of the structure have low modal order in both the axial and circumferential directions. The cylinder characteristics are given in Table 1. The dimensions were measured on the actual structure while the material properties are based on standard values for aluminum. In order to allow structural vibration measurements over the entire surface of the cylinder as well as acoustic measurements over the sphere surrounding the structure, the cylinder’s attachment to its support stand is designed to allow full rotation along its main axis. The cylinder is closed at both ends by aluminum end-caps 12.7 mm in thickness. Each end-cap is attached to the cylinder with a set of 12 small screws equally spaced along the circumference. A steel rod 3.18 mm in diameter is threaded into each end-cap and mounted on a nylon ring. The ring is fixed in a 19.1 mm aluminum section which is bolted to a heavy steel support stand. To allow acoustic measurements of the baffled structure, two sections of “rigid” PVC pipe are installed on each side of the cylinder along its main axis. The two pipes extend in length up to the walls of the anechoic chamber. A picture of the complete rig including the baffle is shown in Figure 3.

All structural vibration inputs, i.e., both disturbance and control inputs, are applied through single-sided piezoelectric actuators. No curved actuators were available for the experimental testing and flat actuators [21] were mounted on the cylinder by cutting them across their width into a set of eight strips of same dimensions. The original actuators are 63.5 mm in length and 38.1 mm in width, which results in eight 7.9 mm by 38.1 mm strips. They are mounted on the cylinder’s outside surface side by side with their length along the cylinder’s axis and wired in phase. A gap of approximately 1 mm remains between each actuator strip to avoid short circuits between the electrodes of two adjacent actuators. The total surface area covered by the set of eight actuators is 69.9 mm by 38.1 mm. Table 1 presents the dimensions and material properties of the piezoelectric actuators. Four sets of the actuator arrangement described above are mounted on the cylinder according to the center locations given in Table 2. The disturbance actuator center location serves as the origin of the circumferential direction, $\phi = 0^\circ$. Its axial location, $z/L = -0.328$, ensures
that all flexural modes present in the 0–1000 Hz bandwidth are excited. The three other actuator sets are implemented as control inputs. Their location was determined in order to allow various control test configurations of interest.

The discrete structural acoustic sensor implemented on the cylinder uses 12 accelerometers (PCB Piezo-electric ICP accelerometers - Model 352A10). The point sensors are arranged as 2 rings of 6 accelerometers equally spaced according to

\[
\begin{align*}
\phi_{q_1} &= q_1 \Delta \phi, & q_1 = 0, 1, \ldots, Q_\phi - 1, & \Delta \phi = 2\pi / Q_\phi \\
z_{q_2} &= -L + \Delta z / 2 + q_2 \Delta z, & q_2 = 0, 1, \ldots, Q_z - 1, & \Delta z = 2L / Q_z
\end{align*}
\]

where \(Q_z = 2\) and \(Q_\phi = 6\) are the number of points along the axial and circumferential direction, respectively. Note that the accelerometers are aligned with the disturbance actuator such that the modes excited by the disturbance input have anti-nodes along the circumferential directions aligned with the point sensor locations. As it will be discussed later, the sensor accuracy is independent of the point sensor circumferential locations when the number of sensors along the circumferential direction is greater than the highest circumferential order of the modes present in the bandwidth. When this condition is not satisfied however, care must be taken so that the point sensors do not coincide with the nodal lines of a given mode (mode \(n = 3\) in this case). Furthermore, placing the sensors on anti-nodal lines ensures better signal to noise ratio thus improving the accuracy of the sensor estimate.

4.2 CONTROL AND MEASUREMENT SETUPS

For all experimental testing, the cylinder is excited through the disturbance actuator with a band-limited random noise. The sensor accuracy tests are performed over a 200–630 Hz bandwidth while the control tests use a 200–500 Hz bandwidth. The reduced bandwidth associated with the control tests ensures that the 2 by 6 sensor (i.e., 12 accelerometers total) yields relatively accurate estimates over the frequency range. A three channel Filtered-x LMS algorithm [22] is implemented on a Texas Instrument TMS320C30 digital signal processor (DSP) to provide up to three control signals (see Figure 2). The results presented in this paper are limited to cases where only the first and second control actuators were excited (see
Table 2). The controller’s reference signal is taken from the signal fed to the disturbance actuator. Most of the control tests include an artificial delay of 20 samples in the disturbance path so as to improve system causality. A few cases were also run with a zero delay in order to evaluate the influence of system causality on the control performance. All control tests use the following settings: the sampling frequency is set to $F_s = 2000 \text{ Hz}$, the FIR compensators have 50 coefficients, and the IIR filters modeling the filtered-$x$ path transfer functions have 60 coefficients in both numerator and denominator. All tests use three error signals based on the structural acoustic sensor and far-field error microphones (B&K 1/2 in microphone, Model 4166), respectively. The sensor’s array of filters is implemented on a second TMS320C30 digital DSP. Note that both controller and sensing code could be implemented on a single DSP if desired. Three error microphones are also located along $\theta = 70^\circ$, $\theta = 90^\circ$, and $\theta = 110^\circ$, in the $\phi = 0^\circ$ plane at $R = 1.85 \text{ m}$ while several sets of FIR filters were designed to provide pressure estimates for various radiation angles. All sensor FIR filters have 22 coefficients with a sampling frequency, $F_s = 6000 \text{ Hz}$. The sensor transfer functions are accurately modeled up to about 2500 Hz. This wide bandwidth relative to the actual bandwidth of excitation was found necessary in order to ensure stability for the control system. Cases where the response of the sensor filters is not constrained at higher frequencies can lead to unwanted amplification of high frequency content remaining in the system due to the finite roll-offs of the low-pass filters.

All tests were conducted in a 4.2 m by 2.2 m by 4.5 m anechoic chamber at the Vibration and Acoustics Laboratories (VAL), Virginia Tech. The chamber has an approximate cut-off frequency of 250 Hz. Out-of-plane structural vibrations are measured with a Politec laser vibrometer (Model OFV-2600/OFV-501). To allow measurements over the entire radiating surface, the laser head is mounted on a one-dimensional linear traverse driven by a stepper motor while a second stepper motor mounted on the end-cap assembly rotates the cylinder around its main axis. The structural velocity measurements use a grid of 13 points along the axial direction and 18 point along the circumferential direction. The measurement point locations are defined by equation (7) with $N_z = 13$ and $N_\phi = 18$. This discretization level proved to be sufficient to accurately measure the structural response over the frequency bandwidth of excitation. The sound pressure radiated from the cylinder is measured inside
the anechoic chamber with a B&K 1/2 in microphone (Model 4166) mounted on a circular traverse. The traverse microphone is located at a radius of 1.85 m from the center of the cylinder. Due to the legs of the support stand, the traverse can only move in the $x$-$z$ plane from $\theta = 10^\circ$ to $\theta = 170^\circ$. The rotation of the cylinder also allows measurements along the circumferential direction from $\phi = 0^\circ$ to $\phi = 360^\circ$. All far-field measurements use a grid of $N_\theta = 13$ points along the azimuthal direction, $\theta$, and $N_\phi = 18$ points along the circumferential direction, $\phi$.

4.3 NUMERICAL SIMULATIONS

In order to study various sensor configurations, numerical simulations were performed prior to the experiments described above. The cylinder structure is modeled under steady-state harmonic excitation of point forces and piezoelectric actuators with a variational approach implementing the Rayleigh-Ritz formulation [15]. In this model, the mechanical displacements and electrical fields within the piezoelectric actuators are fully coupled thus including the mass and stiffness loading of the actuators. This energy based formulation also allows modeling of arbitrary boundary conditions applied along the edges of the cylinder. To this purpose, the model includes translational and rotational springs along the axial, circumferential, and radial directions. The stiffness factor of each spring can then be adjusted to model arbitrary conditions. The reader is referred to [15] for a complete description of the model.

The optimal control voltage to each actuator is computed using standard Linear Quadratic Optimal Control theory [2], where the cost function to be minimized is a quadratic function of the control voltage amplitudes.

Table 3 presents the natural frequencies of the first few modes of the cylinder as obtained from the numerical model and experimental modal analysis of the test structure, respectively. Examination of the associated mode shapes reveals the structure's attachment creates approximately simply-supported boundary conditions where the first modal index, $m$, is associated with the axial direction, and the second index, $n$, with the circumferential direction. Note that each mode of vibration is associated with two distinct natural frequencies and mode shapes (i.e., $\cos(n\theta)$ and $\sin(n\theta)$ angular variation) rotated along the circumferential direction by $\pi/(2n)$ relative to one another. This behavior is expected due to the asymmetry
introduced by the added mass and stiffness of the piezoelectric actuators. Disregarding the mismatch of mode (1, 1), good agreement between numerical and experimental natural frequencies can be observed. Including the circumferential dependence of the stiffness factors used to model the boundary conditions would possibly improve the match especially for the (1, 1) "beam" mode.

5 SENSOR ACCURACY

This section successively presents analytical and experimental results showing the accuracy of the structural acoustic sensor. The sensor estimate is compared to the actual sound pressure radiated in the far field over the 200–680 Hz bandwidth.

5.1 ANALYTICAL RESULTS

Figure 4 shows the magnitude of the far-field radiated pressure in direction \((\theta, \phi) = (70^\circ, 240^\circ)\) (solid line) along with the sensor estimate based on the piston approximation and two different point sensor configurations. Both results were calculated using the analytical model of the cylinder and its associated radiation field which are described in detail in Reference [15]. The "actual" pressure corresponds to the full analytical prediction while the sensor estimate is the far-field pressure estimated using the analytical model in conjunction with the structural acoustic sensor theory of Section 2.2. The dashed line corresponds to a 8 by 7 sensor, i.e., the acceleration measurement points are located according to equation (7) with \(Q_x = 8\) and \(Q_\phi = 7\), and the dotted line to a 2 by 12 sensor. Recalling the natural frequencies given in Table 3, the resonance peaks correspond, as frequency increases, to modes (1, 2), (1, 3), (2, 3), (1, 4), (1, 1) and (2, 4) where the first and second index refers to the axial and circumferential order, respectively. Note the unusual characteristics of cylinders for which the fundamental mode (1, 1) is not associated with the lowest resonance frequency [18]. As seen on the plot, the 8 by 7 sensor yields excellent accuracy at the resonance frequency of all modes included in the bandwidth except for modes (1, 4) and (2, 4) where large errors can be observed. Considering the 2 by 12 sensor, excellent accuracy is obtained at the resonance frequency of all modes.
except for a small error near resonance of modes (2, 3) and (2, 4). These results illustrate two fundamental properties of the sensor estimate for cylindrical geometries as outlined below.

Analogous to the case of planar radiators, the far-field radiated pressure in equation (3) can be expressed in terms of the two-dimensional wave-number transform of the structural out-of-plane acceleration [15]. The wave-number transform along the circumferential direction maps a periodic spatial distribution with period $2\pi$ into two sets of wave-number components (or Fourier coefficients) defined over a discrete set of wave-numbers, $n = 0, 1, \ldots, +\infty$, while the transform along the axial direction maps a finite length axial distribution into a wave-number distribution extending from $-\infty$ to $+\infty$. Now considering a sensor based on the monopole approximation and a set of equally spaced measurement points, the resulting estimate can be expressed in terms of the associated discrete wave-number transforms as [15]

$$p_d(R, \theta, \phi) = - \frac{\rho e^{-jkR}}{\pi kR \sin \theta} \sum_{n=0}^{+\infty} \frac{j^{n+1}}{H_n^{(2)}(ka \sin \theta)} \left\{ \tilde{w}_{d,n}^c(k \cos \theta) \cos(n\phi) + \tilde{w}_{d,n}^s(k \cos \theta) \sin(n\phi) \right\} \tag{8}$$

where

$$\left\{ \begin{array} {c}
\tilde{w}_{d,n}^c(\gamma) = \frac{\Delta \phi}{\pi \epsilon_n} \sum_{q_1=0}^{Q-1} \tilde{w}_d(\phi_{q_1}, \gamma) \cos(n\phi_{q_1}) \\
\tilde{w}_{d,n}^s(\gamma) = \frac{\Delta \phi}{\pi \epsilon_n} \sum_{q_1=0}^{Q-1} \tilde{w}_d(\phi_{q_1}, \gamma) \sin(n\phi_{q_1})
\end{array} \right. \quad \text{and} \quad \tilde{w}_d(\phi, \gamma) = \Delta z \sum_{\eta_2=0}^{Q-1} \tilde{w}(\phi, z_{\eta_2}) e^{i\eta_2 \gamma} \tag{9}$$

Note that the actual radiated pressure is obtained by replacing $\tilde{w}_{d,n}^c(\gamma)$ and $\tilde{w}_{d,n}^s(\gamma)$ in equation (8) by their continuous equivalent

$$\left\{ \begin{array} {c}
\tilde{w}_n^c(\gamma) = \frac{1}{\epsilon_n \pi} \int_{-\infty}^{+\infty} \int_0^{2\pi} \tilde{w}(\phi, z) \cos(n\phi) e^{i\gamma z} \, d\phi \, dz \\
\tilde{w}_n^s(\gamma) = \frac{1}{\epsilon_n \pi} \int_{-\infty}^{+\infty} \int_0^{2\pi} \tilde{w}(\phi, z) \sin(n\phi) e^{i\gamma z} \, d\phi \, dz \end{array} \right. \tag{10}$$

The pressure radiated in the far field at a particular angle is associated with a single axial wave-number component, $k \cos(\theta)$, within the supersonic region, $[-k, +k]$, and an infinite number of circumferential wave-numbers, $n = 0, 1, \ldots, +\infty$. In practice, the infinite summation can be truncated based on the highest order of the circumferential modes included in
the response and the range of the non-dimensional parameter, \( ka \), as the magnitude of the Hankel derivative tends towards infinity as \( n \) increases.

