Compensating for Pneumatic Distortion in Pressure Sensing Devices

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Abstract

In this paper a technique of compensating for pneumatic distortion in pressure sensing devices is developed and verified. This compensation allows conventional pressure sensing technology to obtain improved unsteady pressure measurements. Pressure distortion caused by frictional attenuation and pneumatic resonance within the sensing system makes obtaining unsteady pressure measurements by conventional sensors difficult. Most distortion occurs within the pneumatic tubing which transmits pressure impulses from the aircraft's surface to the measurement transducer. To avoid pneumatic distortion, experiment designers mount the pressure sensor at the surface of the aircraft, (called in situ mounting). In situ transducers cannot always fit in the available space and sometimes pneumatic tubing must be run from the aircraft's surface to the pressure transducer. Research began at the National Aeronautics and Space Administration Ames Research Center Dryden Flight Research Facility, supported by the University of California, Los Angeles, to develop a technique to measure unsteady pressure data using conventional pressure sensing technology. A pneumatic distortion model is reduced to a low-order, state-variable model retaining most of the dynamic characteristics of the full model. The reduced-order model is coupled with results from minimum variance estimation theory to develop an algorithm to compensate for the effects of pneumatic distortion. Both postflight and real-time algorithms are developed and evaluated using simulated and flight data.

Nomenclature

Acronyms

BVP boundary value problem

Symbols

\( A_c \) cross-sectional area of pneumatic tubing

\( B \) state equation input geometry matrix

\( C \) observation matrix

\( COV \) covariance

\( c \) sonic velocity

\( D \) pressure tubing cross-sectional diameter

\( dx \) spatial coordinate differential

\( G \) stationery measurement error variance

\( G_{k+1} \) measurement noise covariance matrix

\( G(1) \) first diagonal component of \( G_{k+1} \)

\( G(2) \) second diagonal component of \( G_{k+1} \)

\( g_{k+1} \) measurement noise function

\( I \) identity matrix

\( I \) input to deconvolution algorithm

\( \hat{I} \) deconvolution input estimate

\( K_{k+1} \) Kalman filter gain matrix

\( k \) data frame index

\( L \) length of pneumatic tubing

\( N \) number of data frames available

\( O \) output from pressure sensor

\( \hat{O} \) deconvolution output estimate

\( P_0 \) initial local pressure

\( P_{k+1/k} \) prediction error covariance matrix

\( P_{k+1/k+1} \) filter error covariance matrix

\( P_{k+1/N} \) smoother error covariance matrix

ESP electronically scanned pressure module

DC direct current
Introduction

Recent advances in aircraft performance and maneuver capability have complicated the problem of flight control augmentation. Increasingly, aircraft system designs use aerodynamic parameters derived from pneumatic measurements as control system feedbacks. To be used as control feedbacks, pneumatically derived data must be measured with accuracy and dynamic fidelity. This has been difficult because of pressure distortion caused by frictional attenuation and pneumatic resonance within the sensing system. Typically, most distortion occurs within the pneumatic tubing used to transmit pressure impulses from the aircraft’s surface to the measurement transducer.

To avoid pneumatic distortion, experiment designers mount the pressure sensor at the aircraft’s surface, called in situ mounting. In some cases this technique is a viable solution. In most cases, as when many pressures must be measured in a small surface area or when pressure is sensed in an external environment, pressure sensor installations require pneumatic tubing from the aircraft’s surface to the pressure transducer.

New non-intrusive pressure sensing technologies, such as piezo-film or optical techniques such as laser/Doppler, need to be developed. These technologies, however, are still being developed and will not be reliably proven for at least a decade. Additionally, these technologies are expensive. A research effort was initiated at the National Aeronautics and Space Administration Ames Research Center Dryden Flight Research Facility, supported by the University of California, Los Angeles, to develop a technique for measuring unsteady pressure data with conventional pressure sensing technology. Preliminary results of this research have been published in Refs. 1–3. This paper presents further results.

A pneumatic distortion model, developed in the early stages of this research, will be reduced to a low-order, state-variable model which retains most of the dynamic characteristics of the full model. The reduced-order model will be coupled with standard results from minimum variance estimation theory to develop an algorithm to compensate for the effects of pneumatic distortion. Both postflight and real-time algorithms are developed. Algorithm characteristics are evaluated using simulation and flight data.

