METHODS STUDIES ON SYSTEM IDENTIFICATION
FROM TRANSIENT ROTOR TESTS

Part I of Second Yearly Report under Contract NAS2-7613

Prepared for the Ames Directorate, USAAMRD/L
at Ames Research Center, Moffett Field, California

by

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CHAPTER 1

INTRODUCTION TO AIRCRAFT PARAMETER IDENTIFICATION METHODS

K. H. Hohenemser and D. Banerjee

ABSTRACT

Some of the more important methods are discussed that have been used or proposed for aircraft parameter identification. The methods are classified into two groups: Equation error or regression estimates and Bayesian estimates and their derivatives that are based on probabilistic concepts. In both of these two groups the cost function can be optimized either globally over the entire time span of the transient, or sequentially, leading to the formulation of optimum filters. Identifiability problems and the validation of the estimates are briefly outlined, and applications to lifting rotors are discussed.

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Preface to Second Yearly Report under Contract NAS2-7613

Work under Contract NAS2-7613 started on July 1, 1973 as a continuation of research conducted under Contract NAS2-4151 since February 1, 1967. The research goals stated in Contract NAS2-7613 are
(a) Assess analytically the effects of fuselage motions on stability and random response. The problem is to develop an adequate but not overly complex flight dynamics analytical model and to study the effects of structural and electronic feedback, particularly for hingeless rotors.
(b) Study by computer and hardware experiments the feasibility of adequate perturbation models from non-linear trim conditions. The problem is to extract an adequate linear perturbation model for the purpose of stability and random motion studies. The extraction is to be performed on the basis of transient responses obtained either by computed time histories or by model tests.
(c) Extend the experimental methods to assess rotor wake-blade interactions by using a 4-bladed rotor model with the capability of progressing and regressing blade pitch excitation (cyclic pitch
stirring), by using a 4-bladed rotor model with hub tilt stirring, and by testing rotor models in sinusoidal up or side flow.

Work on research goal (a) has been reported in Part I of the First Yearly Report under subject contract titled "Methods Studies Toward Simplified Rotor-Body Dynamics". The results were published in the paper: Hohenemser, K. H. and Yin, S. K., "On the Use of First Order Rotor Dynamics in Multiblade Coordinates" presented at the 30th Annual National Forum of the American Helicopter Society, May 1974, Preprint 831.

Initial work on research goal (b) has been reported in Part II of the First Yearly Report under subject contract titled "Computer Experiments in Preparation of System-Identification from Transient Rotor Model Tests".

Initial work on research goal (c) has been reported in Part III of the First Yearly Report under subject contract titled "Experiments with a Four-Bladed Cyclic Pitch Stirring Model Rotor."

The second Yearly Report under Contract NAS2-7613 is subdivided into two parts, whereby Parts I and II are related to the research goals (b) and (c) respectively. The authors and titles of the two parts are:

Part II  Hohenemser, K. H. and Crews, S. T., "Additional Experiments with a Four-Bladed Cyclic Pitch Stirring Model Rotor."

Part I begins with an introduction to aircraft/identification methods, it contains the results of computer experiments with several selected methods of system identification applied to lifting rotors, and contains the development of a new method for optimal data utilization.

Part II presents extended frequency response measurements with the four-bladed model rotor including dynamic wake measurements at zero advance ratio. It describes the modifications of the model and its instrumentation for transient pitch stirring tests, and it discusses the development of software for the data processing.
IV

Scope of Part I of Second Yearly Report

The computer experiments presented in the First Yearly Report under Contract NAS2-7613 assumed a rotor condition with .4 advance ratio. The input transients consisted of rectangular and wave shaped normal flow impulses. Noise polluted blade flapping responses represented the system output. A linear identification scheme was applied that worked well if data preprocessing by a digital filter and by a Kalman filter was adopted. Blade Lock number and collective pitch angle were assumed to be not well known and were identified. During the past year this work was extended in several directions:

1. The normal flow transients were replaced by constant acceleration pitch stirring transients, since the wind tunnel model is particularly suited to accept such transients.

2. For zero advance ratio a two parameter dynamic wake model was adopted that included a time constant. A single blade representation is now no more applicable. A multiblade identification analysis was developed.

3. The previously used identification method was modified and extended to an iterative equation error estimation with updated Kalman filter.
4. The maximum likelihood method was applied to both the single blade and the multiblade identification.

5. The maximum likelihood scheme, assuming the absence of system noise, was used to obtain optimum data utilization, that is to avoid inaccuracies of the parameter estimates from insufficient data processing, and to avoid excessive data processing for a given desired accuracy of the parameter estimates.

Since parameter identification practice is a still developing and somewhat controversial discipline, it was believed of interest to provide a brief introduction to aircraft parameter identification methods in Chapter 1 of this report. Chapter 2 describes the extensions of the work with respect to items 1 to 4, while Chapter 3 describes the work performed with respect to item 5. The 3 chapters are written as independent self sufficient reports with separate abstracts and lists of references for each.
The application of methods for system parameter identification from transients is a rapidly growing field of study. There is a confusing multiplicity of methods and of names for these methods, and contradictory claims are made as to the efficiency and usefulness of the various methods. Widely varying parameter identification methods are being used at the various aircraft flight test centers. There also is a substantial difference between what is theoretically possible and what is practically desirable to perform a given job with adequate accuracy and with a reasonable computer effort. Since a comprehensive review of aircraft parameter identification methods is not available in the literature, it was believed useful to provide a brief introduction to this field in Chapter I before presenting the applications to lifting rotors in Chapters II and III. The review of identification methods is by no means complete. Only the most important methods are discussed. Only rough outlines are given for the various methods. Details of the derivations and of the application algorithms are found in the cited literature.

ELEMENTS OF SYSTEM IDENTIFICATION FROM TRANSIENTS

System identification is the process of extracting numerical values for system parameters and other subsidiary parameters (process and measurement noise
covariances, bias, initial states, etc.) from the time history of control or other inputs and of the resulting system responses. A schematic for the measurements is shown in Fig. 1. The process of system identification involves four steps:

1. Selection of a suitable input that insures participation of all important modes of the system in the transient response.

2. Selection of sufficiently complete and accurate instrumentation to measure the key input and output variables.

3. Selection of a mathematical model that adequately represents the actual system characteristics.

4. Selection of an efficient criterion function and estimation algorithm for the identification of the unknown system parameters.

The concept of system identification is illustrated in Fig. 2. The design input is fed both to the actual system and to its mathematical model that contains the unknown parameters. The measured response, polluted by measurement noise, is compared with the computed response from the mathematical model. The difference between these two responses, the response error, is used in the parameter estimation technique based on the
Figure 1 Schematic of Measurements for System Identification

Figure 2 Illustration of System Identification
criterion function and optimizing technique. The estimation algorithm may also use apriori information, e.g., initial statistics of the parameters. Here we will be mainly concerned with the fourth of the previously listed steps, that is with the various estimation algorithms.

The mathematical representation of the system will be given in the non-linear case by:

System equation \[ \dot{x}(t) = f(x,u,t) + \Gamma(t)w(t) \] (1)
Initial condition \[ x(t = 0) = x_0 \]
Measurement Equation \[ y(t) = h(x,u,t) + v(t) \] (2)

where \( x(t) \) n x 1 state vector
\( u(t) \) p x 1 input vector
\( w(t) \) q x 1 system noise vector, covariance \( Q \)
\( y(t) \) r x 1 output or measurement vector
\( v(t) \) r x 1 measurement noise vector, covariance \( R \)

If the system is linear, equations 1 and 2 reduce to

\[ \dot{x}(t) = F(t) x(t) + G(t) u(t) + \Gamma(t) w(t) \] (3)
\[ y(t) = H(t) x(t) + D(t) u(t) + g + v(t) \] (4)

where \( g \) is the vector of bias errors.
CLASSIFICATION OF IDENTIFICATION ALGORITHMS

The various estimation algorithms can be classified into two groups presented in Table 1. The first group listed in Table 1 above the double line is based on statistical regression and does not admit a probabilistic interpretation. The algorithms listed in Table 1 below the double line are based on probabilistic interpretation. In the equation error estimate no measurement noise is modeled, the following 4 methods include both measurement and system noise, while in the output error estimate no system noise is modeled. The various algorithms listed in Table 1 will be discussed in the following sections.

EQUATION ERROR ESTIMATES

Equation error methods assume a performance criterion that minimizes the square of the equation error (process noise). They are least squares techniques and they require the knowledge of all response variables (states) and their derivatives. In the so called least squares method the unknown parameters are selected such that the integral over the square of the state equation error is minimized, see for example reference 1. With equation 1 we have the error function (the upper integration limit T is the time over which the measurements are taken).
<table>
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<th>Criterion for Estimate</th>
<th>Solution for Estimate</th>
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<tr>
<td>Equation Error Estimate</td>
<td>$\hat{\theta}$ minimizes equation error squares and/or integrated equation error squares</td>
<td>Closed form global or sequential optimization by optimum filter (Method function $\delta(t-t_i)$ or $e^{-st}$).</td>
</tr>
<tr>
<td>(no measurement noise is modeled)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>Bayesian Estimate</td>
<td>$\hat{\theta} = E(\theta/Z)$</td>
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<tr>
<td>Quasi-Bayesian Estimate with augmented state</td>
<td>$\hat{\theta}, \hat{x}$ such that $\max f(\theta, x/Z) = f(\hat{\theta}, \hat{x}/Z)$</td>
<td>Sequential optimization by optimum non-linear filter (extended Kalman filter with or without local iteration and/or smoothing).</td>
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<tr>
<td>Quasi-Bayesian Estimate</td>
<td>$\hat{\theta}$ such that $\max f(\theta/Z) = f(\hat{\theta}/Z)$</td>
<td>Iterative global optimization with state Kalman filter equations as constraints.</td>
</tr>
<tr>
<td>Maximum Likelihood Estimate</td>
<td>$\hat{\theta}$ such that $\max f(Z/\theta) = f(Z/\hat{\theta})$</td>
<td></td>
</tr>
<tr>
<td>Output Error Estimate</td>
<td>$\hat{\theta}$ minimizes output error squares.</td>
<td>Iterative global optimization with system equations as constraints.</td>
</tr>
<tr>
<td>(no system noise is modeled)</td>
<td></td>
<td></td>
</tr>
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Table 1 Classification of Estimation Algorithms.
where $\Theta$ is the $m \times 1$ vector of unknown parameters. $W$ is a positive definite weighting matrix. An appropriate choice for $W$ would be $Q^{-1}$ where $Q$ is the covariance of the process noise. For the usual digital data processing, the variables $\dot{x}$, $x$, $u$ are sampled, and only available at discrete time points $t_i$. Mathematically the sampling process can be expressed by multiplying the system equation with the delta function $\delta(t - t_i)$. The integral of equation 5 then becomes a sum. One can use instead of the delta function also a different "method function", for example $e^{-st}$, that would allow taking the Laplace transforms.

If the system is linear in the unknown parameters $\Theta$, the system equation can be written in the form

$$\dot{x}(t) = F(x,u,t)\Theta + \Gamma(t) w(t)$$

and the performance criterion (5) becomes

$$J = \int_{0}^{T} \left[ \dot{x} - F(x,u,t)\Theta \right]^T W \left[ \dot{x} - F(x,u,t)\Theta \right] dt$$

Since the function inside the integral has continuous derivatives with respect to $\Theta$ we set

$$\frac{\partial J}{\partial \Theta} = 0$$
thus resulting in the closed form solution

\[
\hat{\theta} = \left[ \int_0^T F^T(x,u,t)W F(x,u,t)dt \right]^{-1} \int_0^T F^T(x,u,t)W \dot{x}(t)dt
\]

(9)

The first factor is the covariance matrix of the estimate. If the system is non-linear in the unknown parameters, the solution equation 9 can be replaced by an iterative solution where \( F(x,u,t) \) is substituted by \( \partial f(x,u,\theta_k,t)/\partial \theta \) and \( \hat{\theta} \) on the left hand side is replaced by \( \hat{\theta}_{k+1} \). It can be shown (for example reference 1) that the parameters in the \( n \)th row of \( f(x,u,\theta,t) \) are independent of all the elements of \( \dot{x}(t) \) except \( x_n(t) \). This independence is one of the drawbacks of the least squares method, in that only one of the measured state derivatives is used in determining a given row of the \( f(x,u,\theta,t) \) matrix. If one of the signals has not been measured, the least squares method does not provide an estimate of the parameters related to that signal. This independence also illustrates the fact that the estimate of one row of the \( f(x,u,\theta,t) \) matrix is obtained independently of the other rows, and no "trade-off" can be made between elements in different rows to improve the estimate.

For some applications it is practical to include the state vectors in the error minimization. In the
modified least squares method a combination of the standard least squares with the integrated least squares is used. The parameters obtained by this method not only trace the derivative of the state but also the state itself over the selected time interval. The performance criterion now includes in addition to the equation error also the integrated equation error:

\[
J = \int_0^T |\dot{x}(t) - F(x,u,t)\dot{\theta} + \int_0^t \dot{\theta}(\tau)d\tau - \int_0^T F(x,u,\tau)\Theta d\tau|^2_W dt \tag{10}
\]

where \( W \) is a positive definite weighting matrix and where

\[
||A||_W^2 = A^T W A \tag{11}
\]

Minimizing the expression, equation 10 results in the estimate

\[
\hat{\theta} = \left[ \int_0^T \{F(x,u,t) + \int_0^t F(x,u,\tau) d\tau\}^T W \{F(x,u,\tau) + \int_0^t F(x,u,\tau) d\tau\} dt \right]^{-1}
\]

\[
\left[ \int_0^T \{F(x,u,t) + \int_0^t F(x,u,\tau) d\tau\}^T W \{F(x,u,t) + \int_0^t \dot{x}(\tau) d\tau\} dt \right]
\]

REPRODUCIBILITY OF THE ORIGINAL PAGE IS POOR
This method has the same row independence of \( f(x,u,\theta,t) \) as the standard least squares method.

Since these methods do not allow for measurement errors, they result in biased estimates when this type of error does exist. When measurement errors are small, as is increasingly the case in modern instrumentation, the equation error method is preferrable over other methods because of its simplicity. It is widely used also when measurement errors are substantial and then serves as start-up technique for the output error and other iterative methods.