From equation (9), the accuracy of the sensor estimate is dictated by the levels of aliasing occurring in the axial and circumferential discrete wave-number transforms. The number of point measurements along the axial direction, \( Q_z \), should be such that the Nyquist axial wave-number, \( K_z/2 = Q_z\pi/(2L) \), remains above the main axial wave-number components of the acceleration distribution. Likewise, the circumferential Nyquist wave-number, \( K_\phi/2 = Q_\phi/2 \), should be greater than the highest circumferential order of the modes found in the structural response. This requirement is a direct consequence of the sampling theorem commonly applied to the sampling of time domain signals. It should be pointed out that unlike the axial wave-number transform which extends up to infinity regardless of the spatial distribution due to the finite cylinder length, the circumferential wave-number transform only contains the components associated with modes found in the distribution. Therefore, while the axial discrete wave-number transform always result in some level of aliasing, the discrete circumferential wave-number transform will yield no aliasing, i.e., perfect estimates, provided all components above the Nyquist wave-number have zero amplitude. In addition, all or part of the wave-number components associated with circumferential wave-numbers above the Nyquist wave-number will yield large errors due to the periodicity of the discrete wave-number transform. Thus, the Nyquist circumferential wave-number should be high enough such that errors associated with higher wave-numbers are canceled by the large magnitude of the Hankel derivative term. Finally, it can be shown that the circumferential wave-number component estimate is independent of the origin of the point sensors locations along \( \phi \) provided the number of measurement points satisfies the sampling theorem along this direction. Note that similar trends are expected for the piston approximation.

Returning to Figure 4, the axial wave-number transform on resonance features a main peak around \( \gamma = m\pi/(2L) \) where \( m \) is the axial modal index. Consequently, the number of measurement points along the axial direction should be greater than the modal index of the mode dominating the response, i.e., \( Q_z > m \). Similarly, the number of point measurements along the circumferential direction should be greater than twice the modal index of the associated mode, i.e., \( Q_\phi > 2n \). Cases where at least one of the above conditions is not satisfied
yield aliasing errors as shown in Figure 4. At off-resonance frequencies, the response includes higher order modes which reduce the accuracy of the estimate. Note that the discretization level is the critical parameter affecting the sensor estimate. In other words, good sensor accuracy is ensured over the entire radiating field as long as the number of measurement points is high enough relative to the dominant modes of the response. Finally, it should be mentioned that the above results assume perfect modeling of the sensor transfer functions. However, due to the relative simple characteristics of the sensor transfer functions, excellent modeling accuracy can be achieved with only a few FIR coefficients [15]. Other analytical results, not presented here for brevity, also confirm that Discrete Structural Acoustic Sensing will provide good estimates of radiated pressure as long as no significant aliasing occurs.

5.2 EXPERIMENTAL RESULTS

This section briefly presents the accuracy of the 2 by 6 sensor which is implemented experimentally on the cylinder. The cylinder is excited through the disturbance piezoelectric actuator with a band-limited random noise over 200–630 Hz. As true far-field conditions do not exist in the anechoic chamber especially at low frequencies due its limited size relative to the dimensions of the cylinder and the acoustic wave-length within the frequency bandwidth of excitation, the pressure radiated in the far field is "reconstructed" from the laser measurements of the cylinder out-of-plane velocity. Analogous to the real time pressure estimate implemented in the sensor, the pressure field is computed off-line from the structural laser measurements and the associated Green’s function replacing the continuous integral in equation (3) by its discrete approximation [15]. However in this case, the discretization level is much higher with a grid of 13 by 18 measurement points (see Section 4.2) thus ensuring a high fidelity in the predicted far-field sound pressure. The magnitude of the reconstructed pressure at an angle \((\theta, \phi) = (76.8^\circ, 0^\circ)\) is compared to the associate sensor estimate in Figure 5. Recalling the system’s natural frequencies presented in Table 3, five flexural modes have their natural frequencies within the 200–630 Hz bandwidth. With increasing frequency, the five main resonance peaks noticed on the plot correspond to modes \((1, 2), (1, 3), (2, 3), (1, 4), \) and \((2, 4), \) respectively. Notice that the response also exhibits small contribution from “double” modes \((1, 2)^*, (2, 3)^*, \) and \((2, 4)^*. \)
Examining the sensor output (dotted line), the pressure estimate shows good accuracy around the resonance frequency of the (1, 2) mode. A small variation of about 2 dB is observed at the resonance frequency of mode (1, 3). The reconstructed pressure is also relatively well estimated at off-resonance frequencies around the (1, 2) mode. The sensor accuracy then deteriorates as the frequency increases. A 6 dB variation between reconstructed and estimated pressure is noticed at the resonance frequency of mode (2, 3) while the estimated pressure at the resonance frequencies of the last two modes in the bandwidth (modes (1, 4) and (2, 4)) is off by more than 20 dB.

The above tendencies confirm the analytical results discussed in the previous section and agree well with the properties of the sensor estimate. As mentioned earlier, $Q_z = 2$ measurement points along the axial direction yield good estimates for modes of axial order 1 or less. In the circumferential direction, $Q_\phi = 6$ measurement points ensure accurate estimates of modes up to $n = 2$. That is to say, spatial distributions that are dominated by modes with axial and circumferential order larger than one and two, respectively, do not yield accurate pressure estimates. Increasing the number of point structural sensors would in turn increase the accuracy of the sensor prediction for these higher order modes.

6 RADIATION CONTROL

The following two sections discuss the analytical and experimental control results. In both cases, the cylinder is under broadband excitation through the disturbance actuator. In order to evaluate the performance of the structural acoustic sensor, results are presented which compare its control performances to those obtained with error microphones located in the far field. Note that the following analysis focuses on the influence of the “error” information on the control performances. In particular, the system is not optimized in terms of control input and/or error measurement locations to achieve the best possible attenuation in radiated power over the frequency bandwidth of excitation.
6.1 ANALYTICAL RESULTS

This section presents three control cases calculated using the analytical model. The first two cases are based on the minimization of the estimated and actual radiated pressure in directions \((\theta, \phi) = (70^\circ, 0^\circ), (110^\circ, 40^\circ), (90^\circ, 160^\circ), (70^\circ, 240^\circ),\) and \((110^\circ, 320^\circ)\). These five radiation angles were chosen such as to obtain good global sound attenuation, i.e., none of the modes included in the frequency bandwidth have radiation nodal lines along all five angles. A third control case involves the direct minimization of the out-of-plane acceleration measured at the point structural sensor locations corresponding to the Active Vibration Control case for this system. Note that for the AVC case, the number of error signals then equals the number of accelerometers implemented in the sensor, i.e., 12, while the ASAC cases use five error signals corresponding to the directions of pressure estimates. All three cases use the same control input configuration: the actuators locations are those given in Table 2 except for the circumferential angle of the third control actuator which was set to \(\phi = 225^\circ\). All three control actuators are implemented as control inputs in the following results. Note that this configuration yields an over-determined system such that the optimal control amplitudes are solved in the least square sense. The cost function thus does not include the control effort usually required to condition the solution of under-determined systems [22].

Figure 6 shows the calculated radiated sound power before control (solid line) and after control based on the three cost functions. The dashed line corresponds to the minimization of the five pressure estimates using a 3 by 9 structural acoustic sensor \((Q_x = 3, Q_\phi = 9)\), the dotted line corresponds to the minimization of the actual radiated pressure in the same directions, and the dashed-dotted line corresponds to the minimization of the out-of-plane acceleration at the point sensor locations (see equation (7)).

As seen on the controlled response, the first two sensing approaches (dashed and dotted lines) yield very close control performances. The radiated sound power is attenuated over the entire bandwidth with very small control spillover. The total attenuation across the bandwidth is 20.2 dB for the structural acoustic sensor and 19.4 dB for the error microphones. These results are expected since the discretization level of the sensor ensures reduced aliasing errors for acceleration distributions including modes with axial and circumferential orders up to 2 and 4, respectively. In other words, all modes included in the bandwidth yield
accurate pressure estimates. Note that minimizing the actual radiated pressure yields a slight decrease in overall sound attenuation thus suggesting the small errors introduced in the pressure estimates result in a slightly more global error information. These results show that the structural acoustic sensor implemented in this case can effectively replace the error microphones. Comparatively, a reduced number of acceleration measurement points would increase aliasing errors in the sensor estimate and in turn reduce the global sound attenuation levels.

The third cost function associated with the minimization of the discrete acceleration over 27 points (dashed-dotted line) does not perform as well as the two previous cost functions based on radiation information. While similar levels of attenuation are achieved at on-resonance frequencies (except near resonance of the (1, 1) mode), the control performance significantly deteriorates at off-resonance frequencies. This behavior suggests that some level of modal restructuring occurs off-resonance when minimizing radiation information. In this case, the controller achieves attenuation in the far field by combining several structural modes that are present in the uncontrolled response. The overall vibration levels do not necessarily decrease in this case. At on-resonance frequencies, a single mode dominates the structural response and radiation control is then achieved through modal reduction. To further illustrate these results, the overall attenuation levels of mean-square velocity and sound radiated power are presented in Table 4 for the three cost functions. Again, the first two cost functions based on radiation information perform almost identically. The third cost function based on structural information increases the overall attenuation level in mean-square velocity by more than 10 dB. However, the attenuation in sound radiated power decreases by about 8 dB. In other words, minimizing the structural vibrations yields, in this case, decreased control performances in terms of radiated sound attenuation when compared to the use of structural acoustic sensing or error microphones.

It should be mentioned that the above results are based on an optimal control solution in the frequency domain which presents a number of limitations when used to predict the performance of a real time domain control system under broadband frequency disturbances. Specifically, the optimal control transfer functions are not constrained to yield realizable FIR filters. This often leads to over-estimating the controller performance and, in some cases,
to control spillover not observed experimentally due to the finite number of coefficients in
the control compensator among other factors. Consequently, the analytical results presented
above do not accurately model the performance of the control system investigated experi-
tently. However, they still provide insight into the control performance of Discrete Structural
Acoustic Sensing compared to other sensing approaches and also indicate the maximum
achievable performance.

6.2 EXPERIMENTAL RESULTS

This section discusses some of the control results obtained experimentally on the system
described previously. The first two cases use the 2 by 6 structural acoustic sensor to provide
error signals associated with pressure estimates in three directions. A third case uses three
error microphones located along the same directions for comparison purpose. All three cases
use the first two control actuators, leaving the third actuator unexcited. The radiated sound
power presented below is estimated from the multiple traverse microphone measurements
over the sphere surrounding the cylinder.

The first control case corresponds to the minimization of the three sensor outputs associ-
ated with pressure estimates in directions \((\theta, \phi) = (76.8^\circ, 340^\circ), (63.5^\circ, 0^\circ),\) and \((103.2^\circ, 0^\circ),\)
respectively. Figure 7 shows the radiated sound power before and after control. The dashed
line corresponds to the standard configuration where a 20 sample delay is included in the dis-
turbance path. The dotted line corresponds to an additional case based on the same control
and error configuration with no delay included in the disturbance path. In both cases, the
radiated sound power is attenuated by 20 dB or more near all resonance frequencies of the
uncontrolled response. Small increases in sound power can be observed at off-resonance fre-
quencies for both cases. Note that the dotted line associated with a possible acausal system
does not show significant loss in performance compared to the dashed line. In other words,
system causality does not appear to be a critical factor in this case. Due to the small levels
of damping present in the system, the response is largely dominated by the five resonance
frequencies of the modes included in the bandwidth. It is therefore highly predictable allow-
ing control regardless of the system's causality. The total reduction level in radiated sound
power achieved across the frequency bandwidth of excitation is 15.4 dB with the 20 sample
delay and 15.0 dB with no delay. Note that the three directions of minimization ensure good global control, i.e., no spillover is observed on the radiated sound power of the controlled response. In other words, the controller is forced to attenuate the amplitude of all modes in order to minimize all three error signals. Modal reduction is therefore the main control mechanism involved in this case.

The second control case based on the structural acoustic sensor uses three pressure estimates in directions $(70^\circ, 0^\circ)$, $(90^\circ, 0^\circ)$, and $(110^\circ, 0^\circ)$. These three directions of minimization correspond to the locations of the error microphones implemented in the next control case. Figure 8 shows the radiated sound power before and after control. Excellent attenuation levels can be observed near resonance of modes $(1, 2)$, $(1, 3)$, $(2, 3)$ and $(2, 4)$ while significant control spillover is noticed near resonance of modes $(1, 2)^*$ and $(1, 4)^*$. The total attenuation in radiated sound power is 5.9 dB. In this case, all directions of minimization are in the $\phi = 0^\circ$ plane which coincides with radiation nodal line of modes $(1, 2)^*$, $(1, 4)^*$ and $(2, 4)^*$. In other words these modes are not well observed by the three error signals. This explains the increase in radiated sound power noticed near resonance of mode $(1, 2)^*$ and $(1, 4)^*$: rather than canceling the associated modal amplitudes, the controller recombines the modal amplitudes of the “double” modes and rotates the acceleration distribution such that the resulting nodal lines are aligned with the minimization angles. This modal restructuring mechanism is illustrated in Figure 9 where the out-of-plane velocity distribution measured by the laser vibrometer is shown before and after control at 541 Hz, i.e., near resonance of mode $(1, 4)$. The acceleration distribution after control is clearly rotated. Note that modal restructuring does not occur near resonance of mode $(1, 3)$ and $(2, 3)$. The first two control actuators are aligned with the anti-nodal lines of the associated “double” modes $(1, 3)^*$ and $(2, 3)^*$ therefore preventing their excitation. It should be pointed out that the previous case does not allow the above modal restructuring as the error signals do not correspond to pressure estimates along circumferential angles multiple of 45°, i.e., one error signal at least observes the “double” modes thereby forcing the controller to reduce the associated vibrations. These results therefore confirm the relatively good accuracy of the structural acoustic sensor implemented on the cylinder. Despite the small level of discretization ($Q_z = 2$, $Q_\phi = 6$) which only provides accurate estimates near resonance of mode $(1, 2)$, the sensor still yields error
information somewhat related to the radiated sound pressure at higher frequencies. This is further illustrated by comparing these results with the next control case based on far-field error microphones.