Background

For most flight test applications pressure sensing devices are located remotely from the aircraft surface, usually within the internal cavity of the aircraft. A connecting length of pneumatic tubing transmits pressure changes at the aircraft’s surface to the pressure transducer. Pressure variations at the surface propagate as waves from the upstream end through the pneumatic tubing to the transducer. The wave propagation is damped by frictional attenuation along the walls of the pneumatic tubing. This damping causes spectral attenuation of the pressure response and produces both a magnitude attenuation and a phase lag. When the wave arrives at the downstream (transducer) end of the pneumatic tubing, it is reflected back up the tube. Depending on the frequency of the incoming waves and the length of the pneumatic tubing, the reflected waves either cancel or reinforce incoming pressure waves. If the waves cancel, further spectral attenuation occurs; if the waves reinforce, the power of the incoming wave is amplified and acoustic resonance occurs. Spectral distortion will not occur in zero volume configurations (with in situ mounting) only. This paper will discuss modeling and compensating for the effects of internal acoustic dynamics of the pressure sensor. External effects, such as orifice disturbance of the surface flow, will not be modeled.

A large body of information on the effects of pneumatic distortion is available. Early analyses modeled the pneumatic tubing dynamics for specialized inputs assuming either highly damped or undamped measurement configurations. While accurately predicting pressure responses for
defined types of inputs, these methods could not predict the 
behavior of pneumatic systems subjected to arbitrary pres-
sure inputs.

A complex wave model which evaluates the sensor out-
put for arbitrary inputs was developed in Refs. 2 and 3. The 
model, based on the unsteady Navier-Stokes equations of 
momentum and continuity, was verified by comparisons to 
special case analytical solutions, laboratory data, and flight 
data. The results of this verification indicate that predic-
tions of the wave model are extremely accurate. Refer-
ences 1–3 give detailed descriptions of the tests used to ver-
ify the model.

Because the wave model is complex, it cannot be easily 
inverted to develop a compensation routine which predicts 
the sensor input (the surface pressure) from the sensor out-
put (the measured pressure). Thus the wave model must 
be reduced to a simpler, lower order model retaining most of 
the dynamic characteristics of the full wave model. The 
lower order model is more amenable to analytical manipula-
tion and will be used to derive the compensation algorithms.

Modeling and Analysis

This section presents the idealized configuration to be an-
alyzed first. Next, the full wave model of Refs. 2 and 3 will 
be approximated by a simple second-order filter which re-
tains most of the dynamic characteristics of the complete 
model. The reduced-order model will be written in state-
variable form and its frequency response will be compared 
with the full wave model’s.

Idealized Pressure Sensor Geometry and Conditions

The pressure sensor configuration (Fig. 1) is modeled as 
a straight, cylindrical tube with an axisymmetric volume at-
tached at the downstream end. The tube is of constant di-
ameter (D) and length (L). The attached volume (V), rep-
resenting the internal volume of the pressure transducer, is 
also constant. A longitudinal coordinate (x) starts at the up-
stream end of the tube, and a time coordinate (t) starts at 
some initial time.

A small pressure pulse is introduced at the upstream end 
of the pneumatic tubing and propagates down the tube as a 
longitudinal wave. The flow velocity is approximated by the 
radial average, and as a result the pressure within the sensor 
is modeled as a function of x and t only. Since the sensor 
is closed at one end, flow velocities are small and temperature 
within the sensor has only small variations. Since the flow 
velocities are small, temperature within the sensor is 
nealy constant.

Governing Equations

The governing equations of the full wave model are de-

\[
\frac{\partial^2 P(x,t)}{\partial t^2} + \frac{R}{\rho_0} \frac{\partial P(x,t)}{\partial t} = c^2 \frac{\partial^2 P(x,t)}{\partial x^2}
\]

where \(\rho_0\) is the initial density, \(P(x,t)\) is the local pressure 
within the tubing, and \(c\) is the local sonic velocity. The pa-
rameter \(R\) is the acoustical resistance which for laminar flow 
given by

\[
R = \frac{32\mu}{D^2}
\]

where \(\mu\) is the local dynamic viscosity. The system is 
assumed to be at rest initially; thus the defined initial condi-
tions are

\[
P(x,0) = P_a, \quad \frac{\partial P(x,t)}{\partial t} \bigg|_{t=0} = 0
\]

where \(P_a\) is an arbitrary local starting pressure and the de-
defined pressure function at the upstream boundary is

\[
P(0,t) = P_0(t)
\]