In many applications, measurements of some of the responses or their derivatives are not available. If the response, but not the rate of response is measured, it is tempting to differentiate the measured response. However, the differentiation of measured data introduces additional uncertainty so that this technique is usually inaccurate. If \( e^{-st} \) is used as a methods function, Laplace transforms can be used. The estimation then reduces to an algebraic manipulation of the data that avoids their differentiation. The Laplace transform technique as a substitute of differentiating measurement data is discussed in reference 2.
In the preceding methods we specified a cost criterion $J$ that represented the "loss" resulting from an incorrect estimation of the unknown system parameters. The parameters were then selected in such a way as to minimize the loss. If a priori probabilities exist not only for the measurement errors but also for the unknown parameter vector $\theta$ then one can define an expected loss and select the parameter vector in such a way as to minimize this expected loss. Such an estimate is called a Bayesian estimate, see for example reference 3.

The form of a Bayesian estimate depends on the form of both the loss function and of the a priori distribution of the measurement and parameter noise. For the particular case of positive semi-definite quadratic loss functions the Bayesian estimate is the mean of $\theta$ conditioned on the observations. This is true regardless of the distribution of measurement and parameter noise. It has also been shown that for the case of unimodal symmetric a posteriori distribution of the parameters given the observations, the Bayesian estimate is the conditional mean for a considerably wider class of loss functions. For these reasons the Bayesian estimate can be defined as the conditional mean of the parameter distribution.
In order to compute the conditional mean it is first necessary to determine the conditional probability density for \( \theta \). This density can be written from Bayes rule as (\( Z \) is the set of all observations)

\[
 f(0/Z) = f(Z/0) f(0)/f(Z) \tag{13}
\]

The denominator is a normalizing factor determined from

\[
 f(Z) = \int_{\text{all } \theta} f(Z/\theta) f(\theta) \, d\theta \tag{14}
\]

The optimal Bayesian estimate is now given by

\[
 \hat{\theta} = \left(1/f(Z) \right) \int_{\text{all } \theta} f(Z/\theta) f(\theta) \, d\theta \tag{15}
\]

In general the evaluation of equations 14 and 15 would require the solution of the system equations for all possible values of the parameter vector \( \theta \). This is a large effort, especially if the dimension of \( \theta \) is large. If \( f(\theta/Z) \) is unimodal and symmetric about its mean value, the conditional mean corresponds to the mode. Since \( f(Z) \) is merely a scale factor the finding of the mode requires neither the evaluation of the integral in equation 14 nor that in equation 15. The mode \( \tilde{\theta} \) of \( \theta \) has the property

\[
 \max_{\theta} f(\theta/Z) = \max_{\theta} f(Z/\theta) f(\theta) = f(Z/\tilde{\theta}) f(\tilde{\theta}) \tag{16}
\]
Even if the a priori density \( f(\theta) \) is symmetric it does not follow that the conditional density \( f(\theta|Z) \) is also symmetric since in general the observations depend non-linearly on the parameters. Estimation according to equation 16 is, therefore, called "quasi-Bayesian" estimation. Another designation used for example in reference 4 is maximum a posteriori probability (MAP) parameter estimate. Since the logarithm is a monotonic function of its argument we can replace equation 16 by maximizing the expression

\[
J = \log f(Z|\theta) + \log f(\theta)
\]  

(17)

If no a priori information about the parameters \( \theta \) is available, that is, if the a priori density is uniform, \( f(\theta) = \) constant, the quasi-Bayesian estimate reduces to the "maximum likelihood" estimate which involves finding the maximum of \( f(Z|\theta) \).

**ESTIMATES ASSUMING GAUSSIAN DISTRIBUTIONS**

The evaluation of equation 17 becomes particularly convenient if we assume Gaussian densities for the parameters, for the observations and for the system states. In linear systems and linear measurement equations (equations 3 and 4) one needs only to assume that the system noise \( w(t) \) and the measurement noise \( v(t) \) is Gaussian. It then follows that states \( x(t) \) and
observations \( y(t) \) are also Gaussian. For non-linear systems with Gaussian noise \( f(Z/\theta) \) tends to a Gaussian density as the sampling rate is increased (see for example reference 5 p. 29). The assumption of Gaussian densities for all variables is, therefore, a reasonable one. Since \( \theta \) is a \( m \times 1 \) vector we now have the a priori density

\[
f(\theta) = \left| P_{\theta} \right|^{-1/2} (2\pi)^{-m/2} \exp -\frac{1}{2} (\theta - \theta_0)^T P_{\theta}^{-1} (\theta - \theta_0) \tag{18}\]

\( P_{\theta} \) is the parameter covariance matrix, \( \theta_0 \) is the parameter mean. Except for a constant additive term, \( \log f(\theta) \) is now given by

\[
\log f(\theta) = \frac{1}{2} (\theta - \theta_0)^T P_{\theta}^{-1} (\theta - \theta_0) \tag{19}\]

In order to obtain an expression for \( \log f(Z/\theta) \) in equation (17), we assume that \( Z \) consists of \( N \) consecutive observations \( y(1) ... y(N) \).

\[
Z = Y_N = \{y(1), \ldots, y(N)\} \tag{20}\]

With successive application of Bayes rule we obtain

\[
f(Y_N/\theta) = f(y(1), \ldots, y(N)/\theta) = f(y(N)/Y_{N-1}, \theta) f(Y_{N-1}/\theta) = \prod_{j=1}^{N} f(y(j)/Y_{j-1}, \theta) \tag{21}\]

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Taking the logarithm we have

\[ \log f(Y_N/\theta) = \sum_{j=1}^{N} \log \frac{f(y(j)/Y_{j-1}, \theta)}{Y_{j-1}} \]  (22)

\( (y(j)/Y_{j-1}, \theta) \) is the observation estimate at time \( j \) given all preceding observations and given the parameters.

We denote the observation by \( y(j) \) and its expected value and covariance respectively by \( \hat{y}(j/j-1) \) and \( B(j/j-1) \).

We further denote the "innovation" by

\[ y(j) - \hat{y}(j/j-1) = v(j) \]  (23)

Since \( y(j) \) is a \( r \times 1 \) observation vector, its Gaussian density is

\[ f(y(j)) = \left| B(j/j-1) \right|^{-1/2} (2\pi)^{-r/2} \exp \left\{ -\frac{1}{2} v^T(j) \right\} \left\{ \frac{1}{2} v(j) \right\} \]  (24)

Taking the logarithm of equation 24, summing according to equation 22, inserting in equation 17 and inverting the sign we have now to minimize the expression (see also equation 19)

\[ \sum_{j=1}^{N} \{ v^T(j) B^{-1}(j/j-1) v(j) + \log |B(j/j-1)| \} \]

\[ + (\theta - \bar{\theta})^T P_\theta^{-1} (\theta - \bar{\theta}) \]  (25)
If no a priori information is available before taking observations, the last term in the expression 25 is constant and we then have the criterion for the maximum likelihood estimation. Bayesian or quasi-Bayesian estimation is rarely used since a priori densities for the parameters are in most applications not available.

**MAXIMUM LIKELIHOOD PARAMETER ESTIMATION**

According to expression 25 maximum likelihood estimation is equivalent to minimizing the so-called likelihood function

$$J(\theta) = \sum_{j=1}^{N} [v^T(j)B^{-1}(j/j-1)v(j) + \log |B(j/j-1)|] \quad (26)$$

In the presence of system noise the minimization of the expression 26 is very difficult. When going from time $j-1$ to time $j$ one first has to solve the prediction equations for the estimate of the state and for its covariance. Assuming the linear system equation 3 with zero mean Gaussian system noise $w(t)$ the prediction is given by

$$\hat{x}(j/j-1) = F \hat{x}(j/j-1) + G u(t), \ (j-1) \leq t \leq j \quad (27)$$

$$\dot{P}(j/j-1) = F P(j/j-1) + P(j/j-1)F^T + \Gamma Q \Gamma^T \quad (28)$$
where $Q$ is the system noise covariance and $P$ the state covariance. These equations use the estimated state and its covariance at time $j-1$: $\hat{x}(j-1/j-1)$ and $P(j-1/j-1)$, to predict the state and its covariance at time $j$: $\hat{x}(j/j-1)$ and $P(j/j-1)$. This is the prediction before we know the result of the observations at time $j$. After the observations $y(j)$ have been made the optimum estimate is given by the Kalman filter equations for the state and for its covariance:

$$
\hat{x}(j/j) = \hat{x}(j/j-1) + K(j) (y(j) - H \hat{x}(j/j-1))
$$

(29)

$$
P(j/j) = (I - K(j) H) P(j/j-1)
$$

(30)

with the filter gain

$$
K(j) = P(j/j-1) H^T (H P(j/j-1) H^T + R)^{-1}
$$

(31)

where $R$ is the measurement noise covariance. The covariance of the observations $B(j/j-1)$ that occurs in the cost functional 26 is given in terms of the state covariance before observations by

$$
B(j/j-1) = H P(j/j-1) H^T + R
$$

(32)

Thus the terms in the expression 26 that is to be minimized require the solution of the prediction equations 27 and 28 for each time interval and of the up-date
equations 29 and 30 at each sampling time together with
the solution of the measurement equation 4. Reference
5 gives an algorithm for the solution of the problem.
However, due to its complexity this algorithm has not
as yet been applied to a practical problem of aircraft
parameter estimation, see for example reference 6.

The problem of minimizing the expression 26 is
greatly simplified if the observation covariance
$B(j/j-1)$ can be assumed constant. This is for example
true for zero system noise, when according to
equation 32 $B(j/j-1) = R$. The problem then reduces
to minimizing the cost function

$$\sum_{j=1}^{N} v^T(j) R^{-1} v(j)$$

where $v(j)$ is given by the innovation term 23. Since
equation 33 represents (according to equations 2 and 4)
the sum of the measurement error squares, the estimation
with equation 33 is also called output error method of
estimation. There are several algorithms available to
perform the optimization of $J(\theta)$ from equation 33. The
most widely used is the modified Newton-Raphson or
quasilinearization method. It has the advantage that
the sensitivity or information matrix is obtained as a
byproduct. The inverted information matrix gives the
Cramer-Rao lower bound for the parameter covariance. This lower bound is found in many applications to be a more meaningful measure of the accuracy of the parameter estimate than the parameter covariance obtained from the equation error method (first factor in equation 9).

**OUTPUT ERROR METHOD USING QUASILINEARIZATION**

We use an iterative method beginning with an initial parameter estimate \( \hat{\theta} = \theta_0 \). The problem is to find a zero of the gradient of the cost function 33, \( \partial J/\partial \theta = 0 \). Consider a two-term Taylor series expansion of \( \partial J/\partial \theta \) about the kth value of \( \theta \)

\[
(\partial J/\partial \theta)_{k+1} \approx (\partial J/\partial \theta)_k + (\partial^2 J/\partial \theta^2)_k \Delta \theta_{k+1} \tag{34}
\]

where

\[
\Delta \theta_{k+1} = \theta_{k+1} - \theta_k
\]

\( (\partial^2 J/\partial \theta^2)_k \) is the second gradient of the cost function with respect to \( \theta \) at the kth iteration. If equation 34 is a sufficiently close approximation, the change in \( \theta \) for the \((k+1)\)th iteration to make \( (\partial J/\partial \theta)_{k+1} \) approximately zero is

\[
\Delta \theta_{k+1} = - [(\partial^2 J/\partial \theta^2)_k]^{-1} (\partial J/\partial \theta)_k \tag{35}
\]

Using now for \( v(j) \) the two-term expansion

\[
v(j)_k \approx v(j)_{k-1} + \frac{\partial}{\partial \theta} (v(j))_k \Delta \theta_k \tag{36}
\]
one obtains for the first and second gradients of the cost function

\[
\left( \frac{\partial J}{\partial \theta} \right)_k = 2 \sum_{j=1}^{N} \frac{3}{\partial \theta} \left[ (v(j))_k^T R^{-1} [v(j)]_k \right]
\]

\[
M = \left( \frac{\partial^2 J}{\partial \theta^2} \right)_k = 2 \sum_{j=1}^{N} \frac{3}{\partial \theta} \left[ (v(j))_k^T R^{-1} \frac{3}{\partial \theta} [v(j)]_k \right]
\]

We thus need solutions for \( v(j)_k \) and \( \frac{3}{\partial \theta} v(j)_k \). For this purpose we first solve the system and measurement equations

\[
\hat{x} = f(x', u, t)
\]
\[
\hat{y} = h(\hat{x}, u, t)
\]

for each iteration whereby the initial conditions are either obtained from the measurements or are included in the unknown parameters \( \theta \). The innovation is now obtained from equation 23. Next we solve the "sensitivity equations" for each iteration

\[
\frac{\partial x}{\partial \theta_i} = \frac{\partial f}{\partial \theta_i} + \frac{\partial f}{\partial x} \frac{\partial x}{\partial \theta_i}
\]

\[
\frac{\partial \hat{y}}{\partial \theta_i} = \frac{\partial h}{\partial x} \frac{\partial x}{\partial \theta_i}
\]
The initial conditions of $\frac{dx}{d\theta}$ are zero except when $x(0)$ is identified as part of the parameters $\theta$. In this case the initial partials have the value one. With equation 23 we can now compute the first and second gradient of the cost function, equations 37 and 38, and then obtain the change in parameters for the next iteration from equation 35. This involves the inversion of the sensitivity matrix $M$ (equation 38), whereby $M^{-1}$ is the Cramér-Rao lower bound for the covariance of the parameters.

The method is easily extended to the case with a priori information on the parameters, equation 25. The sensitivity matrix 38 is then augmented by the term $2P^{-1}_\theta$, and the gradient 37 is augmented by the term $2P^{-1}_\theta (\theta_0 - \theta_{k-1})$, see reference 2.

**PARAMETER ESTIMATION BY FILTERING**

The parameter estimation methods discussed so far can be denoted as "global" methods. The performance criterion includes the test data for the entire duration of the transient. Filtering is an important tool in state and parameter estimation. It can be used either in conjunction with global estimates, or it can be used as a direct approach to state and parameter estimation. An example of the first type of filter applications is the prefiltering of test data before
using them in a least squares regression estimate, see for example reference 4. The Graham digital filter can remove high frequency noise. A Kalman filter can be used to estimate state variables and their rates not directly measured. It also removes the noise in the measurements. The role of the Kalman filter in maximum likelihood estimation has been shown in equations 29 to 31, where it is used to establish the innovation sequence.