The third control case replaces the outputs of the structural acoustic sensor by three error microphones located in the far field. To facilitate the comparison of the two sensing approaches, this case uses the same control configuration as previously. In particular, the three error microphones are located along the same directions of pressure estimates implemented in the previous case. The radiated sound power estimated from the traverse microphone measurements is presented before and after control in Figure 10. As for the case of pressure estimates using Discrete Structural Acoustic Sensing (see Figure 8), excellent global sound attenuation is achieved near resonance of modes (1, 2), (1, 3) and (2, 3). Modal reduction occurs at these frequencies therefore canceling the sound pressure radiated over the entire field. Similarly, the sound power near resonance of mode (1, 2)* increases as in the previous control case based on the structural acoustic sensor. Again, this mode has a radiation nodal line along $\phi = 0^\circ$ and thus remains unobserved by any of the three error microphones. Note that the control spillover at this frequency is larger than in the previous control case. As a result, the total sound radiated power across the frequency bandwidth of excitation increases by 0.7 dB. Similar behavior can be observed near resonance of modes (1, 4)* and (2, 4)*. This case clearly illustrate the importance of the control actuator and error sensor configurations. In particular, global sound attenuation for lightly damped cylindrical radiators can only be achieved through modal reduction. As a final remark, it should be mentioned that the controlled response based on pressure estimates in the same three directions and the same two control actuators exhibits very similar trends as the ones obtained with error microphones. In particular, the structural acoustic sensor yields nearly the same response near resonance of modes (1, 2) and (1, 2)* which confirms the accuracy of the pressure estimates at these frequencies.
CONCLUSIONS

Broadband radiation control from a finite cylinder has been demonstrated both analytically and experimentally. The ASAC system implements a multi-channel Filtered-\(x\) LMS control algorithm. Structural control inputs are applied through piezoelectric actuators while error information is provided by a discrete structural acoustic sensor. Results show attenuation in total radiated sound power of up to 15 dB is achieved across a frequency bandwidth encompassing the first five flexural modes of the cylinder.

These results also validate the extension of Discrete Structural Acoustic Sensing to baffled cylindrical geometries. In this technique, the sensor constructs real time estimates of the pressure radiated in the far field at prescribed angles from discrete structural acceleration measurements and associated signal processing. Sensor accuracy and comparisons of control performances based on Discrete Structural Acoustic Sensing and the use of far-field error microphones, respectively, demonstrate the ability of the structural acoustic sensor to replace direct far-field pressure measurements. In particular, the analytical results show that minimizing a cost function based on sensor estimates yields similar control performance to that obtained with a cost function based on actual far-field pressure, provided accuracy is ensured for the sensor. The experimental results on broadband radiation control confirm this result. Similar control performances are obtained for both types of error information. Examination of the system's response at single frequencies within the bandwidth of excitation also reveals the controller behaves similarly in each case. This includes frequencies where modal restructuring is the main control mechanism which confirms the ability of the sensor to provide accurate radiation information.

The accuracy of the sensor estimate over a given frequency bandwidth is primarily related to the number of structural measurements implemented in the sensor. In particular, for equally spaced measurement points, accuracy of the sensor estimates is ensured provided the following two conditions are satisfied. First, the number of measurement points along the cylinder main axis should be such that the associated Nyquist wave-number (half the spatial sampling frequency) is located on the wave-number axis above the main content of the wave-number transform of the structural out-of-plane motion along this axis. In terms of simply-supported mode shapes, this first requirement is satisfied if the number of
measurement points along the main axis is greater than the highest order along the same axis of the modes which dominate the response. Second, the number of measurement points along the circumferential direction of the cylinder should be greater than twice the order of the highest circumferential order of the modes dominating the response. These conditions define the frequency bandwidth of accuracy for the structural acoustic sensor.

More generally, the design and implementation of the sensor do not require precise knowledge of the vibration characteristics of the structure. While some knowledge of the structural velocity distributions allows for predicting the accuracy of the pressure estimate and predetermining the appropriate discretization level, the sensor transfer functions strictly depend on the geometry of the structure and the surrounding fluid properties. This differs from a number of alternative structural sensing techniques whose design and implementation are directly based on dynamic properties of the structure, such as mode shapes. In comparison, Discrete Structural Acoustic Sensing provides a more robust error information. The modeling of the sensor transfer functions involves simple Finite Impulse Response (FIR) filters. This type of filter is well suited to represent the radiation of monopole and piston sources into an unbounded medium. In particular, the sensor transfer functions do not feature resonances, which would otherwise require the use of the more complex Infinite Impulse Response (IIR) filter model. Finally, when implemented in a far-field radiation control system, the time delay associated with the acoustic path can be removed from the sensor transfer functions without affecting the controller performance. This property greatly simplifies the sensor digital filters by reducing the number of coefficients required for accurate modeling of the sensor transfer functions. The sensor DSP implementation then becomes very efficient in terms of computations, allowing higher sampling frequencies or the combination of the sensor and controller code on a single DSP.

8 ACKNOWLEDGMENTS

The authors gratefully acknowledge the Office of Naval Research, Dr. Kam Ng, Technical Monitor. The authors also wish to thank Kent Schuster and Scott Kelly at Reynold’s Metal Company in the Extrusion Division for donating the aluminum cylinder used in the
experiments.
REFERENCES


List of Figures

1 Baffled cylindrical geometries .................................. 31
2 Schematic of the discrete structural acoustic sensor along with the three channel feedforward controller. ........................................ 32
3 Cylinder rig inside the anechoic chamber. ................................ 33
4 Actual and estimated far-field pressure in direction $(\theta, \phi) = (70^\circ, 240^\circ)$. ——, actual; ——, $Q_z = 8$, $Q_\phi = 7$; ——, $Q_z = 2$, $Q_\phi = 12$. ........................................ 34
5 Reconstructed and estimated far-field pressure in direction $(\theta, \phi) = (76.8^\circ, 0^\circ)$. ——, reconstructed pressure; ——, sensor output. ........................................ 35
6 Calculated radiated sound power - Minimization of pressure estimate in directions $(\theta, \phi) = (70^\circ, 0^\circ)$, $(110^\circ, 40^\circ)$, $(90^\circ, 160^\circ)$, $(70^\circ, 240^\circ)$, and $(110^\circ, 320^\circ)$. ——, before control; ——, after control, 3 by 9; ——, after control, microphones; ——, after control, acceleration. ........................................ 36
7 Measured radiated sound power - Minimization of pressure estimates in directions $(\theta, \phi) = (76.8^\circ, 340^\circ)$, $(63.5^\circ, 0^\circ)$, and $(103.2^\circ, 0^\circ)$. ——, before control; ——, after control (20 sample delay); ——, after control (no delay). ........................................ 37
8 Measured radiated sound power: Minimization of pressure estimates in directions $(\theta, \phi) = (70^\circ, 0^\circ)$, $(90^\circ, 0^\circ)$, and $(110^\circ, 0^\circ)$. ——, before control; ——, after control. ........................................ 38
9 Measured out-of-plane velocity distribution at 541 Hz: Minimization of pressure estimates in directions $(\theta, \phi) = (70^\circ, 0^\circ)$, $(90^\circ, 0^\circ)$, and $(110^\circ, 0^\circ)$. ........................................ 39
10 Measured radiated sound power - Minimization of pressure at error microphones in directions $(\theta, \phi) = (70^\circ, 0^\circ)$, $(90^\circ, 0^\circ)$, and $(110^\circ, 0^\circ)$. ——, before control; ——, after control. ........................................ 40
Figure 1: Baffled cylindrical geometries.
Figure 2: Schematic of the discrete structural acoustic sensor along with the three channel feedforward controller.
Figure 3: Cylinder rig inside the anechoic chamber.
Figure 4: Actual and estimated far-field pressure in direction $(\theta, \phi) = (70^\circ, 240^\circ)$. ——, actual; - - - - , $Q_x = 8$, $Q_\phi = 7$; · · · · , $Q_x = 2$, $Q_\phi = 12$. 
Figure 5: Reconstructed and estimated far-field pressure in direction $(\theta, \phi) = (76.8^\circ, 0^\circ)$. 
—, reconstructed pressure; ·····, sensor output.
Figure 6: Calculated radiated sound power - Minimization of pressure estimate in directions \((\theta, \phi) = (70^\circ, 0^\circ), (110^\circ, 40^\circ), (90^\circ, 160^\circ), (70^\circ, 240^\circ),\) and \((110^\circ, 320^\circ)\). ——, before control; ——, after control, 3 by 9; ·····, after control, microphones; ——, after control, acceleration.
Figure 7: Measured radiated sound power - Minimization of pressure estimates in directions $(\theta, \phi) = (76.8^\circ, 340^\circ), (63.5^\circ, 0^\circ), \text{ and } (103.2^\circ, 0^\circ)$. ——, before control; - - -, after control (20 sample delay); - - - - - - - - - - , after control (no delay).
Figure 8: Measured radiated sound power: Minimization of pressure estimates in directions $(\theta, \phi) = (70^\circ, 0^\circ), (90^\circ, 0^\circ), \text{ and } (110^\circ, 0^\circ)$. —, before control; ——, after control.
Figure 9: Measured out-of-plane velocity distribution at 541 Hz: Minimization of pressure estimates in directions \((\theta, \phi) = (70^\circ, 0^\circ), (90^\circ, 0^\circ),\) and \((110^\circ, 0^\circ).\)
Figure 10: Measured radiated sound power - Minimization of pressure at error microphones in directions \((\theta, \phi) = (70^\circ, 0^\circ), (90^\circ, 0^\circ), \) and \((110^\circ, 0^\circ)\). ——, before control; ·····, after control.
List of Tables

1 Dimensions and material properties of the cylinder and piezoelectric actuators. 42
2 Actuators center location. 43
3 Comparison of the numerical and experimental natural frequencies. 44
4 Calculated total attenuation levels over the frequency bandwidth of excitation. 45
<table>
<thead>
<tr>
<th>Parameter</th>
<th>Cylinder</th>
<th>Actuators</th>
</tr>
</thead>
<tbody>
<tr>
<td>length (mm)</td>
<td>987</td>
<td>38.1</td>
</tr>
<tr>
<td>outside diameter / width (mm)</td>
<td>254</td>
<td>69.9</td>
</tr>
<tr>
<td>thickness, (mm)</td>
<td>2.36</td>
<td>0.1905</td>
</tr>
<tr>
<td>Young's modulus, (N/m²)</td>
<td>$7.1 \times 10^7$</td>
<td>$6.1 \times 10^{10}$</td>
</tr>
<tr>
<td>Poisson ratio,</td>
<td>0.31</td>
<td>0.33</td>
</tr>
<tr>
<td>mass density, (kg/m³)</td>
<td>2700</td>
<td>7750</td>
</tr>
<tr>
<td>$d_{31}$ constant (m/V)</td>
<td>-</td>
<td>$171 \times 10^{-12}$</td>
</tr>
<tr>
<td>hysteretic damping factor,</td>
<td>0.002</td>
<td>0</td>
</tr>
</tbody>
</table>

Table 1: Dimensions and material properties of the cylinder and piezoelectric actuators.
<table>
<thead>
<tr>
<th>Actuator</th>
<th>$z/L$</th>
<th>$\phi$ (degrees)</th>
</tr>
</thead>
<tbody>
<tr>
<td>disturbance PZT</td>
<td>-0.328</td>
<td>0</td>
</tr>
<tr>
<td>control PZT # 1</td>
<td>0.370</td>
<td>180</td>
</tr>
<tr>
<td>control PZT # 2</td>
<td>0.220</td>
<td>60</td>
</tr>
<tr>
<td>control PZT # 2</td>
<td>-0.395</td>
<td>250</td>
</tr>
</tbody>
</table>

Table 2: Actuators center location.
<table>
<thead>
<tr>
<th>mode \ ($(m, n)$)</th>
<th>modeled (Hz)</th>
<th>measured (Hz)</th>
</tr>
</thead>
<tbody>
<tr>
<td>(1, 2)</td>
<td>240.8</td>
<td>241.2</td>
</tr>
<tr>
<td>(1, 2)*</td>
<td>241.0</td>
<td>244.2</td>
</tr>
<tr>
<td>(1, 3)</td>
<td>304.9</td>
<td>302.9</td>
</tr>
<tr>
<td>(1, 3)*</td>
<td>305.5</td>
<td>303.1</td>
</tr>
<tr>
<td>(2, 3)</td>
<td>498.8</td>
<td>497.0</td>
</tr>
<tr>
<td>(2, 3)*</td>
<td>499.2</td>
<td>500.3</td>
</tr>
<tr>
<td>(1, 4)</td>
<td>547.6</td>
<td>540.6</td>
</tr>
<tr>
<td>(1, 4)*</td>
<td>547.8</td>
<td>541.1</td>
</tr>
<tr>
<td>(1, 1)</td>
<td>565.3</td>
<td>708.0</td>
</tr>
<tr>
<td>(1, 1)*</td>
<td>565.5</td>
<td>–</td>
</tr>
<tr>
<td>(2, 4)</td>
<td>609.1</td>
<td>601.7</td>
</tr>
<tr>
<td>(2, 4)*</td>
<td>609.2</td>
<td>604.6</td>
</tr>
</tbody>
</table>

Table 3: Comparison of the numerical and experimental natural frequencies.
### Table 4: Calculated total attenuation levels over the frequency bandwidth of excitation.