Satisfying momentum and continuity at the downstream 
boundary

\[
\frac{\partial^2 P(L,t)}{\partial t^2} + \frac{R}{\rho_0} \frac{\partial P(L,t)}{\partial t} + \frac{A_c}{V} \frac{\partial P(x,t)}{\partial x} \bigg|_{x=L} = 0
\]

where \(P(L,t)\) is the sensor response at the pressure trans-
ducer. Equations (1) to (5) represent a completely and prop-
erly posed BVP. The existence and uniqueness of its so-
lution is demonstrated in Ref. 3. Numerical solutions are 

obtained using implicit differencing in which the pressure 
function is approximated by a series of discrete gridpoints, 
with partial derivative operators approximated by finite 
differences. The resulting nonsymmetric tridiagonal system is 
solved using traditional elimination techniques. 10

Reducing the Wave Model to a Second-Order Equation

Analyses of numerical solutions to the wave model indi-
cate that for small-amplitude pressure inputs to low-volume 
sensors, the pressure gradient at the downstream boundary 
can be approximated by the spatial average of the gradient 
within the pneumatic tubing without substantially altering 
the dynamic characteristics of the model. If this approxima-
tion is performed and the BVP is integrated twice with re-
spect to position, the wave model reduces to a second-order 
equation of the form

\[
\frac{d^2 P_L(t)}{dt^2} + 2\xi \omega_n \frac{dP_L(t)}{dt} + \omega_n^2 P_L(t) = \omega_n^2 P_0(t)
\]

where \(\omega_n\) is the natural frequency, and \(\xi\) is the equivalent 
damping ratio. For this model

\[
\omega_n^2 = \frac{C}{L V_c}
\]

\[
2\xi \omega_n = \frac{R}{\rho_0}
\]
If equation (6) is integrated using a backward zero-order hold, the reduced-order model can be written in state-variable form as the matrix equation

$$
\begin{pmatrix}
P_{k+1} \\
\end{pmatrix} = 
\begin{pmatrix}
0 & 1 \\
-1 & \frac{2(t+1) + (u_0\Delta t)}{1 + 2(u_0\Delta t)^2 + (u_0\Delta t)^2}
\end{pmatrix} 
\begin{pmatrix}
P_{k-1} \\
P_k
\end{pmatrix} + \begin{pmatrix}
0 \\
\frac{(u_0\Delta t)^2}{1 + 2(u_0\Delta t)^2 + (u_0\Delta t)^2}
\end{pmatrix} P_{0k+1}
$$

(7)

As mentioned earlier, equation (7) retains most of the frequency and time response characteristics of the original wave model up through the first wave harmonic. This is shown by comparing the analytical frequency responses for the full and reduced-order models. Figure 2 shows frequency response results for an overdamped sensor with pneumatic tubing 8 ft long and 0.06-in. diameter and a negligible volume transducer. The operating altitude is 40,000 ft. The equivalent damping ratio of the reduced-order model is 1.93, and the natural frequency is 26.75 Hz. Figure 3 shows frequency response comparisons for a lightly damped sensor with the same configuration operating at an altitude of 20,000 ft. For this configuration, the equivalent damping ratio of the reduced-order system is 0.25 and the natural frequency is 108.93 Hz.

For both figures the comparisons are good up to the natural frequency of the reduced-order model. Beyond this both the magnitude and phase angle comparisons begin to diverge. The reason for this divergence is clear; the second-order approximation does not allow for higher order wave harmonics, while the full wave model does. Thus at frequencies beyond the first harmonic, the reduced-order model may not be acceptable and results of the compensation routine in the next section will probably not be accurate beyond the first wave harmonic. For analysis beyond the first wave harmonic a higher order model which allows for additional wave harmonics can be used. The model parameters can be matched to the full wave model using system identification or resonance peak matching techniques.