In addition to applications in global estimation methods, filters can also be used as substitutes for global methods. The advantage of such direct filter methods is a reduction in computer effort particularly in cases with a large number of parameters. The disadvantage is that unlike the inverted information matrix of the maximum likelihood method that provides a lower bound on the parameter covariances, no physically meaningful parameter covariances are obtained with the direct filter methods. The covariance propagation equations require initial values that are usually impossible to obtain in any rational way. Though improvements of the filter solution (forward time integration) can be achieved by smoothing (backward time integration), the final parameter covariances remain arbitrary, since they evolve from arbitrary initial covariance estimates.
Assuming that all state variables and their rates have been either measured or are otherwise known from manipulating the measurement data, the unknown parameters, if they occur in linear form in the state equation, can be found by application of a linear filter, see for example reference 7. The classical regression method is a special case of this direct filtering method, namely for infinite initial parameter covariances. In classical regression one obtains a single value of the error covariance matrix. The direct filter application allows the use of a finite initial error covariance matrix and it gives the evolution of this matrix as a function of time. One thus obtains an indication when to stop processing the test data after their information contents has been exhausted. As mentioned before, the absolute values of the error covariances are meaningless, since one usually does not have a rational way of establishing initial values for the parameter covariances.

A method that appears to be economic of computer time for large numbers of unknown parameters was used in reference 4 for application to helicopters. The method consists of a simultaneous identification of
states and parameters with the help of a non-linear filter. In other words, the unknown parameters are treated as additional state variables. Since there occur products of state variables and parameters, the filter is a non-linear one. The so called extended Kalman filter appears to be particularly useful for this purpose. Either non-linear filtering alone or linear filtering in combination with smoothing is performed. The absolute values of the parameter covariances are again of no physical significance since they depend on the arbitrary initial values. In the following, a brief discussion of the direct use of filters in parameter estimation is given.

LINEAR FILTER METHOD OF PARAMETER ESTIMATION

The equation error estimate based on the performance criterion of equation 5 was given in global form by equation 9, which is valid provided that the system is linear in the unknown parameters \( \theta \). The global estimate can be replaced by a filter solution.

Consider a system with the process equation

\[ \dot{\theta} = 0 \]  

(41)

and with the measurement equation

\[ \dot{x}(t) = f(x,u,\theta,t) + v(t) \]  

(42)
The extended Kalman filter that minimizes sequentially the performance criterion is given by the filter equation

\[
\dot{x} = P(\alpha f/\alpha x)^T W (x - (\alpha f/\alpha x) \dot{x})
\]  \hspace{1cm} (43)

and by the covariance equation

\[
\dot{P} = -P(\alpha f/\alpha x)^T W (\alpha f/\alpha x) P
\]  \hspace{1cm} (44)

x and \(\ddot{x}\) are here quantities that are measured or that have been reconstituted from measurements by prefiltering. The weighting matrix W can be interpreted as the inverse of the error covariance \(R^{-1}\) for \(\nu(t)\). The sequential estimate from the filter solution gives the optimum based on all preceding data. One can either select an initial estimate \(\theta(0)\) to integrate equation 43, or one can also assume a zero value for the initial estimate. The initial error covariance \(P_\theta(0)\), as mentioned before, is usually not known. One should, therefore, use a rather large value for the initial covariance. Though the resulting values of \(P_\theta(t)\) obtained as function of time by solving equation 44 have no physical meaning as far as absolute values are concerned, \(P_\theta(t)\) will asymptotically approach zero, and it can be used as
a criterion when to stop processing the data. As can be seen from equation 43, no further changes in the parameters $\theta$ will occur, once $P_\theta$ has approached zero.

The covariance equation 44 has the closed form solution (see reference 7)

$$P_\theta(t) = [P^{-1}_\theta(0) + \int_0^t \left(\frac{\partial f}{\partial \theta}\right)^T W \left(\frac{\partial f}{\partial \theta}\right) \, dt]^{-1}$$

(45)

and the filter equation 43 has the closed form solution

$$\hat{\theta}(t) = P_\theta(t) [P^{-1}_\theta(0)\theta(0) + \int_0^t \left(\frac{\partial f}{\partial \theta}\right)^T W \hat{x} \, dt]$$

(46)

Whether it is computationally simpler to numerically integrate equations 43 and 44 or to evaluate equations 45 and 46 depends on how many points in time are to be covered. For infinite initial parameter covariance, $P^{-1}_\theta(0) = 0$, equation 46 becomes identical to equation 9.

**BAYESIAN ESTIMATION AS A FILTERING PROBLEM**

If we extend the quasi-Bayesian or maximum a posteriori probability (MAP) criterion 16 to include both the parameters $\theta$ and the states $x(t)$, we have

$$\max_{x(t), \theta} f(x(t), \theta/Z) = \max_{x(t), \theta} f(Z/x(t), \theta) f(x(t), \theta)$$

(47)
Assuming now the non-linear system and measurement equations 1 and 2, and assuming further that states and parameters have Gaussian distributions of the form of equation 18, the criterion 47 becomes (see for example references 4 and 8) one of minimizing the quadratic function

\[
J = \frac{1}{2} \left[ \| \Theta(0) - \hat{\Theta}(0) \|_P^{-1} + \| \hat{x}(0) - \hat{\Theta}(0) \|_P^{-1} \right] \\
+ \int_0^T \left\{ \| y(t) - h(x,u,t) \|_R^{-1} + \| w(t) \|_Q^{-1} \right\} dt
\]

subject to the constraint equation 1. If the system and measurement equations 1 and 2 are linearized about the current estimates \( \hat{x} \) and \( \hat{\Theta} \) the recursive solution of the minimization problem 48 results in the extended Kalman filter equations given for the continuous case by (see reference 4)

\[
\dot{x} = (\partial f/\partial x)\hat{x} + (\partial f/\partial u)u + (\partial h/\partial x)R^{-1}(y - (\partial h/\partial x)\hat{x}) \\
\dot{\Theta} = P_{\Theta x}(\partial h/\partial x)R^{-1}(y - (\partial h/\partial x)\hat{x}) \\
\dot{p} = (\partial f/\partial x)p + (\partial f/\partial x)^T - (\partial h/\partial x)R^{-1}(\partial h/\partial x)p + Q + (\partial f/\partial \Theta)p_{\Theta x} + [(\partial f/\partial \Theta)p_{\Theta x}]^T \\
\dot{p}_{\Theta x} = P_{\Theta x}(\partial h/\partial x)^T + P(\partial f/\partial \Theta)^T - P_{\Theta x}(\partial h/\partial x)R^{-1}(\partial h/\partial x)p
\]

with \( p' = -p_{\Theta x}(\partial h/\partial x)R^{-1}(\partial h/\partial x)p_{\Theta x}^T \)
P is as before the state covariance, \( P_{0} \) the parameter covariance, and \( P_{0x} \) is the combined covariance where the parameters \( \theta \) are included in an augmented state. \( P' \) is an abbreviation given in the last of equations 49. Even if the original system is linear, the augmented system is non-linear and hence the filtering problem must be solved by a non-linear filtering technique. In reference 4 the raw data are preprocessed by a digital filter and by a Kalman filter that does not use the unknown parameters but merely makes use of the transformation equations from a space-fixed to a body-fixed reference system (Euler equations).

Lebacqz in reference 9 applies basically the same method except for a discrete instead of the continuous filter formulation. He further uses a one stage filtering-smoothing algorithm which has the advantages of reducing the bias due to non-linearities and of making the algorithm less sensitive to initial conditions. Mehra in reference 4 is critical of using an extended Kalman filter for the augmented state including the unknown parameters. His arguments are that the uncertainties in the states are usually much smaller than the uncertainties in the parameters. Therefore the assumption of local linearization about the latest
estimate which are acceptable for state estimation with an extended Kalman filter are generally less valid for parameter estimation. Moreover, the filter for the augmented state assumes knowledge of the a priori parameter covariances which are unknown. As mentioned before, the arbitrary a priori parameter covariance used as initial conditions for a filter that includes parameters as state variables gives unreliable confidence limits on the parameter estimates. An added difficulty of applying a filter to the augmented state is that poor a priori estimates of the parameters make the convergence rate slow or may even cause divergence of the filter solution. Though improvements can be applied to the extended Kalman filter like local smoothing and local iteration and smoothing, the basic shortcomings of this method appear to have been correctly described in reference 4. Unfortunately, the application of the complete algorithm of maximum likelihood identification given in reference 5 is for a large system much more demanding of computer size and time than the filter solution with the augmented state. While aircraft parameter identification with the complete maximum likelihood algorithm of reference 5 has not as yet been accomplished, the method of filtering the augmented state has been applied to several aircraft parameter identification cases, for example in references 4 and 9.
IDENTIFIABILITY PROBLEMS

Identifiability problems can occur no matter what identification algorithm is used. They are related to the initial 3 steps involved in system identification as listed at the beginning of this chapter: the selection of a suitable input, the selection of the instrumentation, and the selection of the mathematical model. A few comments are added here to point out some difficulties that have been encountered due to these three initial steps.

If the input does not adequately excite some of the system modes, the associated parameters cannot be adequately identified. Sometimes it is practical to combine the responses to various types of inputs into a single identification run, see reference 4. While each of the single inputs excites only a limited number of modes, the combination of inputs provides an adequate excitation of all modes required for the estimation of the parameters. Efforts have also been made to design inputs on the basis of certain optimization criteria. More details on this problem are given in Chapter 3 of this report.

If there are large unaccounted for instrumentation errors non-physical parameter values may result. In reference 10 instrumentation lags and control measurement
errors were found to be most significant. Static measurement errors and instrumentation lags can be a much greater source of parameter inaccuracies than white noise. A detailed analysis of the relationship between static and dynamic measurement errors in states and control inputs and the accuracy of the parameter estimates is required.

If the selected mathematical model for the system is inadequate, the parameters are forced to account for some unmodeled effects. The estimated parameters may, therefore, be quite different from those determined by aerodynamic theory or wind tunnel tests would indicate. A good example is given in reference 11 where a six degree of freedom mathematical model for a helicopter gave unrealistic derivatives, since it had to account for effects of some neglected modes. A unique six degree of freedom linear model for the helicopter flight dynamics does actually not exist. When a nine degree of freedom mathematical model is used, these difficulties disappear. Modeling errors are also a major cause for the lack of convergence of iteration procedures or of parameter identification by filtering methods. The best remedy against difficulties from modeling errors is the adoption of a more suitable mathematical model. Some other measures to improve the convergence of iteration procedures or of filtering methods will be briefly discussed.
In the cases where a priori values of parameters, for example from theory or from wind tunnel tests, are available, one can use an a priori weighting matrix that expresses the confidence in these values and prevents the algorithm from deviating too much from the a priori values. Sometimes there exist some relationships between the parameters. These should then be used as constraints in the optimization problem to avoid non-physical parameter estimates. If parameter dependencies exist, difficulties are encountered in inverting the information matrix. An exact dependency between parameters should result in a zero eigenvalue of the information matrix. A rank deficient solution makes use of the fact that in case of near parameter dependencies there is a large spread between a set of small eigenvalues and another set of much larger eigenvalues of the information matrix.

In filter solutions, divergence because of modeling errors can occur when the covariance matrix becomes prematurely too small, thus preventing further test data to be of influence. There are several ways to prevent premature small covariances. One can provide fictitious noise input to the system or one can directly increase the parameter covariance in each time step according to some rule. One can also overweigh the most recent data thus causing the filter to reduce its memory of the data.
of the more distant past. This indirectly increases the parameter covariance matrix. Since too short data length and too large errors in the initial parameter estimates may also result in non-physical parameter values or in divergence of the identification algorithm, longer transients and better a priori parameter estimates can lead to the avoidance of these difficulties.

VALIDATION OF ESTIMATES

Once a set of parameter estimates has been obtained the question arises; what confidence can be associated with this set? As mentioned before, the parameter covariance matrix obtained by filtering the augmented state is not a good measure of this confidence. The inverted information matrix obtained with the maximum likelihood method represents the Cramer-Rao lower bound for the parameter covariances and is a better measure of this confidence. Using the parameter estimates to predict the transients from which the estimates have been obtained, and computing the rms error with respect to the measured transients, gives another confidence measure. However, if the system is inadequately modeled, one may obtain a small rms error despite the fact that the parameter values are wrong in comparison to theoretical or wind tunnel results, see reference 11. A better way of validation is to compare the prediction with the results of test data not used in the identification process. In fact, it is good practice not to use all of the
available test data for the parameter identification but to reserve some of the runs for such a comparison. Sometimes it is desirable to perform the parameter identification not just with one mathematical model but with a variety of models. In the case described in reference 11 a mathematical model with more parameters gave a much better identification result than a model with fewer parameters, better in the sense of an improved correlation with theoretically and wind tunnel generated parameters. However, there are also cases where mathematical models with a larger number of parameters gave worse identification results than a model with fewer parameters, see reference 12.

Adequate parameter estimation from transients requires careful attention to the many contributing factors in the input, instrumentation, mathematical modeling, and the estimation algorithm; and the validation of this process can only be considered complete after the rms errors of the prediction with the estimated parameters as compared to test data have been found acceptably small for all types of possible transient excitations of the system.
APPLICATIONS TO LIFTING ROTORS

Contrary to the fixed wings of airplanes, lifting rotor characteristics are not well approximated by the usual set of aerodynamic derivatives. One reason is blade modes that must be considered particularly in rapid transients. Another reason is the dynamic rotor wake that is produced by the time varying rotor thrust and rotor pitching and rolling moments and that has a feedback effect on the rotor forces and moments. The omission of the blade modes, as shown in reference 11, results in non-unique and non-physical rotorcraft derivatives. The identification is better if separate rotor degrees of freedom are introduced even in the crude form of a first order lag as was done in reference 13.

A variety of identification methods have been used with respect to lifting rotors. After preprocessing the test data with a digital filter followed by a Kalman filter that does not contain the aerodynamic derivatives (transformation or Euler equations), least squares identification is applied to rotorcraft transient flight test data in references 4 and 11. Each identification run is made with several transients simultaneously. The least squares results are then used as start-up values for an extended Kalman filter for the augmented state. It is not obvious that the extended Kalman filter
actually improves on the least squares results, though filter convergence is achieved. In reference 13 the output error method with quasilinearization is applied without preprocessing the flight test data. The flight data of both references 4 and 13 were obtained in calm air. The equation error method in its filter form was applied in reference 7 to simulated noisy blade flapping and torsion measurements at high rotor advance ratio. The simulated data were preprocessed by a Graham digital filter, but not by a Kalman filter. Reference 7 assumed that all states and their derivatives had been measured. In contrast reference 14 assumed that only flapping deflections are measured but not flapping rates or flapping accelerations. For the dynamic wind tunnel tests simulated in reference 14 there is no way of applying a Kalman filter that does not contain the unknown parameters. However, it was found that for the cases studied, a Kalman filter with considerable errors in the unknown parameters was useful in obtaining the non measured flapping rates and accelerations. The parameter identification was then performed by the equation error method in its filter form.