<table>
<thead>
<tr>
<th>Control case</th>
<th>Attenuation (dB)</th>
<th></th>
<th></th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>Mean-square velocity</td>
<td>Sound radiated power</td>
<td></td>
</tr>
<tr>
<td>3 by 9 structural acoustic sensor error microphones</td>
<td>8.1</td>
<td>20.2</td>
<td></td>
</tr>
<tr>
<td>3 by 9 accelerometer array</td>
<td>8.3</td>
<td>19.4</td>
<td></td>
</tr>
<tr>
<td></td>
<td>20.7</td>
<td>12.4</td>
<td></td>
</tr>
</tbody>
</table>
APPENDIX 20
IMPLEMENTATION OF FAST RECURSIVE ESTIMATION TECHNIQUES FOR ACTIVE CONTROL OF STRUCTURAL SOUND RADIATION

M. J. Bronzel, C. R. Fuller

Vibration and Acoustics Laboratories, Virginia Polytechnic Institute and State University
Blacksburg, VA 24061-0238, USA

INTRODUCTION

Recently, much research has been conducted in the area of active control of sound radiated from a vibrating structure. The minimization of the radiated sound can be accomplished using different approaches. Active Structural Acoustic Control (ASAC) changes the radiation characteristics of the vibrating structure by applying forces or moments directly to the structure. In contrast, Active Noise Control (ANC) involves additional noise sources usually by means of loudspeakers to superimpose a secondary sound field which attenuates the primary noise. The effectiveness of ASAC techniques has successfully been reported by Fuller using point actuators [1]. More recently, Smith [2] has demonstrated the potential attenuation of broadband sound radiation from a simply supported plate by means of piezoelectric actuators directly mounted on the vibrating structure. ASAC effectively reduces the control effort needed to minimize the sound radiation by forcing the structure to vibrate in non-volumetric modes which are inefficient radiators [3]. Structural time domain wavenumber sensing has successfully been demonstrated by Maillard and Fuller [4, 5]. They have developed a technique to estimate the sound radiated in the farfield using time domain structural wavenumber sensors based on out-of-plane acceleration signals measured directly on the structure. This eliminates the need of microphones in the farfield to obtain error signals needed to optimize the impulse response of a feedforward adaptive controller.

The implementation of an Active Noise Control System for controlling the sound radiation from a vibrating structure requires the selection of suitable reference signals. Causality and coherence considerations have to be taken into account with respect to the reference signal filtered through a representation of the control and error path. This filtering is inevitable to feedforward control implementations based on adaptive filtering. Sensing of the reference signals from the vibrating structure is inevitable for real applications since in most cases the original disturbance signal is either unknown or cannot be measured directly. This will effectively result in a structural filtering of the unknown noise source and limit the bandwidth of the measured reference signals. Most feedforward control algorithms used for adapting the coefficients of an FIR filter rely on certain assumptions concerning the statistical properties of the provided reference signals. However, these do not hold for real applications and results in non-optimal convergence properties of the implemented adaptation algorithm.
The deficiencies of stochastic gradient type algorithms for non-white input data have been overcome by exact recursive least squares type algorithms which converge independently of the statistical properties of the input or reference signal. In active noise control applications the secondary path transfer functions have to be taken into account within the update loop of the adaptive filter. The commonly used Filtered-X approach applies only to algorithms which convergence in a stochastic sense. This paper describes the modifications needed to make exact recursive least squares algorithms feasible for active noise control applications. Experiments for controlling the sound radiated from a vibrating plate have been carried out using a stabilized version of the Fast A-priori Error Sequential Technique (SFAEST) algorithm. This paper discusses design issues for development and implementation of an active noise control system based on fast feedforward control algorithms. It presents a method which effectively reduces the limitations imposed by bandlimited reference signals exhibiting a large eigenvalue spread of the associated covariance matrix. This paper will demonstrate the superior control authority of fast transversal filter algorithms in the context of an overall ASAC design approach. Experiments for control of sound radiation from a vibrating plate are carried out and results demonstrate the effectiveness of the proposed design method.

FAST RECURSIVE ESTIMATION

During the past decade fast adaptive algorithms have been developed and further refined in order to reduce their computational complexity and insure their stability under noisy or time-varying constraints. But only recently these algorithms have been applied to noise control problems. Alexander [6], Bronzel [7] and Carayannis [8] provide an overview of fast recursive estimation techniques. Recursive identification is based on optimizing a set of model parameters given the sampled signals rather than relying on assumptions concerning the statistical properties of the signals involved. The Recursive Least Squares (RLS) algorithm implements a sequential iterative procedure for inverting the sampled autocorrelation matrix based on the Sherman-Morrison Lemma [9] resulting in a computational complexity of $O(N^2)$ operations. Fast algorithms have been developed which further utilize the shift-invariance properties of the autocorrelation matrix. They usually circumvent the direct inversion of the covariance matrix by means of feedforward and feedbackward transversal filters to calculate the Kalman gain for updating the unknown parameters of the adaptive filter. These fast variants of RLS algorithms have a computational complexity down to $O(7N)$ for the Fast A-posteriori Error Sequential Technique (FAEST) algorithm [10]. Hence these algorithms are becoming feasible for real-time applications. In our ASAC application we have implemented a numerically stabilized version (SFAEST) of this algorithm which has been developed by Moustakides [11]. The iterative adaptation loop of the SFAEST algorithm is summarized in the Appendix.

In order to apply adaptive filters in feedforward control applications, it is necessary to estimate the contribution of the filter output to the error signal. A common approach is to obtain a representation $S$ of the control and error path and change the order of filter operations. This results in the standard Filtered-X algorithm, depicted schematically in Figure 1. Changing the order of filter operations is possible only for Linear Time-Invariant (LTI) systems. Adaptive filters are time-varying by definition and the corresponding filter operators do generally not commute. However, the errors resulting from a change of the filter operations are small in the limit of slow adaptation. This assumption is valid for stochastic gradient algorithms such as the LMS algorithm which is commonly used in feedforward adaptive control applications.
Figure 1: Filtered-X configuration for adaptive feedforward controllers based on stochastic gradient adaptation schemes.

The failure of the Filtered-X algorithm, when applied to fast recursive estimation schemes has been reported by Bronzel [7], who successfully introduced the Extended-Filtered-X algorithm which enables the implementation of fast recursive identification techniques for feedforward control applications. The basic scheme of this approach is shown in more details in Figure 2. The Extended-Filtered-X algorithm calculates an adjustment $\Delta e[k]$ which compensates the error introduced by exchanging the non-commuting filter operators in the standard Filtered-X algorithm. The output $y[k]$ of the adaptive filter $A$ is filtered through a representation $S$ of the control and error path, yielding an estimate $c_e[k]$ of the contribution from the control signal to the error signal $e[k]$. Exchanging the order of the corresponding filters yields another estimate $c_s[k]$. The necessary adjustment to the sampled error signal $e[k]$ is given by the difference of these estimates:

$$\Delta e[k] = \sum_i a_i[k] r[k-i] - \sum_j s_j y[k-j].$$

Schirmacher [12] has reported the active control of airborne sound using the SFAEST algorithm with a slightly modified form of the Extended-Filtered-X configuration which has been developed by Bjarnason [13], who applied his Modified-Filtered-X scheme to the LMS algorithm. Although the Extended-Filtered-X technique could be applied to stochastic gradient algorithms it will generally yield no performance improvements and is furthermore not essential to ensure stability.

Figure 2: Extended-Filtered-X configuration for adaptive feedforward controllers based on fast recursive estimation techniques.
EXPERIMENTAL SET-UP

The experiments were performed on a simply supported plate made of plain carbon steel with dimensions 380 x 298 x 1.93 mm. Figure 3 shows a schematic display of the experimental arrangement. The simply supported boundary conditions were provided by thin flexible metal shims connecting the edges of the steel plate with a heavy support stand. The lower order resonance frequencies of this plate are presented Table 1.

![Schematic of experimental setup](image)

Nine accelerometers were mounted on the plate as depicted in Figure 3 to measure the structural vibration. The acceleration signals were sampled at 1 kHz and passed through a digital filter network to provide time domain estimates of 3 wave-number components. One of these is used as an error signal for the adaptive feedforward controller. The calculations are carried out in real-time on a TMS320C30 DSP board. A more detailed description of the wave-number sensor is given by Maillard and Fuller [5].

Table 1: Plate modal resonant frequencies.

<table>
<thead>
<tr>
<th>Mode (m,n)</th>
<th>(1,1)</th>
<th>(2,1)</th>
<th>(1,2)</th>
<th>(2,2)</th>
<th>(3,1)</th>
<th>(3,2)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Frequency [Hz]</td>
<td>87</td>
<td>183</td>
<td>244</td>
<td>330</td>
<td>343</td>
<td>474</td>
</tr>
</tbody>
</table>

Two G1195 PZT piezoelectric actuators with dimensions 38 x 32 x 0.19 mm were mounted on each side of the plate and wired out of phase to provide bending moments as control inputs. The controller and a digital random noise signal generator were implemented on a separate TMS320C30 DSP board with 2 channel built-in analog I/O. The signal taken from the output of the built-in signal generator was filtered through a low-pass Butterworth filter with a cut-off frequency of 400 Hz to excite the disturbance shaker. While the reference signal for the adaptive feedforward controller is directly available from the random noise sequence, an adjustable delay was introduced in the disturbance path to investigate the effects of causality. The I/O data stream is sampled at 1 kHz. The adaptive controller uses 180 FIR filter coefficients which are updated.
using the Extended-Filtered-X SFAEST algorithm as outlined in the Appendix. The control and error path is represented using a transversal filter with 180 coefficients. The windowing parameter $\lambda$ in the SFAEST algorithm is set to $1.0$ which resembles the stationarity of the disturbance signal and of the plate dynamics. Suitable values for the initialization parameter $\mu = 1.0$ and for the stability parameter $\rho = 1.0$ were established during experimental testing. The coherence between the disturbance signal and the error signal as well as the auto-spectrum of the estimated wavenumber components were monitored using a portable B&K 2148 dual channel signal analyzer.

**EXPERIMENTAL RESULTS**

The estimate of the sound pressure based on structural time domain wavenumber sensing provides a signal related to the noise radiation from the vibrating simply supported plate coupling in the directions corresponding to $\theta = -30^\circ$, $\theta = 0^\circ$ and $\theta = 30^\circ$ (see Figure 3). The implemented controller is currently designed for SISO systems. Therefore it is only possible to minimize the sound radiation in one direction using a single PZT actuator mounted on the vibrating plate. However, the estimated sound pressure corresponding to the other two directions was monitored, as well.

The following results present an optimal controller, where a 20 ms delay has been added to the disturbance path to make the system more causal. This enables the controller to adapt the coefficients necessary to compensate the non-minimum phase zeros of the control and error path. Figure 4 shows the auto-spectrum of the estimated error signal at $\theta = -30^\circ$ for the uncontrolled (solid line) and controlled (dashed line) system after 5000 iterations using the Extended-Filtered-X SFAEST based controller. Significant reduction of the radiated sound is achieved with peak attenuation levels up to 28 dB for the entire bandwidth ranging from 50 Hz to 400 Hz. The total sound pressure level attenuation is 7.9 dB across that frequency range. Figure 5 displays the noise reduction being achieved. Large reductions in radiated sound level are obtained for on-resonance frequencies. Figure 6 and Figure 7 show the estimated sound pressure level before (solid line) and after (dashed line) control for the additional signals corresponding to pressure radiation in $\theta = 30^\circ$ and $\theta = 0^\circ$. Note that these signals were not included in the costfunction to be minimized using the adaptive feedforward controller. Cancellation occurs at resonant frequencies while the noise level off-resonance is increased in general.

---

**Figure 4:** Auto-spectrum of the estimated error signal ($\theta = -30^\circ$, $\phi = 0^\circ$) before and after control.

**Figure 5:** Noise attenuation of the estimated error signal ($\theta = -30^\circ$, $\phi = 0^\circ$) after 5000 iterations.
The previously presented results were obtained after 5000 iterations using the Extended-Filtered-X SFAEST algorithm. In order to demonstrate its effectiveness, the experiments were repeated using the common Filtered-X LMS algorithm. A convergence factor of $\mu = 2.0 \times 10^{-10}$ was chosen as the maximum possible value to maintain stability during the process of adaptation. Figure 8 and Figure 9 show respectively the auto-spectrum of the error signal before (solid line) and after (dashed line) control and the achievable noise attenuation after 20000 iterations. Although the radiated pressure associated with most dominant radiating modes has been reduced, the overall broadband performance of the Filtered-X LMS based controller has still not achieved that obtained using the Extended-Filtered-X SFAEST algorithm. The total sound pressure attenuation results in 4.0 dB in the frequency range between 50 and 400 Hz.

A direct comparison of the learning curves for the SFAEST and the LMS based controllers demonstrates the superior performance of the fast recursive least squares controller. The SFAEST algorithm converges very fast although the covariance matrix of the filtered reference signals exhibits an extremely high eigenvalue ratio of $\eta = 4.42 \times 10^5$. The residual error signals for the first 5000 iterations are shown in Figure 10 for the two different controllers. Note that the output of the SFAEST based controller was switched off during the first 1000 iterations. This results in better control performance since the SFAEST and other fast adaptation schemes tend to overshoot the residual error during the first iterations in a noisy environment.
Figure 10: Residual adaptation error during the initial 5000 iterations for the SFAEST. The results are obtained as a smoothed ensemble average from 10 experiments.

Figure 11: Coherence function $\gamma_{xy}^2$ before control and after 1000 and 5000 iterations using the SFAEST based controller.

Figure 12: Coherence function $\gamma_{xy}^2$ before control and after 1000 and 5000 iterations using the LMS based controller.

Figure 13: Control performance of the SFAEST based controller after 1000 and after 5000 iterations.