Deriving the Compensation Algorithm

The compensation algorithm will be developed in this section. The objective is to develop an algorithm to infer the external pressure input to the measurement system based on observations of the pressure response at the transducer. In general, the technique of inferring a system input when given knowledge of the system structure and a measurement of the system response is called deconvolution. For this application, this is a difficult task.

If equation (7) is solved for \( P_{0k+1} \), in terms of \( P_{k-1} \), \( P_k \), and \( P_0 \), the resulting equation is numerically ill-conditioned; small changes in \( P_k \) produce large changes in \( P_{0k+1} \). As a result, noise in the pressure measurements at the transducer end of the sensor is over amplified by the numerical instability of the computation and overwhelms any estimate of the input pressure. The measurement noise must be identified and controlled as a part of an estimation loop. The algorithm that performs the error control function, presented in this section, will be used to infer the external pressure input to the system based on observations of the pressure response at the transducer.

First, a general deconvolution algorithm as developed in Ref. 3 will be presented. The technique, developed using the techniques of minimum variance estimation theory, is similar to results presented in Refs. 12-14. Both postflight and real-time versions of the algorithm will be presented. After the general algorithms are presented, they will be applied to the pressure compensation problem.

Minimum Variance Deconvolution Algorithm

The deconvolution algorithm defines an input to the state-variable model and identifies the residual between the prescribed input and the actual input using minimum variance estimation techniques. All notational conventions in this paper follow the conventions of Ref. 15. The assumed form of the model is

$$X_{k+1} = \Phi X_k + B[\Gamma_{k+1} + \delta_{k+1}]$$

(8)

$$Z_{k+1} = CX_{k+1} + g_{k+1}$$

(9)

where,

$$U_{k+1} = \Gamma_{k+1} + \delta_{k+1}$$

(10)

In equation (10) \( U_{k+1} \) is the actual input to the system, \( \Gamma_{k+1} \) is the defined (nonrandom) input to the system model, and \( \delta_{k+1} \) is a random process which models the residual between the actual and prescribed inputs. The covariance of \( \delta_{k+1} \) is defined as

$$COV[\delta_{k+1}] = Q_{k+1}$$

and is assumed to be known.

Measurements of the state vector \( Z_{k+1} \) are observed in the presence of noise. The measurement noise covariance is given as

$$COV[g_{k+1}] = G_{k+1}$$

and is assumed to be known.

For notational convenience, the state transition matrix \( \Phi \) and the input and measurement geometry matrices \( (B \) and \( C \) are written as constants.
Two complete estimation algorithms will be presented. The first, intended for postflight data reconstruction, is implemented as a two-pass forward filter, backward smoother. As developed in Ref. 3, the postflight smoothing algorithm relies on the use of all available data measurements, both past and future. The algorithm, a hybrid of the Kalman filter and the Rauch smoother, uses information in the Kalman filter innovations to identify differences between the defined input and the true input during the reverse pass of the algorithm. The second algorithm, intended for real-time operation, will be derived from the smoothing algorithm and implemented as a time-recursive filter. This algorithm does not use information from future observations of the system response.

Postflight Smoothing Algorithm. Assuming that observations of the system response are available for data frames \( k = 0, 1, \ldots, N \), the following sequence of computations are required for the smoothing algorithm:

Kalman filter loop, performed for \( k = 0, 1, \ldots, N \)

\[
\begin{align*}
\hat{X}_{k+1/k} &= \Phi \hat{X}_{k/k} + B \Gamma_{k+1} \\
P_{k+1/k} &= \Phi P_{k/k} \Phi^T + B Q_{k+1} B^T \\
\tilde{Z}_{k+1/k} &= Z_{k+1} - C \hat{X}_{k+1/k} \\
K_{k+1} &= P_{k+1/k} C^T [C P_{k+1/k} C^T + G_{k+1}]^{-1} \\
\hat{X}_{k+1/k+1} &= \hat{X}_{k+1/k} + K_{k+1} \tilde{Z}_{k+1/k} \\
P_{k+1/k+1} &= [I - K_{k+1} C] P_{k+1/k}
\end{align*}
\]

where \( P_{k+1/k} \) is the predicted filter error covariance, \( P_{k+1/k+1} \) is the adjusted filter error covariance based on the current observation, \( K_{k+1} \) is the Kalman filter gain matrix, and \( \tilde{Z}_{k+1/k} \) is the filter innovation. The matrices \( B Q_{k+1} B^T \) and \( G_{k+1} \) are the assumed covariances for the input and measurement noise processes. The vectors \( \hat{X}_{k+1/k} \) and \( \hat{X}_{k+1/k+1} \) are the predicted and corrected estimates of the filter state vector. Resulting estimates for the state vector, the error covariances, and the innovations process must be stored for each recursion, \( k = 0, 1, \ldots, N \).