In Chapter 2 of this report the same method (except for using global estimates) is used in an iterative form. In addition, the output error method with quasilinearization is applied to the same and to more complex rotor
identification problems. As will be shown, the first method is more computer cost effective in cases where it works, while the second method is more reliable and more versatile.
REFERENCES


Some textbooks related to the subject matter of this chapter are:


Chapter 2

COMPUTER EXPERIMENTS WITH PITCH STIRRING TRANSIENTS

K. H. Hohenemser and S. K. Yin

ABSTRACT

Selected methods are applied in the form of computer experiments to two problems of lifting rotor state variable and parameter identification. The first problem refers to a rotor condition at .4 advance ratio. Cyclic pitch stirring with constant acceleration is assumed as known input. Noise polluted blade flapping measurements are assumed as the only measured output. Blade Lock number and collective pitch are the unknown parameters to be identified. The second problem refers to a rotor condition at zero advance ratio again with constant acceleration pitch stirring. A two parameter dynamic inflow model is stipulated. The only measurements are noise polluted blade flapping responses. Blade Lock number and the two parameters of the dynamic inflow model (including a time constant) are the unknown parameters to be identified.

Two parameter identification methods are applied and their relative merits evaluated: Iterative equation error estimation with updated Kalman filter and the maximum likelihood method. The latter method, though requiring increased computer effort per iteration, was found to be more versatile.
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INTRODUCTION

The identification method used in reference 1 is based on the experience gained from references 2 and 3 where states and parameters were determined simultaneously with the help of an extended Kalman filter. The filter can easily diverge unless good initial estimates are available. Particularly in reference 3 a considerable effort was applied to obtain such good initial estimates. The test data were first processed with a digital filter that took out high frequency noise without distorting the main signals. The data were then processed with a Kalman filter based on the Euler equations, which do not contain the unknown parameters. Thus measurement bias was removed and missing channels were reconstituted. Finally a least squares algorithm was applied to obtain estimates of the unknown parameters. The subsequent application of the extended Kalman filter led to modified parameter estimates, however it is not clear whether or not these modifications represent improvements. In any case the modifications were not large, and the initial estimates appeared to be satisfactory approximations.

In trying to apply the experience from reference 3 to wind tunnel model transients a difficulty arises,
in that no equivalent to the Euler equations for the aircraft exists. Thus there is no way of using a Kalman filter which is free of the unknown parameters. Instead, if a Kalman filter is to be applied, estimates of the parameters must be inserted. Another difficulty for our wind tunnel model tests is that only flapping deflection measurements are made, while the rates of deflection and the accelerations are not measured. Thus the Kalman filter with the estimated parameters is called upon to provide both rates and accelerations. Finally, the least squares algorithm of reference 3 was replaced in reference 1 by a sequential linear estimator for the parameters. This has the advantage that finite initial parameter covariances can be used, and that the time history of the parameter covariance provides a measure for the time beyond which no more useful information can be extracted from the test data. As far as digital filtering is concerned it was found in reference 1 that a cut-off frequency range from 2.5 to 2.9 removed the high frequency noise without unduly distorting the main signal.

The method used in reference 1 worked well in the computer experiments using normal flow transients at .4 advance ratio for the identification of the Lock
number and of the collective pitch angle. However, a number of questions had to be answered by the subsequent studies. It had been decided to begin wind tunnel model rotor transient testing with constant acceleration pitch stirring inputs, and later follow up with normal flow transient testing. Therefore, it was desirable to perform computer experiments with pitch stirring transients. All of the identification work reported here concerns such transients. The linear sequential estimator used in reference 1 requires the simultaneous integration of the filter and of the covariance differential equations. A simpler "global" estimate requires only the inversion of a system of linear equations for the unknown parameters and the evaluation of a number of integrals over the time period of the transient. Therefore, a number of comparisons were made between these two methods. Finally, the Kalman filter for the test data requires estimates of the unknown parameters. The question arises whether an iterative form of the method is practical, where the identified parameters are reinserted into the Kalman filter and a second identification is performed, etc. This iteration method was tried out in several cases.
While the substitution of the theoretical blade Lock number by an equivalent Lock number can be expected to provide a reasonably good approximation of dynamic rotor wake effects if the transient is relatively slow and does not contain high frequencies, a better dynamic wake representation is given by introduction of the L-matrix (reference 4) together with a time constant (reference 5). For zero advance ratio the L-matrix degenerates into a L-scalar, so that a two parameter dynamic wake description is obtained. A single blade representation is now no more applicable, and a multi-blade analysis is necessary. A number of computer experiments were conducted for this case whereby the blade Lock number was treated as third unknown parameter.

During the last decade the maximum likelihood method of parameter identification has been successfully applied to airplane and helicopter transient testing. This method does not require preprocessing of the test data and also does not need complete measurements of the deflecitons, of their rates and of the accelerations. The parameter covariance estimates obtained with this method are more meaningful than those obtained with the linear sequential estimator used in reference 1. A number of cases were treated with the maximum likelihood method.
In its most general form presented in reference 6 the maximum likelihood method is capable of handling identification problems for cases with both measurement and system noise. The method then becomes computationally quite complex and has not as yet been applied to aircraft transient testing. The usual assumption is that the system is free of noise. In this case the maximum likelihood method is greatly simplified (output error method). It has been applied in this report in the simplified form. If the random system noise is measured, the simplified method is still applicable. However in the planned wind tunnel model testing the system noise will not be measured, and though it is expected to be small, the question is whether or not it will unacceptably degrade the estimates of the parameters. Therefore a few computer experiments were conducted with both simulated measurement noise and with simulated system noise, whereby the maximum likelihood method was applied in its usual simplified form.

**EXCITATION OF PITCH STIRRING TRANSIENTS**

For the wind tunnel experiments with pitch stirring transients the initial state of the rotor will be given by prescribing the advance ratio, the collective pitch angle, the rotor angle of attack (set at approximately
zero) and the cyclic control setting that will be zero longitudinal cyclic and 1.5° lateral cyclic. At the time \( t_0 \) pitch stirring is initiated. Denoting by \( \omega \) the angular pitch stirring speed, positive in the direction of rotor rotation, and by \( \dot{\omega} \) the pitch stirring angular acceleration assumed to be constant, we have

\[
\omega = \dot{\omega}(t - t_0) \tag{1}
\]

For a progressing mode \( \omega \) is negative, for a regressing mode \( \omega \) is positive. In a rotating reference system the blade pitch angle is given by

\[
\theta = \theta_0 + 1.5 \cos [\omega(t-t_0) + t] = 0 \quad \text{for } t \leq t_0
\]

\[
\omega = \begin{cases} 
0 & \text{for } t \leq t_0 \\
\dot{\omega}(t - t_0) & \text{for } t > t_0 
\end{cases} \tag{2}
\]

In a multiblade representation the blade pitch angle of the kth blade is

\[
\theta_k = \theta_0 - \theta_I \sin \psi_k + \theta_{II} \cos \psi_k \tag{3}
\]

where

\[
\theta_I = \begin{cases} 
0 & \text{for } t \leq t_0 \\
1.5 \sin \omega(t-t_0) & \text{for } t > t_0 
\end{cases} \tag{4}
\]

\[
\theta_{II} = \begin{cases} 
1.5 & \text{for } t \leq t_0 \\
1.5 \cos \omega(t-t_0) & \text{for } t > t_0 
\end{cases} \tag{5}
\]
\( \theta_I \) represents forward cyclic pitch, \( \theta_{II} \) represents left cyclic pitch. The wind tunnel experiments will be conducted with a variety of pitch stirring accelerations. The computer experiments were mostly conducted with a pitch stirring acceleration of

\[ \dot{\omega} = -\frac{1}{\pi} \]  \hspace{1cm} (6)

which is in the progressing sense.

Since in the non-dimensional time units used here the time of one rotor revolution is \( 2\pi \), the angular pitch stirring velocity one rotor revolution after initiation of pitch stirring is \( \frac{2}{5} \), that is one fifth of the rotor angular speed. Figure 1 shows the time history of blade pitch for about two rotor revolutions \( (t_0 = 12, t = 12 \text{ to } 24) \) in a rotating frame of reference. Figure 2 shows the time history of blade pitch in multiblade representation, that is \( \theta_I \) and \( \theta_{II} \) vs. time \( t \). Figures 1 and 2 refer to the progressing mode.

A convenient way of conducting the computer experiments is to impose at time \( t = 0 \) a step input of lateral cyclic control. If \( t_0 \) is sufficiently large, the transient from the step control input will have subsided when pitch stirring begins at \( t = t_0 \). It was found that for single blade identification a value of \( t_0 = 12 \) is adequate. In the multiblade identification
including the time delayed rotor wake this value was found to be insufficient. Instead, \( t_\Omega = 70 \) was used. Measurement noise was simulated by polluting the analytical flapping response with zero mean white Gaussian computer generated random sequences. System noise was simulated by adding to the multiblade inflow coordinates in the system equations zero mean white Gaussian sequences. Data processing extended over nearly two rotor revolutions, from \( t_0 \) to \( t_0 + 12 \). In Chapter 3 it will be shown that this choice leads to approximately optimal data utilization. If a shorter time period were used, large errors in the parameter estimates would occur. If a longer time period were used, the additional data processing would be unnecessary in view of the adequate accuracy of the parameter estimate using a time period of two rotor revolutions.

For the wind tunnel experiments a variety of collective pitch settings \( \Theta_0 \) will be tested. For the computer experiments the collective pitch setting was mostly assumed to be \( \Theta_0 = 2^\circ \).

**LINEAR SEQUENTIAL AND GLOBAL ESTIMATORS**

In reference 7 the parameter identification is performed from a "system equation"

\[
\dot{\Theta} = 0
\]  
(7)
and a "measurement equation"

\[ \xi = h(\xi, \theta) + v \]  \hspace{1cm} (8)

Equation 8 is actually the system equation arranged in a form where the left hand side contains all terms that are free of the unknown parameters \( \theta \). If the system equation is linear in the state variables \( \xi \) and in the unknown parameters \( \theta \), \( h(\xi, \theta) \) is a linear function of the parameters. The noise vector \( v \) refers only to the terms on the left hand side of equation 8. The state variables that are multiplied by the unknown parameters in \( h(\xi, \theta) \) must be noise free. To obtain the \( \theta \) both \( \xi \) and \( \xi \) must be known. If only part of the variables in \( \xi \) and \( \xi \) have been measured, Kalman filtering is required in order to reconstitute the missing terms.

Denoting by \( P_\theta \) the parameter covariance matrix and by \( R \) the noise covariance matrix, assuming that \( v \) is zero mean Gaussian white noise, optimal parameter estimates \( \hat{\theta} \) can be obtained by minimizing the cost function

\[
J = \frac{1}{2} \left\{ (\theta(0) - \hat{\theta}(0))^T P^{-1}_\theta (0)(\theta(0) - \hat{\theta}(0)) + \int_0^T (\xi - h(\xi, \theta))^T R^{-1} (\xi - h(\xi, \theta)) \right\} \hspace{1cm} (9)
\]
where the a priori estimates \( \theta(0) \), \( P_\theta(0) \) are assumed to be given together with the noise covariance matrix \( R \). The differential equations associated with this optimal problem are (see for example reference 8)

\[
\dot{\theta} = P_\theta(\partial h/\partial \theta)^T R^{-1}(z - h(\zeta, \hat{\theta})) \tag{10}
\]

\[
\dot{P}_\theta = - P_\theta(\partial h/\partial \theta)^T R^{-1}(\partial h/\partial \theta)P_\theta \tag{11}
\]

Beginning with the initial a priori estimate for the parameters \( \theta(0) \) and their covariance matrix \( P_\theta(0) \), these equations can be integrated and result at each time \( t \) in the optimal parameter estimate given the preceding measurements. Since the initial parameter covariance is usually not known and the assumed values are rather arbitrary, the matrix \( P_\theta \) from integrating equation 11 is not a useful measure of the actual parameter covariance. However, once \( P_\theta \) has approached zero, the effect of any further measurements on the estimate \( \hat{\theta} \) also approaches zero as is evident from equation 10. \( P_\theta \), therefore, is valuable in judging for what length of time the data should be processed. Equations 10 and 11 represent the "linear estimator" used in references 1 and 7.
Instead of the sequential estimation by integrating equations 10 and 11 with some initial estimates $\theta(0)$ and $P_\theta(0)$, one can also obtain a "global" estimate directly from equation 9. If one assumes that one and the same parameter estimate $\hat{\theta}$ is valid throughout the time range from $0$ to $T$, one obtains by setting $\partial J/\partial \theta = 0$

$$\hat{\theta} = [P_\theta^{-1}(0) + \int_0^T \left( \frac{\partial h}{\partial \theta} \right) R^{-1} \left( \frac{\partial h}{\partial \theta} \right) dt]^{-1} \left[ P_\theta^{-1}(0) \theta(0) \right] + \int_0^T \left( \frac{\partial h}{\partial \theta} \right) R^{-1} \xi dt$$

(12)

see for example the appendix of reference 7. A convenient assumption is $P_\theta^{-1}(0) = 0$, which means an infinite initial parameter covariance matrix. The initial estimate $\theta(0)$ is then not required and the evaluation of equation 12 is reduced to the determination of fixed boundary integrals, a matrix inversion and a matrix multiplication. The parameter covariance matrix at the time $T$ is given by the first factor of equation 12:

$$P_\theta(T) = [P_\theta^{-1}(0) + \int_0^T \left( \frac{\partial h}{\partial \theta} \right) R^{-1} \left( \frac{\partial h}{\partial \theta} \right) dt]^{-1}$$

(13)
which follows from the integration of equation 11, (see for example the appendix of reference 7). \( P_\theta(T) \) from equation 13 can again be used to judge whether or not all the significant information contents has been extracted from the data. A comparison between the estimation with equations 10 and 11 and with the "global" method of equation 12 will be given later for a specific example.

**Iterative Equation Error Estimation with Updated Kalman Filter**

When using the parameter estimation methods of the preceding section it is necessary to first determine from the noisy deflection measurements estimates for the deflections, for their rates, and for the accelerations. In reference 1 this was done by passing the noisy deflection data through a digital filter that takes out the noise above a certain frequency without distorting the signal in the low frequency range. The filtered deflections were then either differentiated twice, or a Kalman filter was applied in order to obtain the derivatives. Later studies showed large errors in the parameters for too low cut-off frequency of the digital filter. It was then decided to omit the digital filter and instead use the Kalman filter in an iterative way.
In typical examples already the second iteration was as accurate as the result with the combined digital and Kalman filter given in reference 1.