Figure 14: Control performance of the LMS based controller after 1000 and after 5000 iterations.
Figure 11 compares the coherence function between the reference signal and the error signal before and after 1000 and 5000 iterations for the SFAEST based controller. The significant decrease of the coherence indicates that most of the correlated content from the reference signal has already been removed from the error signal after 1000 iterations. The corresponding coherence functions for the LMS based controller are shown in Figure 12. The coherence has not remarkably decreased after 5000 iterations. Figure 13 and Figure 14 show respectively the effective noise attenuation for the SFAEST and for the LMS based controller before control and after 1000 and 5000 iterations. The attenuation of the radiated sound pressure corresponds to the decrease of the coherence between the reference and the error signal shown before.

In Figure 15, the previously obtained results (dotted line) are compared with that achieved with a controller implementation where no additional delay has been added in the disturbance path (dashed line). The corresponding optimal controller coefficients after 5000 iterations are shown in Figure 16. As expected, better attenuation over the frequency band is obtained with the delay being added, due to the non-minimum phase characteristics of the control and error path. The controller coefficients with smaller index than the number of additional delay taps in the disturbance path cannot be modeled in realistic applications. This results in some control spillover for off-resonance frequencies which can be reduced using more than only one control actuator. However, significant attenuation was achieved at resonant frequencies using the control implementation without delay. The broadband overall sound pressure reduction is 4.2 dB across the entire bandwidth.

CONCLUSIONS

Active structural acoustic control of structure-born sound has been investigated using a fast RLS algorithm. Application of fast adaptation schemes has been made feasible using an Extended-Filtered-X algorithm which eliminates the inevitable errors associated with the standard Filtered-X approach. The presented results clearly demonstrate the superior control authority of the Extended-Filtered-X SFAEST algorithm over the Filtered-X LMS algorithm if the covariance matrix of the underlying filtered reference signals exhibits a large eigenvalue ratio. Broadband attenuation of the noise radiated from a simply supported plate in one direction has been demonstrated with peak attenuation levels up to 28 dB in the frequency range from 50 Hz to 400
Hz. Most of the reduction in the radiated sound pressure has been achieved after 1000 iterations. However, a significant number of FIR coefficients were necessary to provide attenuation of the levels achieved. Furthermore, results indicate, that a significant performance improvement could be expected from implementing fast IIR type adaptive filters based on recursive least squares estimation techniques.

REFERENCES


ACKNOWLEDGMENTS

This work has been funded under ONR N00014-92-J-470. The authors are grateful to Rolf Schirmacher and Thomas Auspitzer from University of Göttingen for providing a C-code version of the SFAEST algorithm and to Julien Maillard for providing the time domain wavenumber sensor.
APPENDIX

Stabilized Fast A-posteriori Error Sequential Technique (SFAEST) algorithm. The following list summarizes the update equations for the Extended-Filtered-X SFAEST algorithm. The computational complexity of this algorithm counts 8n+41+2n+2m+1 Multiply-And-Add instructions including 5 divisions.

\[ r[k+1] = s_T \, x_n[k+1] \]
\[ e_a'[k+1] = r[k+1] - f_T \, r_n[k] \]
\[ e_b'[k+1] = r[k-n] - b_T \, r_n[k+1] \]
\[ \gamma_{a+1}[k+1] = \frac{\lambda \, \alpha_a'[k]}{\lambda \, \alpha_a'[k] + e_a'[k+1] \, \gamma_a[k]} \]
\[ y_n[k+1] = \frac{\gamma_{a+1}[k+1]}{1 + e_a'[k+1] \, \left( \kappa_n[k] + \frac{c_n[k]}{\lambda \, \alpha_a'[k]} \, f_n[k] \right) \, \gamma_{a+1}[k+1]} \]
\[ c_n[k] = \lambda^{-n} \, y_n[k] \, f_n'[k] \]
\[ \xi_n[k+1] = e_a'[k+1] + \lambda \, \alpha_a'[k] \, \kappa_n[k] + c_n[k] \, e_a'[k+1] \]
\[ \tilde{\xi}_n[k+1] = \frac{\xi_n[k+1]}{1 + \rho \, \left( 1 - \gamma_n[k+1] \right) + \rho \, c_n[k] \, \left( 1 - \gamma_n[k] \right)} \]
\[ \alpha_a'[k+1] = \lambda \, \alpha_a'[k] + \gamma_n[k] \, \xi_n'[k+1] \]
\[ \xi_n'[k+1] = e_a'[k+1] - \left( 1 - \gamma_n[k+1] \right) \rho \, \xi_n[k+1] \]
\[ \alpha_a'[k+1] = \lambda \, \alpha_a'[k] + \gamma_n[k] \, \alpha_a'[k+1] \]
\[ \kappa_{a+1}[k+1] = \begin{pmatrix} 0 \\ \kappa_n[k] \end{pmatrix} - \frac{1}{\lambda \, \left[ \kappa_n[k] - f_n[k] \right]} \, \begin{pmatrix} 1 \\ \alpha_a'[k] \end{pmatrix} \]
\[ f_n[k+1] = f_n[k] - \gamma_n[k] \left( e_a'[k+1] + \rho \, \xi_n'[k+1] \, c_n[k] \right) \, \kappa_n[k] \]
\[ \begin{pmatrix} \kappa_n[k+1] \\ 0 \end{pmatrix} = \begin{pmatrix} \kappa_{a+1}[k+1] - \kappa_{a+1}[k] \, \left[ \begin{pmatrix} -b_n[k] \\ 1 \end{pmatrix} \right] \\ \kappa_{a+1}[k] \end{pmatrix} \]
\[ b_n[k+1] = b_n[k] - \gamma_n[k+1] \left( e_a'[k+1] + \rho \, \xi_n'[k+1] \right) \, \kappa_n[k+1] \]
\[ y[k+1] = a_n^T[k] \, x_n[k+1] \]
\[ \Delta e[k+1] = a_n^T[k] \, r_n[k+1] - s_T \, y_n[k+1] \]
\[ e_a[k+1] = a_n^T[k] \left( e_a[k+1] + \Delta e[k+1] \right) \]
\[ e_a'[k+1] = a_n^T[k] - e_a'[k+1] \, \kappa_n[k+1] \]

filtered reference signal
a priori forward predictor error
a priori backward predictor error
order recursion for angle update (step up)
angle update (step down)
windowed angle parameter
stabilization factor
stabilized forward predictor error
forward angle update
stabilized backward predictor error
backward angle update
alternative Kalman gain update
forward predictor update
partitioning of alternative Kalman gain
backward predictor update
FIR filter output
a priori error adjustment
modified a posteriori error
FIR filter update
Multiple reference feedforward active noise control

Part I  Analysis and simulation of behavior

Yifeng Tu

Chris R. Fuller

Vibration and Acoustic Laboratories
Mechanical Engineering Department
Virginia Polytechnic Institute and State University
Blacksburg, Virginia 24061-0238

Running headline:  Multiple reference feedforward active noise control
Address for correspondence:

Prof. Chris R. Fuller
Mechanical Engineering Department
Virginia Polytechnic Institute and State University
Blacksburg, Virginia 24061-0238

Number of Pages: 30
Number of figures: 12
1. Summary

In this work, two forms of frequency domain optimum solutions for multiple reference active noise control (MRANC) in a feedforward arrangement are given in terms of signal spectra and system frequency response functions (FRF). It is particularly noted that the optimum solutions are uncoupled if the reference signals are uncorrelated or the noise sources are directly available as reference signals. Other than the traditional control configuration which feed each reference signal into a different filter, another control configuration which combines multiple reference signals into a single input is also investigated. Although this configuration generally delivers a compromised control effect, the payoff is a simplified control structure and significant computational savings. In addition, under some circumstances, the desired performance which is comparable to other complex configurations can be achieved. Simulations based on sound transmission through a vibrating plate have been conducted and the results presented are consistent with the theoretical analysis.

PACS numbers: 43.40 Vn, 43.50 Ki
2. Introduction

Over the last two decades, active noise control (ANC) has been demonstrated as an effective approach for low frequency noise reduction. The principle of ANC is based on the superposition of the two acoustical waves from the primary and secondary sources. When the two waves are out of phase and of the same amplitude, the superposition results in complete cancellation of the two waves and therefore generates a silent zone. ANC provides an ideal complement to the conventional passive noise control approach, which generally works efficiently at higher frequencies.

The application of ANC has been extended from single noise source one-dimensional acoustical fields, e.g. a low frequency field in an air duct, to complex multiple noise sources three-dimensional acoustical fields, e.g. a sound field in an aircraft cabin. The extension to three-dimensional acoustical fields requires a number of secondary sources to minimize the mean square signals from a number of error sensors, so that the primary acoustical field can be spatially matched [1]. The extension to a multiple noise source environment usually requires more than one reference sensor to generate a complete set of reference signals [2].

An important issue for multiple reference active noise control (MRANC) is the selection of the number and the positions of reference sensors. The principle for reference sensor selection is based on using a minimum number of reference sensors to
achieve maximum multiple coherence between the reference signals and the primary noise signals. The number of reference sensors should be as small as possible in order to simplify control structure and save computational cost, while the multiple coherence function should be as close to unity as possible in order to achieve maximum noise reduction [2]. Generally speaking, the reference sensors should be able to detect all the independent noise sources, which usually implies more reference sensors than independent noise sources [3]. A number of frequency domain approaches have been applied to identify noise sources based on coherence techniques [4][5] and Principal Component Analysis [6]. The applicability of coherence techniques is determined by the correlation among reference signals. Principle Component Analysis is effective in determining the dominant noise sources and does not depend on the correlation among reference signals. However, the exact locations of reference sensors required to detect those noise sources remain unknown. Recently, attention has been paid on the convergence rate as well as the number of controller coefficients for reference sensor selection [7].

The previous studies of multiple reference active noise control are mainly focused on noise source identification and reference sensor selection. An important issue remains untouched. When MRANC is designed to achieve noise reduction at a number of locations or globally, the control structure becomes very complicated and the amount of computations may exceed the computational limit of many digital signal processors (DSP). It is therefore of much significance to study the possibility of simplifying the
MRANC control structure, while maintaining the desired noise reduction. In addition, the coupling effect between noise sources and reference sensors has not directly investigated.

3. Frequency domain analysis

Figure 1 shows a typical MRANC system, in which there are M noise sources, K reference sensors, one secondary source and one error sensor. Each reference signal is fed into a different filter and the output of each filter is summed together to drive a single secondary source. This multiple reference multiple input (MRMI) configuration for processing reference signals attempts to control the primary noise from multiple primary paths with the secondary noise from multiple filters, thus it enjoys the applicability in general situations and has been adopted in most previous studies [2], [8]. Since the number of the secondary sources and error sensors is chosen to be unity to simplify the analysis, spatial noise reduction effect is not a concern. The signal at the error sensor is the superposition of the primary noise and the secondary noise and is given by

\[ e(w) = d(w) + T(w)x^T(w)W(w) \]  

where \( d(w) \) is the primary noise, \( T(w) \) is the FRF between the secondary source and the error sensor, \( x(w) \) is the reference signal vector, and \( W(w) \) is the controller FRF vector, that is

\[ x(w) = \begin{bmatrix} x_1(w) & x_2(w) & \cdots & x_k(w) \end{bmatrix}^T \]  
\[ W(w) = \begin{bmatrix} W_1(w) & W_2(w) & \cdots & W_k(w) \end{bmatrix}^T \]
The objective of ANC is to minimize the error signal, thus the cost function is defined as

$$\xi(w) = \mathbb{E}(e^*(w)e(w)) \quad (4)$$

Substituting equation (1) into equation (4) and taking the complex derivative with respect to the controller vector \( \mathbf{w} \), the gradient of the cost function is obtained as

$$\nabla = \mathbb{E}(\bar{x}(w)x^T(w))W(w) + \mathbb{E}(d(w)x^T(w)) \quad (5)$$

where

$$\bar{x}(w) = x(w)x^T(w) \quad (6)$$

The optimum controller vector can be obtained by setting the gradient to zero, thus

$$\mathbf{W}_{opt}(w) = -R^{-1}(w)P(w) \quad (7)$$

where \( R(w) \) is a matrix, whose diagonal terms are the auto-spectra of the reference signals and off-diagonal terms are the cross-spectra of the reference signals, \( P(w) \) is a vector representing the cross-spectra between the reference signals and the primary noise, i.e.

$$R(w) = \begin{pmatrix}
S_{aa}(w) & S_{a\bar{a}}(w) & A & S_{\bar{a}a}(w) \\
S_{a\bar{a}}(w) & S_{\bar{a}a}(w) & A & S_{\bar{a}\bar{a}}(w) \\
M & M & M & M \\
S_{\bar{a}a}(w) & S_{\bar{a}\bar{a}}(w) & A & S_{\bar{a}\bar{a}}(w)
\end{pmatrix} \quad (8)$$

$$P(w) = \begin{pmatrix}
S_{a\bar{a}}(w) \\
S_{\bar{a}a}(w) \\
M \\
S_{\bar{a}\bar{a}}(w)
\end{pmatrix} \quad (9)$$

It is interesting to note that if the reference signals are uncorrelated, their cross-spectra would be zero. It follows that all the off-diagonal terms inside the matrix \( R(w) \) are zero. Thus, equation (7) can be rewritten as
The above equation indicates that the optimum solutions of all the filters are independent, and each filter operates without any interference from other filters. In other words, the filters are uncoupled if the reference signals are uncorrelated. The optimum controller vector can be derived in terms of system FRF's. The primary noise at the error sensors is formed by M noise sources passing through M primary paths and is given by

\[ d(w) = n^T(w)P(w) \]  

where \( n(w) \) and \( P(w) \) are noise source vector and primary FRF vector respectively and can be expressed as

\[ n(w) = [n_1(w) \; n_2(w) \; \ldots \; n_M(w)]^T \]  

\[ P(w) = [P_1(w) \; P_2(w) \; \ldots \; P_M(w)]^T \]  

The noise sources and the reference signals are related by

\[ x(w) = H(w)n(w) \]  

where \( H(w) \) is the source coupling matrix, whose element \( H_{ij}(w) \) represents the FRF between the \( i \)th noise source and the \( j \)th reference sensor, and can be written as

\[ H(w) = \begin{bmatrix} H_{11}(w) & H_{21}(w) & \ldots & S_{m1}(w) \\ H_{21}(w) & H_{22}(w) & \ldots & S_{m2}(w) \\ \vdots & \vdots & \ddots & \vdots \\ H_{m1}(w) & H_{m2}(w) & \ldots & S_{mm}(w) \end{bmatrix} \]  

Substituting equations (6), (11), and (14) into equation (5) and moving all the FRF's to the outside of expectation, the gradient of the cost function is obtained as
\[ V = T^*(w)H^*(w)E(n^*(w)n^T(w)) [H^T(w)T(w)W(w) + P(w)] \]  

(16)

Another form of optimum controller vector can be obtained by setting the gradient to zero, i.e.

\[ W_{opt}(w) = -H^*(w)\frac{P(w)}{T(w)} \]  

(17)

where \(H^*(w)\) is the pseudo-inverse [10] of \(H^T(w)\). The optimum controller vector is expressed in terms of FRF’s of the primary path, the error path and the source coupling path. This solution is intuitively clear: the error signal comes from the noise sources through both the primary path and the secondary path, the controller in the secondary path adjusts its FRF so that the two paths have the same magnitude response and 180 degrees phase difference. Thus, noise cancellation is achieved.