At this point the input residual estimation loop is performed backwards for \( k = N, N-1, \ldots, 0 \), using stored data

\[
\begin{align*}
R_{k/N} &= \Delta_k^T R_{k+1/N} + C^T [C P_{k-1/N} C^T + G_k]^{-1} \tilde{Z}_{k-1} \\
\delta_{k/N} &= Q_k B^T R_{k/N}
\end{align*}
\]

where \( R_{k/N} \) is the deconvolution residual vector, defined by

\[
P_k R_{k/N} = \hat{X}_{k/N} - \hat{X}_{k-1}
\]

and

\[
\Delta_k^T = (I - K_k C) \Phi^T = [\Phi (I - K_k C)]^T
\]

Finally, the input estimate is computed as

\[
\hat{U}_{k/N} = \Gamma_{k/N} + \delta_{k/N}
\]

The final value for the residual vector, \( R_{N+1/N} \), is given by equation (11) with \( k \) replaced by \( N \)

\[
R_{N+1/N} = [P_{N+1/N}]^{-1} [\hat{X}_{N+1/N} - \hat{X}_{N+1/N}]
\]

Real-Time Filtering Algorithm. The real-time algorithm relies only on past and present data to perform the estimation. Assuming that observations of the system response are available for data frames \( k = 0, 1, \ldots, k+1 \), the residual estimation is performed with \( k+1 \) replacing \( N \)

\[
\begin{align*}
\delta_{k+1/k+1} &= Q_{k+1} B^T R_{k+1/k} \\
\hat{U}_{k+1/k+1} &= Q_{k+1} B^T [P_{k+1/k}]^{-1} \hat{X}_{k+1/k+1} - \hat{X}_{k+1/k}
\end{align*}
\]

and substituting in the Kalman filter

\[
\delta_{k+1/k+1} = Q_{k+1} B^T [P_{k+1/k}]^{-1} K_{k+1} \tilde{Z}_{k+1/k}
\]

The input may be estimated as a part of the Kalman filtering loop according to the sequence of computations:

Kalman filter step performed for \( k = 0, 1, \ldots, N \)

\[
\begin{align*}
\hat{X}_{k+1/k} &= \Phi \hat{X}_{k/k} + B \Gamma_{k+1} \\
P_{k+1/k} &= \Phi P_{k/k} \Phi^T + B Q_{k+1} B^T \\
\tilde{Z}_{k+1/k} &= Z_{k+1} - C \hat{X}_{k+1/k} \\
K_{k+1} &= P_{k+1/k} C^T [C P_{k+1/k} C^T + G_{k+1}]^{-1} \\
\hat{X}_{k+1/k+1} &= \hat{X}_{k+1/k} + K_{k+1} \tilde{Z}_{k+1/k} \\
P_{k+1/k+1} &= [I - K_{k+1} C] P_{k+1/k}
\end{align*}
\]

Residual estimation step

\[
\delta_{k+1/k+1} = Q_{k+1} B^T C^T [C P_{k+1/k} C^T]^{-1} \tilde{Z}_{k+1/k}
\]

and

\[
\hat{U}_{k+1/k+1} = \Gamma_{k+1/k+1} + \delta_{k+1/k+1}
\]

Applying Minimum Variance Deconvolution Algorithms to the Pressure Compensation Problem

The above algorithms will now be applied to the pneumatic distortion problem. Equation (7) may be written as the first-order matrix difference equation

\[
X_{k+1} = \Phi X_k + BU_{k+1}
\]

If the downstream pressure is observed in the presence of noise, then

\[
Z_{k+1} = X_{k+1} + g_{k+1}
\]

If the substitution

\[
U_{k+1} = \Gamma_{k+1} + \delta_{k+1}
\]
is performed, equations (12) and (13) are the same form as
the assumed deconvolution model, equations (8) and (9),
and the deconvolution algorithms may be directly applied.