Only simulated noisy blade flapping measurements were used in the Kalman filter. The filter provided the deflection rates and accelerations needed for the "global" parameter estimate, but not the deflections themselves. In other words, the parameter estimate was performed with the simulated noisy deflection measurements and with the rates and accelerations from the Kalman filter. In the first iteration a Kalman filter with estimated parameter values was used (typically 20% error). After updated parameter values had been obtained, a second pass with an updated Kalman filter was performed, etc. The deflection data remained the same for each iteration, but the rates of deflection and the accelerations were updated. As will be seen in the numerical examples this method worked well for the single blade identification. The reason for the good parameter identification in spite of substantial measurement noise is probably the following. In the equation of single blade flapping the flapping deflection term includes a large part that is independent of the Lock number, and only a relatively
small part that is dependent on the Lock number. Therefore, the major part of the noisy deflection term can be included in the left hand side of equation 8 in ζ. The method allows noise in this part. The smaller part of the flapping deflection term that is of aerodynamic origin and includes the blade Lock number is part of the right hand side of equation 8 and should be noise free. Apparently the actual noise in this relatively small term does not lead to a substantial bias in the parameter estimate.

MAXIMUM LIKELIHOOD ALGORITHM

The maximum likelihood method for our particular case pertains to a system equation (zero system noise)

\[ \dot{x} = f(x,u,\theta) \]  

(14)

where \( x \) is the state vector, \( u \) the input vector assumed to be known, \( \theta \) is the vector of unknown parameters that may include initial values of state variables. The measurement equation is assumed to be linear and of the form

\[ y = H x + v \]  

(15)

\( y \) is the vector of observed quantities, \( H \) is a matrix relating the state variables to the observations, \( v \) is the vector of random measurement errors, assumed to
be zero mean white noise with given covariance matrix \( R \)

\[
E[v(t)v^T(\tau)] = R \delta(t, \tau)
\]  \hspace{1cm} (16)

\( R \) is assumed to be constant with time. Though the preceding equations do not show bias terms, bias errors could easily be included in the unknown parameter vector \( \theta \).

A sample of measurements \( y_1 \ y_2 \cdots y_n \) is now made during the time of the transient and the parameter estimate \( \hat{\theta} \) is selected such that the conditional probability of this sample of measurements given \( \theta \) is maximized.

\[
\hat{\theta} = \max_{\theta} p(y_1 \cdots y_n | \theta)
\]  \hspace{1cm} (17)

The following steps lead to the maximum of the likelihood function \( p(y_1 \cdots y_n | \theta) \), though there is no assurance that the maximum is global. The method outlined here is called quasilinearization with the modified Newton-Raphson method. It assumes Gaussian distributions of the random variables.

1. Select an initial parameter estimate \( \hat{\theta} = \theta_0 \).
2. Solve the system equation 14 with this parameter estimate

\[
\hat{x} = f(\hat{x}, u, \hat{\theta})
\]  \hspace{1cm} (18)
The initial conditions can either be obtained from the measurements, or, where this is not feasible, they can be included in the unknown parameter vector $\hat{\theta}$.

3 Calculate for each measurement the "innovation term"

$$v_j = y_j - H \hat{x}_j$$  \hspace{1cm} (19)

4 Solve the "sensitivity equations"

$$\dot{x}/\partial \theta_k = \partial F/\partial \theta_k + F(t) \partial x/\partial \theta_k$$  \hspace{1cm} (20)

where $$F(t) = \partial F(t)/\partial x \bigg|_{x = \hat{x}}$$

The initial conditions of $\dot{x}/\partial \theta_k$ are zero except when $x(0)$ is identified as part of the parameter vector $\hat{\theta}$. In this case the initial partials have the value one.

5 The likelihood function for zero system noise is

$$J = \log p(y_1 \ldots y_n/\theta) = -(1/2) \sum_{j=1}^{N} v_j^T R^{-1} v_j$$  \hspace{1cm} (21)

Determine now the gradient of this function with respect to $\theta$

$$\partial J/\partial \theta = \sum_{j=1}^{N} v_j^T R^{-1} \partial v_j/\partial \theta$$  \hspace{1cm} (22)
where \( \partial v_j / \partial \theta = - H \Theta \hat{J}_j / \partial \theta \) \hspace{1cm} (23)

6 Compute the information or sensitivity matrix

\[
M = \partial^2 J / \partial \theta^2 = \sum_{j=1}^{N} (\partial v_j / \partial \theta)^T R^{-1} \partial v_j / \partial \theta \hspace{1cm} (24)
\]

The inverse \( M^{-1} \) of the information matrix provides a lower bound for the covariance of the updated parameter estimates.

7 The updated parameter estimate is

\[ \hat{\Theta} = \Theta_0 + \Delta \Theta \hspace{1cm} (25) \]

where \( \Delta \Theta = -M^{-1} (\partial J / \partial \Theta)^T \)

8 Go now back to equation 18 with the updated parameter estimate and repeat the steps to equation 25. Reiterate until convergence of the information matrix and of the parameter vector is obtained.

**SINGLE BLADE PARAMETER IDENTIFICATION**

We first present a case where the sequential estimate is compared to the global estimate of the parameter \( \gamma \) (blade Lock number). The excitation of the progressing mode is given by
\[ \theta = 1.5 \cos (\omega + 1)t \]

\[
\omega = \begin{cases} 
0 & \text{for } t \leq 12 \\
-(1/\pi)(t-12) & \text{for } t > 12 
\end{cases}
\] (26)

This excitation is somewhat different from that defined by equation 2 and illustrated in Figure 1. While equation 2 defines a constantly accelerated progressive excitation, equation 26 deviates somewhat from a constant acceleration. The difference is, however, not essential. The collective pitch angle is \( \theta_0 = 0 \). The rotor advance ratio is \( \mu = .4 \). The blade frequency in the rotating system is \( P = 1.2 \). System and measurement equations for the single blade case are given by equations 14 and 15 of reference 1.

The sequential estimate of \( \gamma \) and of its covariance, was found from equations 10 and 11, using as initial estimate at time \( t = 12 \) the value zero, and using the initial covariance \( P_0 = 2000 \). The global estimate was determined from equation 12 setting the initial covariance to infinite, or \( P^{-1}_0 = 0 \). Figure 3 gives the estimate \( \hat{\gamma} \) and its covariance from the sequential method for the time \( t = 12 \) to 24 and also the value of the global estimate, \( \hat{\gamma} = 5.05 \). The true value of this
parameter is 5.00. From the covariance plot it is seen that at \( t = 16 \) the covariance is quite small as compared to the initial value so that any further data processing will not appreciably change the estimate. This is born out by the \( \hat{\gamma} \) plot. The global estimate for the time \( t = 24 \) agrees with the sequential estimate at this time, as it should be, since there is little difference between assuming the initial covariance as \( P_o = 2000 \) or \( P_o = \infty \). The method used for figure 3 is the same as that used in reference 1. The flapping response data were noise polluted by white noise with a standard deviation of \( \sigma_\beta = .2 \). This is a large noise since the maximum flapping excursion is only 1.2. The polluted flapping deflections were then passed through a digital filter with the cut-off frequencies of 2.5 to 2.9. Rates of deflections and accelerations were determined from a Kalman filter with the erroneous value of \( \gamma = 4.0 \). Sequential and global estimates were determined with the flapping deflections from the digital filter output and with the flapping rates and accelerations from the Kalman filter output. It is seen that this method worked very well in this case. The effect of the
digital filter cut-off frequencies is, however, substantial. The global method gave for the same case the following estimates

<table>
<thead>
<tr>
<th>Cut-off Frequency</th>
<th>Estimate $\hat{\gamma}$ for $t = 24$</th>
</tr>
</thead>
<tbody>
<tr>
<td>$\omega_c$</td>
<td>$\omega_t$</td>
</tr>
<tr>
<td>1.7</td>
<td>1.9</td>
</tr>
<tr>
<td>2.5</td>
<td>2.9</td>
</tr>
<tr>
<td>3.5</td>
<td>3.9</td>
</tr>
</tbody>
</table>

Because of the sensitivity of the estimate with respect to the cut-off frequencies of the digital filter it was decided to omit the digital filter and to use only the Kalman filter, however in an iterative way.

The following example is again for a rotor advance ratio of $\mu = .4$ blade frequency in the rotating system $\omega_1 = 1.2$ and for a collective pitch setting of $\theta_o = 2^\circ$. Constant acceleration of progressive pitch stirring according to equation 2 is assumed, whereby $t_o = 12$. The polluted flapping response was processed with a Kalman filter using the initial parameter values of $\gamma = 4.0$ and $\delta = \gamma \theta_o = 8.0$ as compared to their true values of 5.0 and 10.0 respectively. The two parameters
assumed to be unknown, γ and δ, were determined by
the global estimate equation 12 with $P_0^{-1}(0) = 0$ and
$R = I$. For the initial conditions of the Kalman
filter $β(0)$ was taken from the simulated measurement
and $\hat{β}(0)$ was taken with a 20% error. The following
table gives in addition to the parameter values for
4 iterations also the diagonal terms of the
covariance matrix. There is some overshoot in both
parameters and the convergence is not very good.
Note that the covariance does not properly reflect
the actual errors in the parameters.

$$δ_β = 0.2, R = I$$

<table>
<thead>
<tr>
<th>Parameter</th>
<th>γ</th>
<th>$δ = γθ_0$</th>
<th>$P_γ$</th>
<th>$P_δ$</th>
</tr>
</thead>
<tbody>
<tr>
<td>True value</td>
<td>5.00</td>
<td>10.00</td>
<td></td>
<td></td>
</tr>
<tr>
<td>Initial estimate</td>
<td>4.00</td>
<td>8.00</td>
<td></td>
<td></td>
</tr>
<tr>
<td>Iteration 1</td>
<td>5.56</td>
<td>8.95</td>
<td>0.092</td>
<td>0.113</td>
</tr>
<tr>
<td>(1.6 CPU</td>
<td>4.76</td>
<td>10.31</td>
<td>0.082</td>
<td>0.111</td>
</tr>
<tr>
<td>seconds per</td>
<td>4.52</td>
<td>10.09</td>
<td>0.073</td>
<td>0.111</td>
</tr>
<tr>
<td>iteration)</td>
<td>4.59</td>
<td>9.93</td>
<td>0.071</td>
<td>0.111</td>
</tr>
</tbody>
</table>

We now treat the same problem with the maximum
likelihood method, and first assume the correct initial
conditions $β(12) = -0.921$ and $\hat{β}(12) = 0.986$. The first 3
iterations are given by
\[ \sigma_\beta = .20 \]

<table>
<thead>
<tr>
<th>Parameter</th>
<th>( \gamma )</th>
<th>( \delta )</th>
<th>( \partial J/\partial \gamma )</th>
<th>( \partial J/\partial \delta )</th>
</tr>
</thead>
<tbody>
<tr>
<td>True value</td>
<td>5.00</td>
<td>10.00</td>
<td></td>
<td></td>
</tr>
<tr>
<td>Initial estimate</td>
<td>4.00</td>
<td>8.00</td>
<td></td>
<td></td>
</tr>
<tr>
<td>Iteration 1</td>
<td>4.94</td>
<td>9.70</td>
<td>-6.11</td>
<td>-5.84</td>
</tr>
<tr>
<td>(2.6 CPU seconds per iteration)</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Iteration 2</td>
<td>4.91</td>
<td>9.82</td>
<td>.10</td>
<td>-.34</td>
</tr>
<tr>
<td>Iteration 3</td>
<td>4.91</td>
<td>9.82</td>
<td>-.002</td>
<td>.001</td>
</tr>
</tbody>
</table>

Though the computer CPU time is now per iteration higher than for the preceding method, the convergence is very rapid and the accuracy is very good. The Cramer-Rao lower bound for the parameter covariance matrix for the second and third iteration is

\[
\begin{bmatrix}
.211 & -.036 \\
-.036 & .365
\end{bmatrix}, \quad R = I
\]

In order to obtain the covariances for the actual simulated measurement noise \( \sigma_\beta = .20 \) one must multiply the above values by \( .04 \). Thus the lower bounds for the parameter standard deviations are \( \sigma_\gamma = .092 \), \( \sigma_\delta = .12 \) which is quite close to the errors in the second iteration.
In a second version of the same problem it is assumed that the initial conditions are unknown and must be included in the parameter identification. The following table gives the results of the first 3 iterations.

\[ \sigma_\beta = .20 \]

<table>
<thead>
<tr>
<th>Parameter</th>
<th>( \gamma )</th>
<th>( \delta )</th>
<th>( \beta(12) )</th>
<th>( \dot{\beta}(12) )</th>
</tr>
</thead>
<tbody>
<tr>
<td>True value</td>
<td>5.00</td>
<td>10.00</td>
<td>-.921</td>
<td>.986</td>
</tr>
<tr>
<td>Initial estimate</td>
<td>4.00</td>
<td>8.00</td>
<td>-.750</td>
<td>1.200</td>
</tr>
<tr>
<td>Iteration 1</td>
<td>4.96</td>
<td>9.68</td>
<td>-.89</td>
<td>1.02</td>
</tr>
<tr>
<td>(4.3 CPU seconds per iteration)</td>
<td>4.90</td>
<td>9.85</td>
<td>-.89</td>
<td>1.01</td>
</tr>
<tr>
<td>Iteration 2</td>
<td>4.90</td>
<td>9.85</td>
<td>-.89</td>
<td>1.01</td>
</tr>
<tr>
<td>Iteration 3</td>
<td>4.90</td>
<td>9.85</td>
<td>-.89</td>
<td>1.01</td>
</tr>
</tbody>
</table>

The parameter covariance lower bounds are

\[
M^{-1} = \begin{bmatrix}
.230 & -.059 & -.048 & .023 \\
-.059 & .405 & .058 & .022 \\
-.048 & .058 & .130 & -.066 \\
.023 & .022 & -.066 & .228 \\
\end{bmatrix}, \quad R = I
\]

The computer CPU time per iteration is once more increased, but the convergence is excellent and the accuracy of the estimate is very good. Again the inverted information matrix gives physically meaningful covariances for the unknown parameters.
One can then conclude that for single blade identification with two unknown parameters $\gamma$ and $\delta = \gamma \theta_0$ the iterative equation error estimation with updated Kalman filter gives the lowest CPU time per iteration, however because of the much faster convergence of the maximum likelihood method and because of its greater versatility (inclusion of initial response values in the set of unknown parameters), it definitely is preferable. For the following cases the maximum likelihood method will be used.