If \(H(w)\) is an identity matrix, which implies that the number of reference sensors is the same as the number of noise sources, and each noise source is available as reference signal, equation (17) reduces to

\[ W_{opt}(w) = -\frac{P(w)}{T(w)} \]  

(18)

Again, equation (18) indicates that the optimum solutions of all the control filters are independent. In fact, each optimum solution is determined only by the corresponding primary path and the error path.

An alternative MRANC system combines all the reference signals into a single input (MRSI) as shown in Figure 2. This configuration generally delivers compromised noise reduction effect. However, under some special circumstances, desired noise
reduction effect along with a large number of computational savings can be obtained.

The coherence function is defined as

$$\gamma(w) = \frac{S_{ad}(w)S_{dc}(w)}{S_{ax}(w)S_{dd}(w)}$$  \hspace{1cm} (19)$$

The primary noise and the reference signal can be written as

$$d(w) = n_1(w)P_1(w) + n_2(w)P_2(w) + \Lambda + n_M(w)P_M(w)$$ \hspace{1cm} (20)$$

$$x(w) = n_1(w)H_1(w) + n_2(w)H_2(w) + \Lambda + n_M(w)H_M(w)$$ \hspace{1cm} (21)$$

where $H_j(w)$ is the summation of all the FRF's from the $i$th noise sources to all the reference sensor and is given by

$$H_i(w) = \sum_{j=1}^{K} H_{ij}(w)$$ \hspace{1cm} (22)$$

Since the noise sources are assumed to be uncorrelated, their cross-spectra are zero, it follows that

$$S_{xx} = S_{n_1n_1}|H_1|^2 + S_{n_2n_2}|H_2|^2 + \Lambda + S_{n_Mn_M}|H_M|^2$$ \hspace{1cm} (23)$$

$$S_{dd} = S_{n_1n_1}|P_1|^2 + S_{n_2n_2}|P_2|^2 + \Lambda + S_{n_Mn_M}|P_M|^2$$ \hspace{1cm} (24)$$

$$S_{ad} = S_{n_1n_1}H_1^*P_1 + S_{n_2n_2}H_2^*P_2 + \Lambda + S_{n_Mn_M}H_M^*P_M$$ \hspace{1cm} (25)$$

$$S_{dx} = S_{n_1n_1}H_1P_1^* + S_{n_2n_2}H_2P_2^* + \Lambda + S_{n_Mn_M}H_MP_M^*$$ \hspace{1cm} (26)$$

Subtracting the numerator from the denominator of the coherence function results in

$$S_{xx}S_{dd} - S_{ad}S_{dx}$$

$$= \sum_{i,j=1}^{K} \left( S_{n_in_j} \left[ |H_i|^2|P_j|^2 + |H_j|^2|P_i|^2 - H_iH_j^*P_i^*P_j - H_j^*H_iP_iP_j^* \right] \right)$$
\[
\sum_{i,j=1 \text{ or } j} \left( S_{n_i n_j} \left( |H_i P_j - H_j P_i|^2 \right) \right) = 0
\]  \quad (27)

Since the auto-spectrum is non-negative, the above expression is also non-negative, which implies that the coherence function is usually less than unity, i.e.

\[
\gamma(w) \leq 1 \quad (28)
\]

This result indicates that perfect noise cancellation is not achievable with MRSI configuration. It should be noted that there are no unaccounted input signals, output signals, or nonlinear components in the system. If the system is configured properly, unity coherence function is expected. Thus, the coherence inadequacy is caused by the defect of MRSI configuration. It is also important to note that the MRSI configuration is defective even when the reference signals are uncorrelated. This conclusion can be seen by assuming that each noise source is available as a reference signal, accordingly, every noise coupling terms \( P_i \) in equation (27) drop out and equation (28) still holds. As a conclusion, compromised noise reduction is anticipated as a result of the defective MRSI control structure.

However, under two special circumstances: (1) when each reference signal occupies different frequency range, i.e. \( S_{n_i n_j} = 0 \) for \( i \neq j \); (2) when each noise source is available as a reference signal, and all the FRF's of the primary paths are equal, i.e. \( |H_i P_j - H_j P_i| = 0 \) for \( i \neq j \), equation (26) is equal to zero, which implies unity coherence function throughout the frequency range. Thus, when either one of the two
conditions is met, the performance of a MRSI configuration is expected to be equivalent to that of a MRMI configuration. For example, when several engines run at different speeds, each engine can be considered as a source radiating only tonal noise, since the base tone frequency is different, a MRSI configuration can be used to obtain desired noise reduction. This falls into the first circumstance. If the noise sources are adjacently located and available as reference signals without coupling, and the error sensors are far away from the noise sources, all the primary paths are expected to be comparable. As a result, desired noise reduction effect can be achieved with a MRSI configuration. This falls into the second circumstance. The choice of a MRSI configuration significantly simplifies a MRANC system as compared to a MRMI configuration.

4. Simulation setup and system modeling

In a complicated ANC system such as an aircraft cabin, the fuselage may generate interior noise due to directly applied structural forces as well as acoustical pressure fluctuations acting on it from exterior. The setup shown in Figure 3 gives consideration to these two types of excitations. As shown in Figure 3, there are two disturbance sources, one secondary source and one error microphone for the plate system. The plate has dimensions of 0.381m long and 0.305m wide and is mounted in a heavy steel frame, which produces negligible rotation and displacement of the boundary, approximating clamped boundary conditions. The steel frame is mounted in a rigid wall with one side facing toward a reverberation chamber and the other side toward an anechoic chamber.
The plate is excited by two distinctive noise sources; one is the acoustical disturbance from a large speaker, while the other is the structural disturbance from a piezoelectric ceramic transducer (PZT #1) mounted on the plate. The secondary control source acting on the plate is another piezoelectric actuator (PZT #2). The positions of both of the PZT’s are selected such that any plate mode of order (4,4) or less can be excited. The error sensor is a microphone located in the direction approximately perpendicular to the center of the plate. The goal of the control is to minimize the total radiated sound at the error microphone.

The system shown in Figure 7 has two primary paths (from the speaker through the plate to the error microphone and from the PZT #1 to the error microphone) and one error path (from the PZT #2 to the error microphone). The FRF’s of the two primary paths and the error path were modeled with FIR filters. In order to obtain the FIR models of the system, the FRF’s were measured with a B&K 2032 digital signal analyzer, and 801 frequency response data samples equally spaced between 0 to 400 Hz were obtained. These frequency response data were fitted with FIR filters using the least square method [11]. The frequency range is constrained to be below 400 Hz. Within the chosen frequency range, a maximum of five structural modes can be excited.

The number of coefficients in each FIR filter was chosen to be 128 such that both the phase and magnitude of the FRF can be well matched at the frequency range where large noise cancellation is desired. The match at other frequency ranges (e.g. below 40 Hz) is not very important, since the noise cancellation at those frequency ranges is
unobtainable due to the dynamic limitations of the PZT actuators. The sampling frequency is chosen to be 800 Hz, which is exactly the Nyquist frequency for the system. The magnitude response of measured FRF's and their corresponding FIR models are shown in Figure 4, 5 and 6.

5. Simulation results

In the first simulation, the performance difference of the two configurations (MRSI and MRMI) is examined with two independent random signal generators as the noise sources. For MRMI configuration, the two reference signals are obtained directly from the two random signal generators. For MRSI configuration, the reference signal is the summation of the signals from the two random signal generators. The optimum filter weight vectors for both configurations are calculated based on the norm equation

$$\mathbf{R} \mathbf{W}_{\text{opt}} = -\mathbf{P}$$

(29)

where

$$\mathbf{R} = \mathbf{E} \begin{bmatrix}
\mathbf{x}(k)\mathbf{x}^T(k) & \mathbf{x}(k)\mathbf{x}^T_2(k) & \Delta & \mathbf{x}(k)\mathbf{x}^T_k(k) \\
\mathbf{x}_2(k)\mathbf{x}^T_1(k) & \mathbf{x}_2(k)\mathbf{x}^T_2(k) & \Delta & \mathbf{x}_2(k)\mathbf{x}^T_k(k) \\
\mathbf{M} & \mathbf{M} & \mathbf{M} & \mathbf{M} \\
\mathbf{x}_k(k)\mathbf{x}^T_1(k) & \mathbf{x}_k(k)\mathbf{x}^T_2(k) & \Delta & \mathbf{x}_k(k)\mathbf{x}^T_k(k)
\end{bmatrix}$$

(30)

$$\mathbf{P} = \mathbf{E} \begin{bmatrix}
\mathbf{x}(k)d(k) \\
\mathbf{x}_2(k)d(k) \\
\mathbf{M} \\
\mathbf{x}_k(k)d(k)
\end{bmatrix}$$

(31)
The expected values inside the matrix \( R \) and the vector \( P \) were approximated using the average of 4096 samples. The residual noise after control is obtained by substituting the optimum weight vectors into controller filters.

The error signal after control (residual noise) in the time domain is shown in Figure 7, which indicates that the error signal for MRMI is much smaller than that for MRSI. In other words, MRMI achieves much better noise reduction than MRSI. Although perfect noise cancellation with MRMI can be achieved theoretically since the system is casual, completely coherent and without any additional noise, the error signal does not reach zero. This is due to the finite filter length effect since each controller filter for MRMI has only 128 coefficients. A comparison of the power spectral density is shown in Figure 8, which clearly indicates that the performance of MRMI is much better than that of MRSI. In fact, 10 dB overall noise reduction for MRMI is achieved, while only about 3 dB overall noise reduction is achieved for MRSI. It should be noted that the maximum noise reduction occurs in the vicinity of the resonance frequencies, while at the off-resonance frequencies, only a small amount of noise attenuation is achieved.

Since both noise sources in the preceding simulation has the same frequency range between zero to 400 Hz, the poor performance of MRSI is expected. However, if the two noise sources have non-overlapped frequency range, MRSI is expected to have comparable performance with MRMI. In this simulation, the first noise source is provided by a random signal generator passing through a low-pass filter with cutoff frequency at 160 Hz, and the second noise source is provided by another random signal generator.
generator passing through a high-pass filter with cutoff frequency at 160 Hz. Thus, each
noise source occupies a different frequency range. Each of the two filters in MRMI has
128 coefficients, and the only filter in MRSI also has 128 coefficients. A sequence of
error signal after control (residual noise) is shown in Figure 9, and the power spectral
density of the error signal before and after optimum control is shown in Figure 10. The
result shows that the performance of both configurations is close. However, since MRSI
has a very simple structure, significant amount of computations can be saved. It is
interesting to note that, under the assumption that each noise source has a unique
frequency range, MRSI approach is actually the inverse of the sub-band approach.

It has been theoretical predicted that if all the primary paths are similar to each
other and each noise source is available as a reference signal, then MRSI is expected to
have comparable performance as MRMI. In this simulation, the first primary path is
replaced by the variation of the second primary path, which deviates about 10% from the
original magnitude and phase, and the second primary path stays the same. Thus, the two
primary paths are similar. A sequence of error signal after control (residual noise) is
shown in Figure 11, and the power spectral density of the error signal before and after
optimum control is shown in Figure 12. The result clearly shows that the performance of
both configurations is very similar.
6. Conclusions

A frequency domain optimum solution of a multiple reference active noise control (MRANC) system based on a general configuration (MRMI) has been derived, and the coupling effect between noise sources and reference sensors has been considered. Another simplified control configuration in which the reference signal are directly added together to form a single signal has been investigated and compared with the MRMI configuration. When there are several noise sources, the MRMI configuration generally performs much better than the MRSI configuration. However, when each reference signal occupies a unique frequency range, or all the primary paths are similar to each other and each noise source is available as a reference signal, the two configurations can result in comparable noise reduction. In this situation, since the MRSI configuration is much less complex, a significant amount of computational and memory savings can be achieved.

7. Acknowledgments

The authors gratefully acknowledge the support of this work by the office of Naval Research under the Grant ONR N00014-94-1-1140, Dr. Kam Ng Technical Monitor.
Figure 1. Multiple reference multiple input (MRMI) system.
Figure 2. Multiple reference single input (MRSI) system.
Figure 3. Experimental setup.
Figure 4. Magnitude of the frequency response function and its FIR model for the primary path No. 1.
Figure 5. Magnitude of the frequency response function and its FIR model for the primary path No.2.
Figure 6. Magnitude of the frequency response function and its FIR model for the error path.
Figure 7. Error signal after control in a typical case, (a) MRSI (b) MRMI.
Figure 8. Spectrum of error signal in a typical case.
Figure 9. Error signal after control when each noise source occupies different frequency range, (a) MRSI, (b) MRMI.
Figure 10. Spectrum of error signal when each noise source occupies different frequency band.