The input function \( \Gamma_{k+1} \) is determined by solving equa-
tion (7) for the input pressure in terms of the downstream
pressures, and evaluating the result using state estimates
from previous time recursions of the filter loop. Amplified
noise in \( \Gamma_{k+1} \) is estimated and controlled by the deconvolu-
tion algorithm through judicious selection of the state and
measurement noise covariance matrices.

The state noise covariance matrix is derived from equa-
tion (7) as

\[
BQ_{k+1}B^T = \begin{pmatrix}
0 & 0 \\
0 & Q_{U,k+1} \left[ \frac{(\omega_0 \Delta f)^2}{1 + 2Q_{U,k+1}(\omega_0 \Delta f)} \right]^2
\end{pmatrix}
\]

(14)

where \( Q_{U,k+1} \) is the variance in the residual \( \delta_{k+1} \). Similarly,
assuming that the measurement noise is white, the measure-
ment noise covariance matrix is

\[
G_{k+1} = \begin{pmatrix}
G(1)_{k+1} & 0 \\
0 & G(2)_{k+1}
\end{pmatrix}
\]

(15)

where \( G(1)_{k+1} \) is the error variance of the measured
downstream pressure at data frame \( k \) and \( G(2)_{k+1} \) is the error
variance in the measured downstream pressure at data frame
\( k + 1 \). For stationary measurement processes

\[ G(1)_{k+1} = G(2)_{k+1} \equiv G \]

Results of Deconvolution Analysis

Selected results of the minimum variance deconvolution
analysis will now be presented. Filtering tuning will be
demonstrated using simulated data. Next the deconvolution
algorithms will be applied to flight data. Both the postflight
smoothing and real-time filtering algorithms will be ana-
yzed and compared with reference pressure data obtained
from flight experiments.

Filter Tuning

Selecting proper assumed values of the error covariance
matrices is critical. This practice, referred to as “filter tun-
ing,” is typical of minimum variance estimators. The appro-
priate values for the covariance matrices are usually unique
to each configuration. Proper tuning of the deconvolution
algorithms will be shown using simulated data generated
from numerical solutions of the full wave model. Simulated
data is used because the actual measurement and input er-
ror covariances can be predefined. The simulated data were
generated using the full wave model assuming the configu-
ration associated with the overdamped case of Fig. 2; a sen-
sor configuration with pneumatic tubing 8 ft long, and of
0.06-in. diameter, a transducer with negligible volume, and
operating at an altitude of 40,000 ft. For this configuration
and altitude, the equivalent damping ratio of the reduced-
order system is 1.93; the natural frequency is 26.75 Hz. Sim-
ulated zero-mean white noise with a variance of 0.1 \( \frac{\text{lb}_f}{\text{ft}^2} \)
was superimposed on the output time history; simulated
zero-mean white noise with a variance of 1.0 \( \frac{\text{lb}_f}{\text{ft}^2} \)
was superimposed on the input time history. The input and out-
put noise were uncorrelated.

The minimum variance deconvolution analysis was sim-
ulated using the postflight smoothing algorithm. Figure 4
shows the simulation process. A defined input is superim-
posed with additive noise and the result (I) is input to the
wave model. The output from the wave model is superim-
posed with additive noise to give a pseudomeasurement (O)
time history. Using values of the natural frequency and the
equivalent damping ratio provided by the wave model and
prescribed values for the input and observation covariances,
the pseudomeasurement is used to perform the deconvolu-
tion. The resulting deconvolution estimates of the model
input and response are indicated by the symbols \( \tilde{I} \) and \( \tilde{O} \).

Figures 5 and 6 show the effects of selected values of \( G \)
on the estimates of \( U \). In these figures, the value of \( Q_U \) was
fixed at the appropriate values of \( 1.0 \), while the value for \( G \)
was varied. Recall that the actual measurement error vari-
ance is \( 0.1 \frac{\text{lb}_f}{\text{ft}^2} \). Figure 5 shows the deconvolution re-
sults for \( G = 10 \), a value which is much too large. Three
time history graphs are shown. Time history overplots of
the measured downstream pressure response (O) and the
smoother estimate of the input pressure (I) are shown in the
top graph. Time history overplots of \( O \) and the estimated
response (\( \tilde{O} \)) are shown in the middle graph, and time his-
tory overplots of the actual input (I) and \( \tilde{I} \) are shown in the
bottom graph. Comparisons are poor. This set of weights does
not allow the estimation algorithm to extract enough infor-
mation from the measured pressure data; resulting estimates
are overly smoothed.