**MULTIBLADE PARAMETER IDENTIFICATION**

The preceding example is now treated with a multiblade representation. The excitation in multiblade coordinates is given by equations 3 to 5 with $t_0 = 12$. The pitch input is shown in Figure 2 assuming a constant progressing stirring acceleration given by equation 6. The multiblade responses $\beta_0, \beta_1, \beta_2$ are polluted from $t = 12$ to $t = 24$ with zero mean Gaussian noise of standard deviation $.2$.

We first identify the parameters $\gamma$ and $\delta = \gamma \theta_0$ with given initial values of $\beta_0, \dot{\beta}_0, \beta_1, \dot{\beta}_1, \beta_2, \dot{\beta}_2$ at $t = 12$. The responses are obtained with the periodic system equations. However, for the parameter identification, constant coefficients are assumed so that
modeling errors are present. The initial states at $t = 12$ are:

$$
\beta_0 = .880, \quad \beta_I = -.983, \quad \beta_{II} = 1.698, \quad \beta_d = .060 \\
\dot{\beta}_0 = .038, \quad \dot{\beta}_I = .034, \quad \dot{\beta}_{II} = .03, \quad \dot{\beta}_d = -.030
$$

$\beta_d$ refers to the reactionless mode of a four-bladed rotor, $\beta_0$ is the coning angle, $\beta_I$ represents forward tilt and $\beta_{II}$ left tilt. For the constant coefficient approximation $\beta_d$ does not couple with the other states and can be omitted. The multiblade equations for a four-bladed rotor are given in reference 9. The exact and the noise polluted responses $\beta_I, \beta_{II}$ are shown in figure 4.

Using the above initial conditions one obtains with the maximum likelihood method the following values:

<table>
<thead>
<tr>
<th>Parameter</th>
<th>$\gamma$</th>
<th>$\delta$</th>
</tr>
</thead>
<tbody>
<tr>
<td>True Value</td>
<td>5.00</td>
<td>10.00</td>
</tr>
<tr>
<td>Initial Estimate</td>
<td>2.50</td>
<td>5.00</td>
</tr>
<tr>
<td>Iteration 1</td>
<td>4.97</td>
<td>8.46</td>
</tr>
<tr>
<td>(4 CPU seconds per iteration) 2</td>
<td>5.00</td>
<td>9.62</td>
</tr>
<tr>
<td></td>
<td>4.99</td>
<td>9.65</td>
</tr>
</tbody>
</table>
The lower bound for the parameter covariance in the 3rd iteration is

\[ M^{-1} = \begin{bmatrix} .125 & .003 \\ .003 & .269 \end{bmatrix}, \quad R = I \]

The result as compared to the preceding single blade identification is better for \( \gamma \) and worse for \( \delta \). Note the assumed large errors of \( 50\% \) in the initial parameter estimate.

In order to assess the effect of errors in the initial states at \( t = 12 \), it is now assumed that all initial states are zero. One then obtains:

\[ \beta_0 = \hat{\beta}_0 = \beta_I = \hat{\beta}_I = \beta_{II} = \hat{\beta}_{II} = 0 \quad \text{at} \quad t = 12 \]

<table>
<thead>
<tr>
<th>Parameter</th>
<th>( \gamma )</th>
<th>( \delta )</th>
</tr>
</thead>
<tbody>
<tr>
<td>True value</td>
<td>5.00</td>
<td>10.00</td>
</tr>
<tr>
<td>Initial estimate</td>
<td>2.50</td>
<td>5.00</td>
</tr>
<tr>
<td>Iteration 1</td>
<td>5.68</td>
<td>9.45</td>
</tr>
<tr>
<td>(4 CPU seconds per iteration)</td>
<td>4.87</td>
<td>10.27</td>
</tr>
<tr>
<td>Iteration 3</td>
<td>4.92</td>
<td>10.26</td>
</tr>
</tbody>
</table>
For $R = I$ the inverted information matrix is

$$M^{-1} = \begin{bmatrix}
.151 & .043 \\
.043 & .278
\end{bmatrix}$$

The result shows that large errors in the initial states can be tolerated.

Finally the 3 initial deflections have been assumed as unknown parameters, increasing the number of parameters to be identified from 2 to 5. The initial values for the response rates were assumed to be zero: $\dot{\beta}_o = \ddot{\beta}_I = \ddot{\beta}_{II} = 0$ at $t = 12$. The initial values of $\beta_o$, $\beta_I$, $\beta_{II}$ are given in the table:

<table>
<thead>
<tr>
<th>Parameter</th>
<th>$\gamma$</th>
<th>$\delta$</th>
<th>$\beta_o(12)$</th>
<th>$\beta_I(12)$</th>
<th>$\beta_{II}(12)$</th>
</tr>
</thead>
<tbody>
<tr>
<td>True values</td>
<td>5.00</td>
<td>10.02</td>
<td>.880</td>
<td>-.983</td>
<td>1.598</td>
</tr>
<tr>
<td>Initial estimates</td>
<td>2.50</td>
<td>5.02</td>
<td>1.00</td>
<td>-1.00</td>
<td>2.02</td>
</tr>
<tr>
<td>Iteration 1</td>
<td>4.94</td>
<td>9.25</td>
<td>.979</td>
<td>-1.039</td>
<td>1.926</td>
</tr>
<tr>
<td>(6 CPU seconds per iteration) 2</td>
<td>5.01</td>
<td>9.97</td>
<td>.920</td>
<td>-.985</td>
<td>1.732</td>
</tr>
<tr>
<td>3</td>
<td>5.00</td>
<td>9.93</td>
<td>.926</td>
<td>-.984</td>
<td>1.730</td>
</tr>
</tbody>
</table>
The inverted information matrix for $\gamma$ and $\delta$ is:

$$M^{-1} = \begin{bmatrix} 0.133 & -0.006 \\ -0.006 & 0.304 \end{bmatrix}, \quad R = I$$

It is seen that now the convergence is much faster, the second iteration being even better than previously the 3rd iteration. However, the CPU time per iteration is increased from 4 seconds to 6 seconds.

The following table compares the results of the various methods. The last 4 rows refer to the maximum likelihood method.

<table>
<thead>
<tr>
<th></th>
<th>Iterations</th>
<th>$\gamma$</th>
<th>$\delta$</th>
<th>Variances of $\gamma$</th>
<th>Variances of $\delta$</th>
<th>Total CPU time, sec.</th>
</tr>
</thead>
<tbody>
<tr>
<td>True Value</td>
<td></td>
<td>5.00</td>
<td>10.00</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Iterated equation</td>
<td>3</td>
<td>4.52</td>
<td>10.09</td>
<td>0.073</td>
<td>0.111</td>
<td>4.8</td>
</tr>
<tr>
<td>error, single blade</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Single blade, correct</td>
<td>2</td>
<td>4.91</td>
<td>9.82</td>
<td>0.0065</td>
<td>0.0146</td>
<td>5.2</td>
</tr>
<tr>
<td>initial conditions</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Single blade, identified initial conditions</td>
<td>2</td>
<td>4.90</td>
<td>9.85</td>
<td>0.0092</td>
<td>0.0162</td>
<td>8.6</td>
</tr>
<tr>
<td>Multiblade, zero</td>
<td>2</td>
<td>4.87</td>
<td>10.27</td>
<td>0.0060</td>
<td>0.0111</td>
<td>8.0</td>
</tr>
<tr>
<td>initial values</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Multiblade, zero</td>
<td>2</td>
<td>5.01</td>
<td>9.97</td>
<td>0.0053</td>
<td>0.0210</td>
<td>12.0</td>
</tr>
<tr>
<td>initial rates identified initial deflections</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

The number of iterations indicated in the table is that for which convergence has been achieved. The iterated equation error estimation with updated Kalman filter
needs the lowest total computer effort, however, the accuracy of the estimate is worst for $\gamma$. The maximum likelihood estimation, due to faster convergence, needs only moderately more computer effort and yields better accuracy. For the single blade the identification of the two initial states, $\beta(12)$ and $\hat{\beta}(12)$, results in the same good accuracy of the parameters as when using the correct initial conditions.

The multiblade identification - despite modeling errors by omitting all periodic terms in the equations of motion and despite assuming zero initial conditions - also converges rapidly and provides only slightly less accuracy of the parameters. The multiblade analysis with identification of the 3 initial deflections gives the best accuracies, however with more than twice the computer effort. Note that the covariance estimate of the Kalman filter method is an order of magnitude greater than for the maximum likelihood method. The latter values have been obtained from $M^{-1}$ for $R = I$ by multiplication with $.04$, since

$$R = \begin{bmatrix} .04 & 0 \\ 0 & .04 \end{bmatrix}$$

They represent lower bounds of the actual parameter covariances but agree fairly well with the errors found for the identified parameters. This is not true for the covariance estimates of the Kalman
filter method. The latter method was attempted also for the multiblade representation but was found to require impractically long computer runs per iteration combined with a lack of convergence. The multiblade identification studies were, therefore, limited to the maximum likelihood method.

**DYNAMIC WAKE PARAMETER IDENTIFICATION FOR ZERO ADVANCE RATIO**

For the case of zero advance ratio reference 10 gives a mathematical model of the dynamic rotor wake that has 2 constants: The quasi-steady wake number \(L\) that relates the pitching and rolling moments of the rotor to the sine and cosine components of the wake velocity, and a time constant \(\tau\). The third rotor constant is

\[
A = B^{\frac{4}{3}} \gamma / 8
\]  

(27)

where \(B\) is the tip loss factor and \(\gamma\) the blade Lock number. The inflow angle from the rotor wake is assumed constant over the radius and represented for the \(k\)th blade by

\[
\lambda_k = \lambda_0 + \lambda_I \cos \psi_k + \lambda_{II} \sin \psi_k
\]  

(28)

In a linearized analysis for zero advance ratio the coning mode is uncoupled from the tilting modes. The system equations of reference 10 are in matrix notation
There are 6 state variables: \( \beta_I, \dot{\beta}_I, \beta_{II}, \dot{\beta}_{II}, \lambda_I, \lambda_{II} \).

The flapping angles of the 4 blades \( \beta_1, \beta_2, \beta_3, \beta_4 \) are measured in the rotating system and combined into \( \beta_1-\beta_3 \) and \( \beta_2-\beta_4 \). The measurement equations - that is the relation between the measured quantities and the system state variables - are given by

\[
\begin{bmatrix}
\beta_{m1} \\
\beta_{m2}
\end{bmatrix} = \begin{bmatrix}
(\beta_1-\beta_3)/2 \\
(\beta_2-\beta_4)/2
\end{bmatrix} = \begin{bmatrix}
\cos t & \sin t \\
-sin t & \cos t
\end{bmatrix}\begin{bmatrix}
\beta_I \\
\beta_{II}
\end{bmatrix} + \begin{bmatrix}
\eta_1 \\
\eta_2
\end{bmatrix}
\]

where \( \eta_1, \eta_2 \) are measurement noise of \( \beta_{m1} \) and \( \beta_{m2} \) respectively. Thus only 2 of the 6 state variables
are measured. The parameters to be identified are $A$, $A_L/\tau$, $l/\tau$, and the blade natural frequency in the rotating system is assumed to be given:

$$\omega_1 = 1.2.$$  

The pitch stirring input is again described by equations 4 to 6 (progressing mode). The rotor wake causes an increased stabilization time. Therefore $t_0 = 70$ was selected instead of $t_0 = 12$. The measurement noise $\eta_1$, $\eta_2$ is assumed to be white and Gaussian with the standard deviations

$$\sigma_{\beta_m1} = \sigma_{\beta_m2} = .10$$

The reduction to one half the noise amplitude as compared to the preceding examples is justified by the fact that $\beta_m1$ and $\beta_m2$ are substantially smaller than $\beta_1$ and $\beta_{II}$ see figures 4 and 5.* The true values of the 3 parameters for which the response was determined are

$$A = .500 \quad A_L/\tau = .250 \quad l/\tau = .125$$

These values are in the range expected to be found for the model rotor. They will depend essentially on the collective pitch setting. In preliminary

*Note that figure 5 refers to $\sigma_{\beta_m} = .05$
computer runs it was found that the initial estimate of $1/\tau$ should not be zero.

Using the true initial conditions one obtains with the maximum likelihood method

<table>
<thead>
<tr>
<th>Parameter</th>
<th>$A$</th>
<th>$AL/\tau$</th>
<th>$1/\tau$</th>
</tr>
</thead>
<tbody>
<tr>
<td>True value</td>
<td>0.500</td>
<td>0.250</td>
<td>0.125</td>
</tr>
<tr>
<td>Initial estimate</td>
<td>0.400</td>
<td>0.200</td>
<td>0.250</td>
</tr>
<tr>
<td>Iteration 1</td>
<td>0.500</td>
<td>0.290</td>
<td>0.086</td>
</tr>
<tr>
<td>(3.8 CPU seconds per iteration)</td>
<td>0.486</td>
<td>0.253</td>
<td>0.125</td>
</tr>
<tr>
<td>2</td>
<td>0.487</td>
<td>0.255</td>
<td>0.128</td>
</tr>
<tr>
<td>3</td>
<td>0.487</td>
<td>0.255</td>
<td>0.128</td>
</tr>
<tr>
<td>4</td>
<td>0.487</td>
<td>0.255</td>
<td>0.128</td>
</tr>
</tbody>
</table>

With the inverted information matrix for the 4th iteration

\[
\begin{bmatrix}
0.052 & 0.059 & 0.013 \\
0.059 & 0.094 & 0.033 \\
0.013 & 0.023 & 0.010 \\
\end{bmatrix}, \quad R = I
\]

Despite the fact that only 2 out of 6 state variables are measured, the parameter identification is very good and the second iteration has almost converged.