J. Sound and Vibration. Y. Tu, C.R. Fuller Page 26
Figure 11. Error signal after control when two primary paths are similar a) MRSI, b) MRMI.
Figure 12. Spectrum of error signal when two primary paths are similar.
Reference


11 A. A. Giordano and F. M. Hsu, Least Square estimation with application to digital signal processing, Wiley, New York, 1985
Multiple reference feedforward active noise control

Part II  Reference preprocessing and experimental results

Yifeng Tu

Chris R. Fuller

Vibration and Acoustic Laboratories
Mechanical Engineering Department
Virginia Polytechnic Institute and State University
Blacksburg, Virginia 24061-0238

Running headline: Multiple reference feedforward active noise control

Address for correspondence:

Prof. Chris R. Fuller
Mechanical Engineering Department
Virginia Polytechnic Institute and State University
Blacksburg, Virginia 24061-0238

Number of Pages: 33

Number of tables: 2

Number of figures: 12

J. Sound and Vibration.  Y. Tu, C.R. Fuller  Page 1
1. Summary

Multiple reference active noise control (MRANC) has been applied to acoustical fields with multiple noise sources to achieve low frequency noise reduction. The traditional control configuration feeds each reference signal into a different control filter. This configuration has been widely adopted due to its potential performance in a general multiple noise source environment. However, it entails the problem of ill conditioning when the reference signals are correlated. In this paper, a time domain analysis has been carried out to investigate the problem of ill conditioning for MRANC. To cope with the problem of ill conditioning, a reference signal preprocessing step is added to the conventional active noise control process. This preprocessing step essentially constructs a new set of reference signals, which preserve all the information of the original reference, but are uncorrelated with each other. An adaptive decorrelation filter based on the Wiener filter theory and Gram-Schmidt orthogonalization theorem is constructed to implement the reference signal preprocessing step. Experiments based on sound transmission through a vibrating plate have been conducted and the results presented are consistent with the theoretical analysis.
2. Introduction

With the breathtaking advance of digital signal processing (DSP) technology and ever-increasing parallelism for real-time computation, complex feedforward active noise control (ANC) systems with multiple reference, multiple actuator and multiple error have been built. These complex ANC systems make noise reduction possible not only in single noise source one-dimensional acoustical fields, but in multiple noise source three-dimensional acoustical fields as well. When an ANC system is applied to a three-dimensional acoustical field, it is usually required to use multiple actuator and multiple error sensor to achieve spatial noise reduction [1].

In addition, to achieve noise reduction in a multiple noise source environment, it is usually required to use multiple reference sensor to generate a complete set of reference signals [2] so that the desired multiple coherence function can be obtained. A previous companion paper [3] studied the behavior of conventional and simplified multiple reference active noise control (MRANC) systems using frequency domain analysis and simulation. It was found that good performance was achieved when the reference signals were uncorrelated. However, if the reference signals are correlated, the MRANC system may become ill conditioned which usually results in a slow convergence rate and high sensitivity of noise attenuation to the measurement contamination.

A method to increase the convergence speed by using decorrelators for a MRANC system was proposed by Masato et. al. [4]. The effectiveness of the proposed
decorrelators to increase convergence speed is strongly affected by the characteristics of
the reference signal correlation. In a general situation, the proposed decorrelators deliver
poor results. Studies on adaptive noise cancellation, in which the cancellation is focused
on electrical signal rather than acoustical wave, have shown that the reference signal
coupling in a multiple noise source environment degrades the control performance. A de-
coupling method was proposed to produce a new set of reference signals [5]. Parallel
adaptive filter structures along with sub-band approaches have also been applied to
adaptive noise cancellation. The results are shown to improve the ability to track non-
stationary noise process [6]. Recently, attention has been paid on the convergence rate as
well as the number of control coefficients for reference sensor selection [7] in a multiple
noise source environment.

The previous studies on MRANC are mainly focused on noise source
identification and reference sensor selection. An important issue remains unsolved, i.e.
when reference signals are correlated, the ANC system becomes ill-conditioned. An ill-
conditioned system usually results in slow convergence rate and high sensitivity of noise
attenuation to measurement contamination. In this paper, the problem of ill conditioning
has been analytically studied using a time domain analysis. A reference signal
preprocessing step using adaptive decorrelation filters is proposed and investigated using
simulations and experiments. It is shown that the preprocessing step significantly improve
the performance of the MRANC system when the reference signals are correlated.
3. Time domain analysis

Figure 1 shows a typical MRANC system, in which there are M noise sources, K reference sensors, one secondary source and one error sensor. Each reference signal is fed into a different filter and the output of each filter is summed together to drive a single secondary source. This structure for processing reference signals attempts to control the primary noise from multiple primary paths with the secondary noise from multiple filters, thus it enjoys the applicability in general situations and has been adopted by most previous studies [2], [8]. Since the number of the secondary sources and error sensors is chosen to be unity to simplify the analysis, spatial noise reduction effect is not a concern in this paper.

Usually, reference signals are obtained through various sensors, and each reference sensor may pick up signals from several noise sources passing through different paths, as shown in Figure 3. As a result, the reference signals maybe correlated with each other. The correlated part of these reference signals represents the common input to different filters, which in turn generates the correlated outputs. The error signal at step k can be written as

\[ e(k) = d(k) + u(k)T(z) \]

where \( T(z) \) is the Z transform of the error path. Here, no assumption is made about the structure of \( T(z) \), it could be in the form of either IIR or FIR filters, and \( u(k) \) is the summation of all the filter outputs, given by

\[ u(k) = \sum_{i=1}^{K} x_i^T(k)w_i = \sum_{i=1}^{K} w_i^T x_i(k) \]
where \( w_i \) and \( x_i(k) \) are the weight vector and the tapped input vector for the \( i \)th filter respectively. Suppose the filter length is \( M \), then

\[
\mathbf{w}_i = \begin{bmatrix}
w_{0,i} & w_{1,i} & w_{2,i} & \ldots & w_{M-1,i}
\end{bmatrix}^T
\]  

(3)

\[
\mathbf{x}_i(k) = \begin{bmatrix}
x_i(k) & x_i(k-1) & x_i(k-2) & \ldots & x_i(k-M+1)
\end{bmatrix}^T
\]  

(4)

Substituting equation (2) into equation (1) yields

\[
e(k) = d(k) + \sum_{i=1}^{K} \mathbf{w}_i^T \mathbf{x}_i(k)T(z)
\]  

(5)

Defining the filtered reference signal as

\[
\bar{\mathbf{r}}_i(k) = x_i(k)T(z)
\]  

(6)

the error signal can be rewritten as

\[
e(k) = d(k) + \mathbf{W}^T \bar{\mathbf{r}}(k)
\]  

(7)

where \( \mathbf{W} \) is a \( K \times M \) vector formed by putting together all the control filter weight vectors and \( \bar{\mathbf{r}}(k) \) is another \( K \times M \) vector formed by putting together all the reference signal vectors, i.e.

\[
\mathbf{W} = \begin{bmatrix}
\mathbf{w}_1^T & \mathbf{w}_2^T & \ldots & \mathbf{w}_K^T
\end{bmatrix}^T
\]  

(8)

\[
\bar{\mathbf{r}}(k) = \begin{bmatrix}
\bar{\mathbf{r}}_1^T(k) & \bar{\mathbf{r}}_2^T(k) & \ldots & \bar{\mathbf{r}}_K^T(k)
\end{bmatrix}^T
\]  

(9)

The cost function is constructed as the mean square error signal, i.e.

\[
\xi = E[e^2(k)]
\]  

(10)

Substituting equation (5) into equation (10), the gradient of the cost function with respect to the weight vector is obtained as

\[
\nabla = \frac{\partial \xi}{\partial \mathbf{W}} = 2E\left[ e(k) \frac{\partial e(k)}{\partial \mathbf{W}} \right]
\]
\[ = 2E[d(k)\bar{x}(k)] + 2E[\bar{x}(k)\bar{x}^T(k)]W \]

Setting the gradient to zero, the optimum weight vector is obtained by solving the norm equation

\[ RW_{opt} = -P \]

where

\[ R = E \begin{pmatrix}
    \bar{x}_1(k)\bar{x}^T_1(k) & \bar{x}_1(k)\bar{x}^T_2(k) & \Lambda & \bar{x}_1(k)\bar{x}^T_k(k) \\
    \bar{x}_2(k)\bar{x}^T_1(k) & \bar{x}_2(k)\bar{x}^T_2(k) & \Lambda & \bar{x}_2(k)\bar{x}^T_k(k) \\
    \Lambda & \Lambda & \Lambda & \Lambda \\
    \bar{x}_k(k)\bar{x}^T_1(k) & \bar{x}_k(k)\bar{x}^T_2(k) & \Lambda & \bar{x}_k(k)\bar{x}^T_k(k)
\end{pmatrix} \]

\[ P = E \begin{pmatrix}
    \bar{x}_1(k)d(k) \\
    \bar{x}_2(k)d(k) \\
    \Lambda \\
    \bar{x}_k(k)d(k)
\end{pmatrix} \]

Each term inside the above \( R \) matrix is a sub-matrix, and the matrix \( R \) is real, symmetric and non-negative definite just like the \( R \) matrix in a single reference ANC system. Therefore, the corresponding eigenvalues of the \( R \) matrix are also non-negative and real. The characteristics of the \( R \) matrix are determined by the auto-correlation and cross-correlation functions of the filtered reference signals. If all the reference signals are uncorrelated with each other, every off-diagonal terms in the \( R \) matrix will be zero and every optimum weight vectors, \( w_1, w_2, \ldots \) and \( w_k \), are uncoupled. On the other hand, if the reference signals are correlated, the condition number [9] of the \( R \) matrix may become large, which implies that the MRANC system is ill-conditioned. This problem becomes even more prominent when each reference signal is orthogonal, i.e. the auto-correlation matrices inside the \( R \) matrix are diagonal.
It is very important to understand *perturbation theory* [9] and its impact on the development of an algorithm to solve the norm equation (11). The perturbation theory states that if the matrix $R$ and the vector $P$ are perturbed by small amounts $\delta R$ and $\delta P$ respectively, and if the relative perturbations, $\|\delta R\|/\|R\|$ and $\|\delta P\|/\|P\|$, are both on the same order of $\varepsilon$, where $\varepsilon \ll 1$, then

$$\frac{\|\delta W\|}{\|W\|} \leq \varepsilon \chi(R) \quad (14)$$

where $\delta W$ is the change of weight vector $W$ as a result of the perturbation from the matrix $R$ and the vector $P$, and $\chi(R)$ is the *condition number* of the matrix $R$, and $\|\|$ is the *norm* operator [10]. The condition number describes the ill condition of a matrix. Since the matrix $R$ is real and symmetric, it can be shown [6] that the condition number equals

$$\chi(R) = \frac{\lambda_{\text{max}}}{\lambda_{\text{min}}} \quad (15)$$

where $\lambda_{\text{max}}$ and $\lambda_{\text{min}}$ are the maximum and minimum eigenvalues of the matrix $R$ respectively. This ratio is also commonly referred to as eigenvalue spread.

The perturbation theory states that if there are some errors in the matrix $R$ or the vector $P$ caused by measurement or some other factors, the ill-condition of the correlation matrix $R$ may lead to a weight vector solution $W$ which is far from the optimum Wiener solution $W_{\text{opt}}$ due to the problem of ill condition. In other words, the ill condition of the correlation matrix $R$ causes the optimum Wiener vector to be very sensitive to various measurement contamination. The measurement contamination may be resulted from A/D and D/A conversion, transducer error, finite precision error, non-linearity and etc.

J. Sound Vibration Y. Tu, C.R.Fuller
On the other hand, the eigenvalue spread of the matrix $R$ has a significant impact on the convergence rate of an ANC system, especially when the LMS based algorithm is applied [2]. An important factor that determines the eigenvalue spread is the cross-correlation among the reference signals. The cross-correlation becomes even more dominant when each reference is an orthogonal signal. In particular, if the reference signal $x_i$ is correlated with the reference signal $x_j$, the $i$th and $j$th columns in the $R$ matrix will exhibit some similarities, which results in large eigenvalue spread. Upon the extreme circumstance when any two reference signals are the same, the determinant of the matrix $R$ becomes zero, and the eigenvalue spread approaches infinity. In this case, the ill-conditioned system deteriorates to an underdetermined system since the solution to the ANC system is not unique.