The above situation cannot be improved by arbitrarily
lowering the value for \( G \). Figure 6 shows similar time his-
tory comparisons for a value of \( G \) which is much too small,
0.01. The end result is a virtually perfect match between
the measured and predicted response. As discussed earlier,
this results in over-amplification of the measurement noise
and the estimate is overwhelmed. The selected value for \( G \)
did not allow the deconvolution algorithm to identify and
control noise in the measured response. The filter is essen-
tially trying to perform an open-loop deconvolution of the
state equation.

Figure 7 shows the effect of selected values of \( Q_U \) on the
estimation of \( U \). The value of \( G \) was fixed at the appropriate
value of \( 0.10 \), while the value for \( Q_U \) was set at \( 10 \). The ac-
tual value for \( Q_U \) is \( 1.0 \). The resulting estimate contains ex-
taneous harmonics that are not present in the original input
signal caused by overamplification of mid-range frequen-
cies. The dynamic constraints of the reduced-order model
have not been enforced strongly enough.
The appropriate level of smoothing and compensation is shown in Fig. 8. Here the proper values for the measurement and input error covariances are used. In this case the estimated input matches the actual input well, and much of the superimposed noise has been removed.

Application to Flight Data

The deconvolution algorithms will now be applied to flight data from the experiments described in Refs. 1–3. The flight test configuration mounted on the wing of a high-performance aircraft (Fig. 9) consisted of a set of static test orifices connected by pneumatic tubing to a single electronically scanned pressure module (ESP), and a single in situ mounted pressure sensor. The test matrix had ports of 0.02, 0.04, and 0.06 in. in diameter and allowed pneumatic tubing sections to be interchangeable. Reference pressure data were provided by a very low volume piezoelectric wafer pressure transducer glued to the aircraft skin adjacent to the test matrix. Since the in situ transducer was mounted on the aircraft skin, it was exposed to temperatures considerably lower than the recommended normal operating range. This caused the direct current (DC) output to drift off the scale. To eliminate the drift, the reference sensor output was electronically coupled with a third-order band-pass filter (1.16 to 101.7 Hz) to remove the DC component. The DC component was restored to the reference data using the output from a pressure port connected to the ESP module by 6 in. of pneumatic tubing. The output from this port was low-pass filtered using a second-order digital filter with a rolloff frequency at 1.0 Hz. The two filtered signals were merged using a complementary filter.

Pressure data obtained from the test matrix measurements were compensated using both the postflight smoothing and real-time filtering algorithms. The results will be compared to the reference data reconstructed using the complementary filter. The flight data deconvolution process (Fig. 10) is similar to the simulation process described earlier. Two flight data cases will be presented; one for an overdamped configuration, and one for an underdamped configuration.

Deconvolution results for the overdamped flight maneuver obtained at 42,000 ft altitude with a sensor configuration with 8 ft of 0.06-in.-diameter pneumatic tubing and negligible transducer volume are shown in Fig. 11 for the smoothing algorithm and Fig. 12 for the filtering algorithm. For this configuration and altitude, the equivalent damping ratio of the reduced-order system is 2.125; the natural frequency is 26.75 Hz. Again, three time history plots are shown. The lower graph compares with . The middle graph compares and . The upper graph compares to . The comparisons are excellent for both the postflight smoothing and real-time filtering algorithms. The estimated input does not contain the high-frequency noise present in the reference data.

Figure 13 shows deconvolution results for an underdamped flight maneuver at 24,000 ft altitude with a sensor configuration having 2 ft of 0.06-in.-diameter pneumatic tubing and negligible transducer volume for the smoothing algorithm. Figure 14 shows results for the filtering algorithm for the same configuration and altitude. For this configuration and altitude, the equivalent damping ratio of the reduced-order system is 0.279; the natural frequency is 107.18 Hz. Both the smoothing and filtering algorithms work well. The smoothing algorithm removes nearly all of the resonant harmonic, approximately 100 Hz, from the measured pressure response. The filtering algorithm removes a sizable portion of the resonance. Both algorithms account for phase lag well.