The following case is the same as before except for regressing excitation ($+\omega$) instead of progressing excitation ($-\omega$)
<table>
<thead>
<tr>
<th>Parameter</th>
<th>A</th>
<th>A/τ</th>
<th>1/τ</th>
</tr>
</thead>
<tbody>
<tr>
<td>True Value</td>
<td>.500</td>
<td>.250</td>
<td>.125</td>
</tr>
<tr>
<td>Initial Estimate</td>
<td>.400</td>
<td>.200</td>
<td>.250</td>
</tr>
<tr>
<td>Iteration 1</td>
<td>.524</td>
<td>.252</td>
<td>.087</td>
</tr>
<tr>
<td>(3.8 CPU seconds per iteration) 2</td>
<td>.540</td>
<td>.264</td>
<td>.134</td>
</tr>
<tr>
<td>3</td>
<td>.536</td>
<td>.264</td>
<td>.139</td>
</tr>
<tr>
<td>4</td>
<td>.536</td>
<td>.264</td>
<td>.139</td>
</tr>
</tbody>
</table>

\[
M^{-1} = \begin{bmatrix} .167 & .179 & .029 \\ .179 & .256 & .045 \\ .029 & .045 & .013 \end{bmatrix}, \quad R = I
\]

Though convergence is good and the second iteration has almost converged, the errors are now larger than before as expressed also by the \( M^{-1} \) matrix.

The physical reason why regressing modes are less suited for rotor wake identification is the fact that at \( \omega = .2 \) the excitation is in resonance with the regressing flapping mode. At this condition no dynamic rotor wake exists since aerodynamic excitation and aerodynamic damping cancel each other. Since regressing mode transients include a frequency region with a weak dynamic rotor wake, the identification of the wake parameters is less good than for progressing mode transients.
Finally the same case is treated with very large errors in the initial parameter estimates. Now the 4 initial displacements are identified in addition to the 3 parameters, resulting in 7 instead of 3 unknown parameters. The initial rates are assumed to be zero: $\dot{\beta}_I = \dot{\beta}_{II} = \dot{\lambda}_I = \dot{\lambda}_{II} = 0$ at $t = 70$.

Using progressing excitation one obtains

$$\sigma_{\beta ml} = \sigma_{\beta m2} = .10$$

<table>
<thead>
<tr>
<th>Parameter</th>
<th>$A$</th>
<th>$AL/\tau$</th>
<th>$1/\tau$</th>
<th>$\beta_I(70)$</th>
<th>$\beta_{II}(70)$</th>
<th>$\lambda_I(70)$</th>
<th>$\lambda_{II}(70)$</th>
</tr>
</thead>
<tbody>
<tr>
<td>True Value</td>
<td>.500</td>
<td>.250</td>
<td>.125</td>
<td>.497</td>
<td>.188</td>
<td>-.874</td>
<td>-.331</td>
</tr>
<tr>
<td>Initial Estimate</td>
<td>.400</td>
<td>.067</td>
<td>.083</td>
<td>.601</td>
<td>.305</td>
<td>-.531</td>
<td>-.268</td>
</tr>
<tr>
<td>Iteration 1</td>
<td>.443</td>
<td>.184</td>
<td>.150</td>
<td>.500</td>
<td>.211</td>
<td>-.864</td>
<td>-.309</td>
</tr>
<tr>
<td>(63 CPU seconds</td>
<td>.471</td>
<td>.233</td>
<td>.115</td>
<td>.497</td>
<td>.210</td>
<td>-.859</td>
<td>-.322</td>
</tr>
<tr>
<td>per iteration ) 3</td>
<td>.474</td>
<td>.235</td>
<td>.119</td>
<td>.497</td>
<td>.211</td>
<td>-.870</td>
<td>-.327</td>
</tr>
<tr>
<td>4</td>
<td>.474</td>
<td>.236</td>
<td>.120</td>
<td>.497</td>
<td>.211</td>
<td>-.871</td>
<td>-.327</td>
</tr>
<tr>
<td>$M^{-1}$ diagonal 4th iteration, $R = I$</td>
<td>.121</td>
<td>.184</td>
<td>.055</td>
<td>.088</td>
<td>.095</td>
<td>1.069</td>
<td>.796</td>
</tr>
</tbody>
</table>

The convergence is again good and the second iteration is almost converged. The errors in the 3 parameters are somewhat larger than before but still acceptable. The errors in the initial conditions, except for $\beta_{II}$, are also small, though the $M^{-1}$ matrix shows larger
covariances for $\lambda_I$ and $\lambda_{II}$. The initial estimate of the initial displacements $\beta_I, \beta_{II}, \lambda_I, \lambda_{II}$ at $t = 70$ are not arbitrary but are determined from the equilibrium equations before the beginning of pitch stirring, using the initial estimates of $A, AL/\tau$ and $1/\tau$. Since the true initial conditions for $\lambda_I$ and $\lambda_{II}$ are unknown, their inclusion in the identification is necessary. $\beta_I$ and $\beta_{II}$ are measured, thus their initial conditions are known within the accuracy of the measurement and their inclusion in the identification may not be required if the measuring error is small. The computer time for identifying all 4 initial displacements ($\beta_I, \beta_{II}, \lambda_I, \lambda_{II}$ at $t = 70$) is however quite moderate, so that their identification is practical.

On the basis of the computer experiments the method to be used for the rotor model tests at zero advance ratio is then as follows:

1. Estimate the parameters $A, AL/\tau, 1/\tau$

2. Determine from the steady state equations 29 with zero rates and zero accelerations the initial values for $\beta_I, \beta_{II}, \lambda_I, \lambda_{II}$ at time $t = t_0$, using the estimated parameters.

3. Assume zero initial rates:

   $\dot{\beta}_I = \dot{\beta}_{II} = \dot{\lambda}_I = \dot{\lambda}_{II} = 0$ at $t = t_0$
4. Apply the maximum likelihood algorithm to equation 29 and determine iteratively the 7 parameters $A$, $AL/\tau$, $1/\tau$, $\beta_1(t_0)$, $\beta_{II}(t_0)$, $\lambda_1(t_0)$, $\lambda_{II}(t_0)$.

SIMULTANEOUS MEASUREMENT AND SYSTEM NOISE

Though the simplified maximum likelihood algorithm as currently used in aircraft parameter identification work and as defined here by equations 18 to 25 does not provide for system noise, such noise is unavoidable in full scale and wind tunnel tests. The origin of the system noise is either in atmospheric or wind tunnel flow turbulence, or in modeling errors. In order to assess the detrimental effects of the system noise on the quality of the maximum likelihood estimates, the preceding example was recomputed under the assumption that equation 29 has an additional noise term on the right hand side of

$$
\begin{bmatrix}
A & 0 \\
0 & A \\
-(AL + 1)/\tau & 0 \\
0 & -(AL + 1)/\tau
\end{bmatrix}
\begin{bmatrix}
V_1 \\
V_2
\end{bmatrix}
$$

(29a)

This means that we have noisy inflow components $\lambda_I$, $\lambda_{II}$. For $V_1$, $V_2$ a zero mean computer generated Gaussian
sequence with standard deviations \( \sigma_{v_1} = \sigma_{v_2} = .1 \) was assumed. Figure 6 shows the noise polluted inflow components \( \lambda_I, \lambda_{II} \). In addition noise was added as before in the measurement equations 30, however now with standard deviations \( \sigma_{\beta m1} = \sigma_{\beta m2} = .05 \) which is one half the previously assumed value and which is more representative of the expected measurement errors.

We first determine the effect of the lower measurement noise on the parameter identification:

\[
\sigma_{\beta m1} = \sigma_{\beta m2} = .05, \quad \sigma_{v_1} = \sigma_{v_2} = 0
\]

<table>
<thead>
<tr>
<th>Parameter</th>
<th>A</th>
<th>( \Delta I/\tau )</th>
<th>I/\tau</th>
<th>( \beta_1(70) )</th>
<th>( \beta_{II}(70) )</th>
<th>( \lambda_I(70) )</th>
<th>( \lambda_{II}(70) )</th>
</tr>
</thead>
<tbody>
<tr>
<td>True Value</td>
<td>.500</td>
<td>.250</td>
<td>.125</td>
<td>.497</td>
<td>.188</td>
<td>-.874</td>
<td>-.331</td>
</tr>
<tr>
<td>Initial Estimate</td>
<td>.400</td>
<td>.067</td>
<td>.083</td>
<td>.601</td>
<td>.305</td>
<td>-.531</td>
<td>-.268</td>
</tr>
<tr>
<td>Iteration 1</td>
<td>.455</td>
<td>.189</td>
<td>.159</td>
<td>.504</td>
<td>.195</td>
<td>-.880</td>
<td>-.330</td>
</tr>
<tr>
<td>2</td>
<td>.483</td>
<td>.238</td>
<td>.119</td>
<td>.500</td>
<td>.194</td>
<td>-.867</td>
<td>-.339</td>
</tr>
<tr>
<td>3</td>
<td>.485</td>
<td>.239</td>
<td>.123</td>
<td>.501</td>
<td>.194</td>
<td>-.874</td>
<td>-.341</td>
</tr>
<tr>
<td>4</td>
<td>.485</td>
<td>.239</td>
<td>.123</td>
<td>.501</td>
<td>.194</td>
<td>-.874</td>
<td>-.341</td>
</tr>
</tbody>
</table>

It is seen that the accuracy of the parameters is improved as compared to the preceding case of

\[
\sigma_{\beta m1} = \sigma_{\beta m2} = .10, \quad \text{as it should be.}
\]
Combining now the measurement and system noise one obtains:

\[ \sigma_{\beta_1} = \sigma_{\beta_2} = .05, \quad \sigma_{V_1} = \sigma_{V_2} = .10 \]

<table>
<thead>
<tr>
<th>Parameter</th>
<th>A</th>
<th>A(L/\tau)</th>
<th>1/(L)</th>
<th>(\beta_1(70))</th>
<th>(\beta_{II}(70))</th>
<th>(\lambda_1(70))</th>
<th>(\lambda_{II}(70))</th>
</tr>
</thead>
<tbody>
<tr>
<td>True Value</td>
<td>.500</td>
<td>.250</td>
<td>.125</td>
<td>.497</td>
<td>.188</td>
<td>-.874</td>
<td>-.331</td>
</tr>
<tr>
<td>Initial Estimate</td>
<td>.400</td>
<td>.067</td>
<td>.083</td>
<td>.601</td>
<td>.305</td>
<td>-.531</td>
<td>-.266</td>
</tr>
<tr>
<td>Iteration 1</td>
<td>.455</td>
<td>.189</td>
<td>.153</td>
<td>.505</td>
<td>.188</td>
<td>-.860</td>
<td>-.319</td>
</tr>
<tr>
<td>2</td>
<td>.482</td>
<td>.236</td>
<td>.115</td>
<td>.502</td>
<td>.188</td>
<td>-.847</td>
<td>-.324</td>
</tr>
<tr>
<td>3</td>
<td>.483</td>
<td>.235</td>
<td>.118</td>
<td>.502</td>
<td>.190</td>
<td>-.853</td>
<td>-.325</td>
</tr>
<tr>
<td>4</td>
<td>.483</td>
<td>.236</td>
<td>.118</td>
<td>.502</td>
<td>.190</td>
<td>-.854</td>
<td>-.325</td>
</tr>
</tbody>
</table>

It is seen that the system noise indicated in figure 6 has only a small detrimental effect on the accuracy of the identified parameters. Of all computer experiments of parameter identification we have shown here only the results of one computer run. Since these results are dependent on the computer generated random sequences, we are actually dealing with identified parameters that are random variables. A repeat run will, however, show only minor differences in the identified parameters, so that the presented results in comparison to the true values can be considered as typical. The parameter that suffers most in its...
accuracy when system noise is added is the reciprocal time constant $1/\tau$. Even so the error is only 5.5% which would appear quite acceptable for all practical purposes.

CONCLUSIONS

The computer experiments presented here can be used to establish preferences with respect to various alternatives in the rotor parameter identification schemes.

1. **Iterated Equation Error Estimation with Updated Kalman Filter vs. Maximum Likelihood Method**

For single blade parameter identification from pitch stirring transients the equation error method applied in an iterative form using a Kalman filter with the latest parameter updates worked well and required the least computer CPU time. For multiblade parameter identification this method became impractical because of slow convergence and high computer CPU time. The maximum likelihood method worked well both for single blade and multiblade applications, though in case of single blade identification it requires somewhat more computer CPU time. The parameter covariances from the maximum likelihood method are
clearly superior to and more meaningful than the covariances determined with the equation error method. The maximum likelihood method also gave good parameter identifications in the presence of both measurement and system noise, though most of the computer experiments were conducted with measurement noise only. Overall one can conclude that the maximum likelihood method in its simplified form in which system noise is not modeled, is for the applications studied superior to the equation error method and thus will represent the method of choice for the parameter identification from wind tunnel rotor model tests.

2. **Fixed vs. Identified Initial State**

Since prior to the initiation of pitch stirring the rotor is approximately in a steady condition as far as the multiblade coordinates are concerned, one can always assume their initial rates as zero. For two parameter identification in forward flight \((\mu = .4)\) the use of zero initial values for all states gave good identification results. Including the initial displacements in the identification - while retaining initial zero rates of displacement -
greatly improved the convergence and the accuracy, though at a cost of about twice the computer CPU time per iteration. For the multiblade analysis it appears nevertheless practical to identify the initial displacements in order to improve convergence and accuracy of the parameters. For the single blade analysis the initial value of the flapping rate is not zero and must be identified.

3. Single Blade vs. Multiblade Analysis
If rotor wake lag is included, only a multiblade analysis is possible. Without rotor wake lag and using an "equivalent" Lock number to describe the wake effects, a single blade analysis is possible. Initial flapping deflection and rate of deflection should be identified. The multiblade analysis with zero initial rates and identified initial deflections is more accurate though more demanding in computer CPU time per iteration. The multiblade analysis is thus preferable also in those cases where a single blade analysis is possible.