4. Preprocessing of reference signals

As discussed in the last section, it is desirable to have uncorrelated reference signals. In the present work, this is achieved through the use of decorrelation filters in a reference signal preprocessing stage. The preprocessing constructs a new set of reference signals, which preserve all the information of the original reference signals, but are uncorrelated with each other. Such a step essentially causes all the off-diagonal blocks in the $R$ matrix to be zero, thus eliminating the ill-conditioning problem due to the cross-correlation among reference signals.
According to the orthogonal theorem of adaptive filter theory, when an adaptive filter shown in Figure 4 is running as an optimum Wiener filter, the error signal is uncorrelated with the reference signal, i.e.

\[
E[x(k-i)e(k)]_{w=w_{op}} = 0 \quad i = 0, 1, 2, \ldots, M-1
\]

where \( x \) is the reference signal, \( e \) is the error signals, and \( M \) is the number of filter coefficients. Based on the orthogonal theorem, adaptive decorrelation filters can be constructed as shown in Figure 5, in which two correlated reference signals are processed by a couple of adaptive filters to generate two uncorrelated reference signals. For the upper filter A, the reference signal is \( \bar{x}_1 \), and the error signal is \( \bar{x}_2 \). Thus, the orthogonal relationship is expressed as

\[
E[\bar{x}_1(k-i)\bar{x}_2(k)]_{A=A_{op}} = 0 \quad i = 0, 1, 2, \ldots, M-1
\]

where \( M \) is the number of filter coefficients corresponding to filter A. For the lower filter B, the reference signal is \( \bar{x}_2 \), and the error signal is \( \bar{x}_1 \). Thus, the orthogonal relationship is expressed as

\[
E[\bar{x}_2(k-i)\bar{x}_1(k)]_{B=B_{op}} = 0 \quad i = 0, 1, 2, \ldots, M-1
\]

where \( M \) is the number of filter coefficients corresponding to filter B. It is important to note that the degree of decorrelation between the reference signal and the error signal depends on the number of filter coefficients. Ideally, an infinite number of filter coefficients is needed to decorrelate the two reference signals. However, since the objective of applying decorrelation filters is to diagonalize the matrix \( R \), it is easy to show that the two transversal filters in the decorrelation filters should have the same
number of coefficients, and the number of coefficients M should be the same as the number of the control filter coefficients.

It should be noted that the first two coefficients $a_0$ and $b_0$ in the filters $A$ and $B$ are redundant, since both of them are trying to achieve

$$E[\tilde{x}_2(k)\tilde{x}_1(k)] = 0$$

This is an over-determined case, which gives infinite solutions to $a_0$ and $b_0$. A practical approach is to force either $a_0$ or $b_0$ to be zero. Using the LMS algorithm and assuming that first coefficient $b_0$ is set to be zero, the uncorrelated reference signals $\tilde{x}_1$ and $\tilde{x}_2$ can be obtained as

$$\tilde{x}_1(k) = x_1(k) + \sum_{i=1}^{M} \tilde{x}_2(k-i)b_i(k) \quad i = 1, 2, \ldots, M-1$$

$$b_j(k+1) = b_j(k) - \mu \tilde{x}_2(k-i)\tilde{x}_1(k) \quad i = 1, 2, \ldots, M-1$$

$$\tilde{x}_2(k) = x_2(k) + \sum_{i=0}^{M} \tilde{x}_1(k-i)a_i(k) \quad i = 0, 1, 2, \ldots, M-1$$

$$a_i(k+1) = a_i(k) - \mu \tilde{x}_1(k-i)\tilde{x}_2(k) \quad i = 0, 1, 2, \ldots, M-1$$

The above decorrelation techniques have been reported to achieve signal separation and restoration of original signals [11], [12]. In these applications, assumptions on the relationship between source signals and input signals have to be made, and also required are some assumptions on the statistical properties of source signals. In the pre-processing, the reference signals are assumed to be wide sense stationary, however, the relationship between noise sources and reference signals is not important. A decorrelation structure for $K$ reference signals is shown in Figure 6.
The correlation matrix $\mathbf{R}'$ of decorrelated reference signals is generally not strictly diagonal, but block-diagonal, since each term inside the matrix is a sub-matrix. Although the eigenvalue spread is generally smaller after the preprocessing with the decorrelation filters, it is not guaranteed at every situation. In particular, if the cross-correlation terms are much smaller than the auto-correlation terms in the matrix $\mathbf{R}$, the decorrelation preprocess may not improve eigenvalue spread at all. However, if reference signal is orthogonal, which implies that each diagonal sub-matrix inside the matrix $\mathbf{R}'$ is diagonal, then the decorrelation process will definitely improve the eigenvalue spread. A potential approaches to achieve this is to combine decorrelation process with lattice structure based FIR filters [13], [14] or frequency-domain block algorithm [15] [16].

5. Experimental setup

In a complicated ANC system such as an aircraft cabin, the fuselage may generate noise due to directly applied structural forces as well as acoustical pressure fluctuations. The setup shown in Figure 7 gives consideration to these two types of excitations. As shown in Figure 7, there are two disturbance sources, one secondary source and one error microphone for the plate system. The plate has dimensions of 0.381 m long and 0.305 m wide and is mounted in a heavy steel frame, which produces negligible rotation and displacement of the boundary, approximating clamped boundary conditions. The steel frame is mounted in a rigid wall with one side facing toward a reverberation chamber and the other side toward an anechoic chamber. The plate is excited by two distinctive noise sources; one is the acoustical disturbance from a large speaker, while the other is the
structural disturbance from a piezoelectric transducer (PZT #1) mounted on the plate. The secondary control source acting on the plate is another PZT (PZT #2). The positions of the PZT's are selected such that any plate mode of order (4,4) or less can be excited as shown in Table 1. The error sensor is a microphone located in the direction approximately perpendicular to the center of the plate. The goal of the control is to minimize the total radiated sound at the error microphone.

The block diagram showing the various elements for the experiments is presented in Figure 8. The heart of the system is a TMS320C30® DSP board, which is used to implement the preprocessing and control algorithms. The A/D and D/A conversions are carried out through two additional I/O boards, which provide 32 input channels and 16 output channels. The DSP board along with two associated I/O boards is plugged into a PC. A graphical user interface (GUI) running under the host PC is provided to adjust various control parameters and display DSP data. Since the disturbance signals generated within the DSP are digital in nature, they are transformed into analog signals through D/A converters to drive the primary noise sources (speaker and PZT #1). Similarly, the control signal is also transformed into analog signal through D/A converters to drive the secondary source (PZT #2). Since the frequency range for the experiments is selected to be below 400 Hz, all the signals are low-pass filtered so that the frequency components above 400 Hz are negligible in order to avoid alias. In addition, since the signals generated within the DSP have very small power, in order to drive the speaker and PZTs, they are also fed into power amplifiers. It should be noted that the signal from the error microphone is fed into a high-pass filter to eliminate the dc signal drift.
The controller is based on adaptive FIR filters, and the number of coefficients for each FIR filter is selected to be 128. The error path between the secondary source (PZT #2) and the error sensor is also modeled with a FIR filter, and the same number of coefficients is used. The frequency range of the noise field is selected to be below 400 Hz. These parameters are selected based on the computational limit of the DSP, since increasing frequency range generally requires more filter coefficients to get satisfactory results, which in turn requires more computations.

Two random signal generators were used to produce two independent noise sources, and the reference signals were obtained indirectly from the two noise sources. The relationship between the reference signals and the noise sources is shown in Figure 9. The first reference signal $r_1$ is exactly the same as the first noise source signal $n_1$. The second reference signal $r_2$ is the combination of the second noise source $n_2$ and the first noise source $n_1$ filtered through a band-pass filter, that is

$$r_1(k) = n_1(k)$$

$$r_2(k) = C_0 * n_2(k) + n_1(k)H(Z)$$

where $C_0$ is a constant. The two uncorrelated noise source signals, $n_1$ and $n_2$, are uniformly distributed between -1 and 1. The cut off frequencies for the band pass filter $H(z)$ are selected to be 160 Hz and 320 Hz. A FIR filter with four coefficients is used here to implement the band pass filter. The windowing method is adopted to design the band pass FIR filter and the resultant four coefficients vector is $[-1.009, 6.875, 6.875, -1.009]$. Thus, the two reference signals in equations (22) and (23) are correlated due to the common contributions from noise source $n_1$, and their correlation function can be varied if a different constant $C_0$ is selected.
The decorrelation filters are implemented with fixed Wiener filters instead of adaptive filters since the correlation between the reference signals is time invariant. In this particular case, the requirement for the decorrelation filter is to remove the component in the reference signal \( r_2 \), which is correlated with the reference signal \( r_1 \). The weight vector for the decorrelation filter \( A \) is calculated as \{1.025, -6.888, -6.863, 1.003\}, and the weight vector for the decorrelation filter \( B \) is a zero vector.

Corresponding to each selected constant \( C_0 \), computations are carried out to obtain the correlation matrix \( R \) for the reference signals, the decorrelated reference signals, the filtered reference signals, and the decorrelated filtered reference signals. Based on the correlation matrix, the corresponding eigenvalue spread is also computed and the results are shown in Table 2. It is clear that the eigenvalue spread of the reference signals is smaller after it is processed through the decorrelation filter. In fact, the convergence speed is determined by the filtered reference signals instead of the reference signals for the Filtered-X LMS (FXLMS) algorithm. Thus, in order to improve the convergence speed, the eigenvalue spread for the filtered reference signals must get smaller as well. This requirement is indeed satisfied since, although decorrelation is only applied to the reference signals, the correlation between filtered reference signals is also affected. It is also interesting to note the eigenvalue spread for the filtered reference signals is much larger than that for the reference signals. This is due to the stretching effect of the error path.
6. Experimental results

If the reference signals are preprocessed through the decorrelation filter, then the FXLMS algorithm is referred to as the DFXLMS algorithm. In order to examine the effect of the decorrelation filter on improving the convergence speed, the conventional FXLMS algorithm was first applied. After letting the controller weight vectors converge for 30 seconds, the convergence process was frozen. Then the power spectral density of the error signal was measured. Next, the DFXLMS algorithm was also applied, the power spectral density of the error signal was also measured after 30 seconds of convergence time. The results, shown in Figure 10, indicate that 9.0 dB noise reduction was achieved with the DFXLMS algorithm, while only 5.3 dB noise reduction with the conventional FXLMS algorithm.

In order to measure the learning curve, the convergence process of the FXLMS algorithm or the DFXLMS algorithm was started with a small convergence parameter, and the error signal power at 5, 10, 15, 25 ... and 360 seconds were measured. These values representing the power of the error signal form the learning curves as shown in Figure 11. The results indicate that the DFXLMS algorithm converges about three times faster than the FXLMS algorithm and the improvement of the convergence speed is in the same magnitude as the improvement of the corresponding eigenvalue spread. It is interesting to note that the error signal power for the DFXLMS is still larger than that for the FXLMS algorithm after 360 seconds. This is not only due to very long convergence time needed for the error signals to reach steady state, but due to various measurement errors (e.g. A/D and D/A conversion, non-linearity, finite precision) introduced in the experiment.
In order to investigate the impact of measurement errors, a simulation based on the same experimental setup was carried out using the approach outlined in the companion paper. The frequency response functions (FRF) of the two primary paths (from the speaker through the plate to the error microphone and from the PZT #1 to the error microphone) and one error path (from the PZT #2 to the error microphone) were measured with a B&K 2032 digital signal analyzer. The frequency response functions were fitted with FIR filters using the least square method [17]. Based on the FIR models, the learning curves of the two algorithms can be obtained through calculation. The results, shown in Figure 12, show the same tendency that the convergence speed of the DFXLMS algorithm is about three times as fast as that of the traditional FXLMS algorithm. However, after 6 minutes of convergence, the power of the error signal without the DFXLMS algorithm is very close to that with FXLMS algorithm since the simulation is free from various measurement errors caused by A/D, D/A converters, transducers, non-linear components and many other ANC elements. These simulation results, as compared to the corresponding experimental results, clearly indicate that the DFXLMS algorithm is effective to improve the noise reduction amount when there is noise on the measurement of signals.

7. Conclusion

The optimum solution of a multiple reference feedforward active noise control system has been obtained in time domain. It was particularly noted that if the reference signals are correlated, the corresponding system would be ill-conditioned, which results in slow
convergence speed and high sensitivity to measurement errors. A decorrelation approach has been constructed for preprocessing reference signals based on the Wiener filter theory and Gram-Schmidt orthogonalization theorem. Experiments based on sound transmission through a vibrating plate has been conducted, and the results demonstrate the effectiveness of the decorrelation approach to improve both the convergence speed and steady state error.

8. Acknowledgments

The authors gratefully acknowledge the support of this work by the office of Naval Research under the Grant ONR N00014-94-1-1140, Dr. Kam Ng Technical Monitor.
Figure 1. Multiple reference multiple input (MRMI) system.
Figure 2. Noise sources and reference sensors.
Figure 4. An adaptive transversal filter.
Figure 5. An adaptive decorrelation filter.
Figure 6. Adaptive Decorrelation Filter for K reference signals.
Figure 7. Experimental setup.
Figure 8. Block diagram of the various elements for the experiments.
Figure 9. Noise sources and reference sensors
Figure 10. Spectrum of error signal after 30 seconds convergence time.
Figure 11. Comparison of learning curves based on experiment.
Figure 12. Comparison of learning curves based on simulation.
Table 1 Excited modes and measured natural frequencies of the clamped plate.

<table>
<thead>
<tr>
<th>Mode</th>
<th>(1,1)</th>
<th>(2,1)</th>
<th>(1,2)</th>
<th>(2,2)</th>
<th>(3,1)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Frequency (Hz)</td>
<td>115</td>
<td>201</td>
<td>265</td>
<td>342</td>
<td>350</td>
</tr>
</tbody>
</table>
Table 2 Eigenvalue spread versus correlation

<table>
<thead>
<tr>
<th>$C_0$</th>
<th>Reference signals without decorrelation</th>
<th>Reference signals with decorrelation</th>
<th>Filtered reference without decorrelation</th>
<th>Filtered reference with decorrelation</th>
</tr>
</thead>
<tbody>
<tr>
<td>1.0</td>
<td>17.7</td>
<td>3.2</td>
<td>$8.4 \times 10^6$</td>
<td>$2.3 \times 10^3$</td>
</tr>
<tr>
<td>0.5</td>
<td>48.2</td>
<td>3.4</td>
<td>$2.9 \times 10^6$</td>
<td>$3.1 \times 10^5$</td>
</tr>
<tr>
<td>0.2</td>
<td>242.5</td>
<td>3.0</td>
<td>$1.1 \times 10^7$</td>
<td>$2.2 \times 10^5$</td>
</tr>
</tbody>
</table>


17 A. A. Giordano and F. M. Hsu, Least Square estimation with application to digital signal processing, Wiley, New York, 1985