These results are typical of most flight maneuvers analyzed. Both the postflight smoothing and real-time filtering algorithms performed well. All tuning of the filter covariances was performed offline using preflight information and arbitrary selection. No on-line adaptation of the covariances was performed.

Summary and Concluding Remarks

In this paper, a general numerical technique for obtaining unsteady pressure measurements using conventional pressure sensing technology was developed. A mathematical model, based on the Navier–Stokes equations of momentum and continuity, was reduced to a simple state-variable model. Frequency responses of the full and reduced-order models were shown.

Using a collection of standard results from minimum variance estimation theory, the reduced-order model was incorporated into a deconvolution algorithm which estimates the system input based on the system output. Both postflight smoothing and real-time filtering algorithms were developed. Overamplification of measurement error is controlled by careful selection of the elements of the algorithm state and measurement error covariance matrices.

Performance characteristics were discussed. Filter tuning characteristics were demonstrated using simulated data. Applications of the techniques to flight data were presented. Both the postflight smoothing and real-time filtering algorithms performed well for overdamped and underdamped sensor configurations.

The results of this research are promising for the modeling and compensation techniques developed. These techniques provide a means of predicting and accounting for the effects of pneumatic distortion in pressure sensing devices. They also offer a reliable and cost effective means of measuring unsteady pressure data. These techniques will be useful until new pressure sensing technologies can be developed and proven.

References


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**Fig. 1** Pressure sensing device.
Fig. 2 Frequency response comparison of full and reduced-order models for overdamped sensor configuration.

Fig. 3 Frequency response comparison of full and reduced-order models for underdamped sensor configuration.

Fig. 4 Deconvolution simulation scheme.
Fig. 5 Effect of improperly selected measurement error covariance, $Q_{u} = 1.0$, $Q = 10.0$. 
Fig. 6 Effect of improperly selected measurement error covariance, $Q_U = 1.0, G = 0.01$. 
Fig. 7 Effect of improperly selected input error covariance, $Q_U = 10.0$, $G = 0.1$. 
Fig. 8 Effect of properly selected input and measurement error covariance, $Q_U = 1.0$, $G = 0.1$. 
Fig. 9 Flight test experiment.

Fig. 10 Flight data deconvolution algorithm.
Fig. 11 Flight maneuver time history comparison for the postflight deconvolution algorithm output for an overdamped system.
Fig. 12 Flight maneuver time history comparison for the real-time deconvolution algorithm output for an overdamped system.
Fig. 13 Flight maneuver time history comparison for the postflight deconvolution algorithm for an underdamped system.
Fig. 14 Flight maneuver time history comparison for the real-time deconvolution algorithm for an underdamped system.
### Abstract

In this paper a technique of compensating for pneumatic distortion in pressure sensing devices is developed and verified. This compensation allows conventional pressure sensing technology to obtain improved unsteady pressure measurements. Pressure distortion caused by frictional attenuation and pneumatic resonance within the sensing system makes obtaining unsteady pressure measurements by conventional sensors difficult. Most distortion occurs within the pneumatic tubing which transmits pressure impulses from the aircraft’s surface to the measurement transducer. To avoid pneumatic distortion, experiment designers mount the pressure sensor at the surface of the aircraft, (called in situ mounting). In situ transducers cannot always fit in the available space and sometimes pneumatic tubing must be run from the aircraft’s surface to the pressure transducer. Research began at the National Aeronautics and Space Administration Ames Research Center Dryden Flight Research Facility, supported by the University of California, Los Angeles, to develop a technique to measure unsteady pressure data using conventional pressure sensing technology. A pneumatic distortion model is reduced to a low-order, state-variable model retaining most of the dynamic characteristics of the full model. The reduced-order model is coupled with results from minimum variance estimation theory to develop an algorithm to compensate for the effects of pneumatic distortion. Both postflight and real-time algorithms are developed and evaluated using simulated and flight data.

### Key Words
- Lag compensation
- Pneumatic distortion
- Pneumatic lag
- Pressure measurements
- Pressure sensors

### Distribution Statement
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