4. Progressing vs. Regressing Pitch Stirring
Progressing pitch stirring leads to more accurate rotor wake parameter identification than regressing
pitch stirring, because the regressing stirring transient passes through the flapping resonance at which no dynamic rotor wake exists.
References


Chapter 2

FIGURE CAPTIONS

Figure 1  Time History of Blade Pitch, Rotating Reference System, Progressing Mode

Figure 2  Time History of Blade Pitch, Multiblade Representation, Progressing Mode

Figure 3  Estimate $\hat{\gamma}$ and its Variance for $P(0) = 2000$, $\dot{\gamma}(0) = 0$, $\sigma_\beta = .2$, $\theta_0 = 0$, $\mu = .4$, $\omega_1 = 1.2$, Progressing Excitation

Figure 4  Exact and Noise Polluted Multiblade Responses $\beta_I$, $\beta_{II}$ to Inputs Shown in Figure 2 for $\mu = .4$, $\theta = 2^\circ$, $\gamma = 5.00$, $\omega_1 = 1.20$, $\sigma_{\beta_0} = \sigma_\beta I = \sigma_\beta II = .2$

Figure 5  Exact and Noise Polluted Multiblade Responses $\beta_{m1}$, $\beta_{m2}$ to Inputs Shown in Figure 2 for $\mu = 0$, $\Lambda = .500$, $\Lambda_l/\tau = .250$, $\lambda/\tau = .125$, $\omega_1 = 1.20$, $\sigma_{\beta_{m1}} = \sigma_{\beta_{m2}} = .05$

Figure 6  Exact and Noise Polluted Inflow Components $\lambda_I$, $\lambda_{II}$ (system noise) for $\mu = 0$, $\Lambda = .500$, $\Lambda_l/\tau = .250$, $\lambda/\tau = .125$, $\omega_1 = 1.20$, $\sigma_\lambda I = \sigma_\lambda II = .10$
Fig. 1
Fig. 2
GLOBAL ESTIMATE $\hat{\gamma} = 5.05$

$\theta_0 = 0$

$p_0 = 2000$

Fig. 3
Fig. 4

REPRODUCIBILITY OF THE ORIGINAL PAGE IS POOR
Fig. 5
Fig. 6
CHAPTER 3

OPTIMAL DATA UTILIZATION FOR PARAMETER IDENTIFICATION

PROBLEMS WITH APPLICATION TO LIFTING ROTORS

K. H. Hohenemser and D. Banerjee

ABSTRACT

A method is developed for optimal data utilization in the maximum likelihood identification of systems without process noise, based on stipulated upper bounds of parameter covariances. The method is applied to a case of simulated transient wind tunnel testing of lifting rotors.

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OPTIMAL DATA UTILIZATION FOR GIVEN INPUT FUNCTION 9
APPLICATION TO A CASE OF LIFTING ROTOR PARAMETER IDENTIFICATION 12
ANALYSIS OF RESULTS 16
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INTRODUCTION

Methods for state and parameter estimation from transients are widely used in aircraft testing.\textsuperscript{1,2,3,4} The problem is to obtain optimum estimates (based on certain performance criteria) of initial states and of unknown parameters (derivatives) from noisy measurements of some inputs and response variables. In most cases of airplane parameter identification a constant system is used as an analytical model. For lifting rotor applications a periodic system model is required.\textsuperscript{5} State and parameter identification from transients looks promising also for wind tunnel testing and may well drastically reduce the amount of test efforts as compared to e.g., frequency response testing of lifting rotors. The following study was performed in preparation for transient wind tunnel tests with a lifting rotor model.

In the most general case the identification method has to account for the following types of uncertainties:
(a) Modeling errors which originate from the difference between the mathematical model and the actual physical phenomenon.

(b) Uncertainties from input noise, e.g., turbulence (in the atmosphere or in the wind tunnel).

(c) Uncertainties from measurement inaccuracies. This includes measurement bias and noise and also incomplete measurements when some of the input and output data are missing.

Three basic methods have been applied to the aircraft identification problem. If the measurements are a linear function of the unknown parameters, the classical least squares regression or equation error method is applicable. This method was widely used in the early stages of aircraft parameter identification. For the application of this method the time histories of all the states and their derivatives are needed together with the input variables. Since the equation error method becomes less reliable the noisier the measurements are, it has been replaced in the last decade by less restrictive methods that allow considerable measurement noise and that work also when some states have not been measured. One of the most widely used is the maximum likelihood method, which, in the absence of modeling errors and input noise, reduces to what is also
called the output error method.\textsuperscript{1,2,4} Another way of overcoming the limitations of the least squares regression method is the application of the extended Kalman filter.\textsuperscript{3} This note is applicable to data analysis with the output error method. This method is not necessarily limited to smooth air testing. Atmospheric or wind tunnel turbulence may be included if appropriate input measurements are taken.\textsuperscript{6,7}

In aircraft or wind tunnel transient testing the question comes up as to what kind of transient should be selected. If the transient is too short, the parameters will be identified with inadequate accuracy. If the transient is too long, an unnecessary amount of data must be processed. The question we pose here for the maximum likelihood method is: Given a required accuracy of the parameter estimate, and given an input function, what is the minimal quantity of measured data necessary to achieve this accuracy? From a survey of the pertinent literature it appears that this question has not been asked before, much less a solution presented. However, there are some recent studies where certain criteria were used to define an optimum input. We will first briefly discuss two of these optimal input proposals, and then proceed to develop the method of optimal data utilization for a given type of input.
GENERAL QUESTIONS OF INPUT DESIGN

(a) What type of input function should be used?

(b) For what time period should the response data be processed to enable identification of the system parameters with a specified accuracy? Are certain time periods of the response particularly rich in information contents and should they, therefore, be preferably used?

There usually are some constraints on the input design like amplitude constraints, smoothness constraints (step or impulse inputs are mathematical idealizations but often practically not realizable), instrumentation constraints, and constraints imposed by the selected analytical model that usually filters out the higher frequency contents of the input.

Analytical solutions of the problem of optimal input design require the minimization of a cost function. Stepner and Mehra use the sensitivity of the system response to the unknown parameters as the performance criterion for optimal input design. The time of the transient is assumed to be fixed. Thus questions (b) are not involved. The measurement equation is

\[ y_m(t) = y(x, \theta, u, t) + v(t) \]  

(1)

x is the state vector, \( \theta \) the parameter vector, u the input vector, v(t) the additive measurement noise with
covariance matrix \( R \). We write the Taylor expansion with respect to the parameter \( \Theta \) about the a priori estimate \( \Theta_0 \) of \( \Theta \) and neglect higher order terms:

\[
y_m(t) = y(x, \Theta_0, u, t) + \frac{\partial}{\partial \Theta_0} y(x, \Theta_0, u, t)(\Theta - \Theta_0) + v(t)
\]

In the output error method \((\Theta - \Theta_0)\) is determined by a least squares solution of Eq. (2) for a fixed time period \((t_0, t_f)\). For a high degree of accuracy in determining \((\Theta - \Theta_0)\) the sensitivity function \(\partial y/\partial \Theta\) must be large. The scalar performance index selected in Reference 1 is

\[
J = \text{Trace}(WM)
\]

where

\[
M = \int_{t_0}^{t_f} (\partial y/\partial \Theta)^T R^{-1}(\partial y/\partial \Theta) dt
\]

Due to the introduction of \( R^{-1} \) in \( M \) the performance criterion favors the measurements which are more accurate. The weighting matrix \( W \) is based on the relative importance of the parameter accuracies.

Assuming now linear system and measurement equations

\[
x(t) = F x(t) + G u(t)
\]

\[
y_m(t) = H x(t) + v(t)
\]
together with an "energy constraint" for the input

\[ E = \int_{t_0}^{t_f} u^T u \, dt \]  \hspace{1cm} (7)

the optimum input \( u \) can be determined as a two point boundary value problem whereby the Hamiltonian includes the term

\[ (1/2) \mu_0 (u^T u - E/t_f) . \]

\( \mu_0 \) is the time invariant Langrange factor (scalar) to be evaluated from the Euler differential equations of the optimization problem. It should be noted that the "energy constraint" Eq. (7) has no physical significance but is a convenient device to obtain smooth input functions. Physically the input will usually be limited by amplitude rather than by the quadratic criterion (7) and quite different "optimal" inputs can then be expected.

Chen\(^8\) attacks the problem of optimal input design in an entirely different way as a time-optimal control problem by minimizing

\[ J = \int_{-t_0}^{t_f} dt \]  \hspace{1cm} (8)

under the following constraints:
System equations

\[ \dot{x} = f(x, u, \theta, t), \quad x(t_0) = x_0 \]  

(9)

Sensitivity equations

\[ \frac{\partial x}{\partial x_o} = (\frac{\partial f}{\partial x})(\frac{\partial x}{\partial x_o}), \quad \frac{\partial x(t_0)}{\partial x_o} = I \]  

(10)

\[ \frac{\partial x}{\partial \theta} = (\frac{\partial f}{\partial x})(\frac{\partial x}{\partial \theta}) + \frac{\partial f}{\partial \theta}, \quad \frac{\partial x(t_0)}{\partial \theta} = 0 \]  

(11)

Information matrix equations

\[ M^{-1} = -M^{-1}(\frac{\partial \varphi}{\partial \theta})^T R^{-1}(\frac{\partial \varphi}{\partial \theta})M^{-1} \]  

(12)

where \( \varphi \) is the innovation:

\[ \varphi = y_m - y \]  

(13)

and where the information matrix \( M \) is given by Eq. 4.

Finally Chen assumes an amplitude constraint

\[ |u| \leq \sigma \]  

(14)

and he prescribes the trace of the information matrix for time \( t_f \)

\[ C_{ii}(t_f) = \sigma_i^2 \]  

(15)

One can show that for linear input \( u(t) \) into the system equation and for an input matrix independent of any unknown parameter, the optimal input is of the "bang-bang" form between the amplitude constraints. The solution of this problem requires a computer search which was not performed in Reference 8. Rather an arbitrary set of
bang-bang inputs in the form of Walsh functions was shown to result in a specific case in lower values of \( M^{-1}(t_f) \) (given \( t_f \)) than those obtained by using Mehra's "optimal input". This apparent contradiction can be explained by the differential equation (12) governing \( M^{-1} \). For a particular value of \( M^{-1} \) the rate of decrease of \( M^{-1} \) with time is dependent on all elements of

\[
(\partial y/\partial \theta)^T R^{-1}(\partial y/\partial \theta)
\]

while Mehra, in his criterion (3), optimizes only the trace of \( WM \).

While the input amplitude constraint Eq. (14) used by Chen is physically more significant than the quadratic constraint Eq. (7) used by Mehra, the actual constraints are usually still more complex. In cases of airplanes or lifting rotors one usually wishes to limit the response to the linear sub stall regime, since the analytical model to be identified is often a linear one. The stall boundary is, however, a complex function of the input and cannot be represented by an amplitude constraint for the input transient. This is particularly relevant for the lifting rotor, so that neither the Mehra nor the Chen input optimization criteria are useful for lifting rotor applications,
quite apart from the excessive computer effort involved in obtaining the optimal inputs. Furthermore, the input matrix usually contains unknown parameters. In this case Chen's optimum solution would not be of the bang-bang type and would be still more difficult to obtain. For all of these reasons it was concluded that at the present state of optimal input design methods an attempt to compare our selected inputs with an "optimum input" would not be practical. Instead, a more limited approach has been taken described in the following section.

**OPTIMAL DATA UTILIZATION FOR GIVEN INPUT FUNCTION**

We first point out the difference between the continuous and the discrete case. In the maximum likelihood method (output error method for zero process noise) using the Newton-Raphson approach with quasi linearization, one obtains for the parameter update increment the following expressions:

**Continuous case:**

\[
\Delta \theta = \left[ \int_{t_0}^{T_f} \left( \frac{\partial v}{\partial \theta} \right)^T R^{-1} \left( \frac{\partial v}{\partial \theta} \right) dt \right]^{-1} \int_{t_0}^{T_f} \left( \frac{\partial v}{\partial \theta} \right)^T R^{-1} v dt \tag{16}
\]

**Discrete case:**

\[
\Delta \theta = \sum_{i=1}^{N} \left( \frac{\partial v}{\partial \theta} \right)_i R^{-1} \left( \frac{\partial v}{\partial \theta} \right)_i \tag{17}
\]
The Cramér-Rao lower bound has been defined only for a vector of sampled measurements and not for the continuous case.\(^2,^9\)

For high sampling rate one can define an approximate differential equation for \(M\) from Eq. (17) in the following way:

Set \(S_i \equiv (\partial y/\partial \theta)_i\) \hfill (18)

then

\[
M = \sum_{i=1}^{N} S_i^T R^{-1} S_i = \frac{1}{\Delta t} \sum_{i=1}^{N} S_i^T R^{-1} S_i \Delta t \hfill (19)
\]

As \(N\) increases, \(\Delta t\) gets smaller and the right hand side of Eq. (19) can be approximated by

\[
M \approx (1/\Delta t) \int_{t_0}^{t_f} S^T R^{-1} S \, dt \hfill (20)
\]

\[
M^{-1} \approx \Delta t \left[ \int_{t_0}^{t_f} S^T R^{-1} S \, dt \right]^{-1} \hfill (21)
\]

Taking the derivative of \(M^{-1}\) with respect to \(t_f\):

\[
dM^{-1}/dt_f = -M^{-1}(\partial M/\partial t_f)M^{-1}
\]

or with Eq. (20):

\[
dM^{-1}/dt_f = -(1/\Delta t)M^{-1} S^T R^{-1} S M^{-1} \hfill (22)
\]
The point is that even in a continuous formulation the time increment \( \Delta t \) between samplings must occur. Eq. (21) is the correct formulation for the Cramér-Rao lower bound of the covariance matrix for the parameters.

We can now use the approximately valid differential equation (22) to obtain some insight into ways of best data utilization. Let us assume that we wish to prescribe certain values for the parameter standard deviations \( \sigma_i \) and that we wish to compare the Cramér-Rao lower bound with these standard deviations. Since we are dealing not with the unknown actual parameter covariances but only with their lower bounds, we should apply some conservatism to the selected \( \sigma_i \), that is we should select \( \sigma_i \) smaller than we really need for the specific data processing case. We thus require

\[
0 \leq M^{-1}_{t_f} (i, i) \leq \sigma_i^2 \tag{23}
\]

whereby \( M^{-1}_{t_f} \) is the value of \( M^{-1} \) at time \( t_f \). For non-zero values of \( S \), the right hand side of Eq. (22) is negative definite and hence \( M^{-1}_{t_f} (i, i) \) are monotonically decreasing functions of \( t_f \). There will thus be a minimum time for which the constraints of Eq. (23) are satisfied.

Another way of reducing the amount of measured data for the parameter identification is to select for
the data processing those time periods for which the components of the matrix

\[ S^T R^{-1} S \]

have significant values. From Eq. (21) it follows that then the Cramér-Rao lower bound \( M^{-1} \) will be particularly small. The components of \( M^{-1} \) also decrease with decreasing time element \( \Delta t \) between samples.

Since it is impractical to use for the integration of Eq. (22) infinity as initial condition, it is recommended to determine \( M^{-1} \) for a small time period, say for \( N = 10 \), from Eq. (21) and integrate Eq. (22) with the solution to Eq. (21) as initial conditions. Since \( S \) includes parameter estimates, one needs a preliminary estimation of the unknown parameters in order to use Eq. (22).

APPLICATION TO A CASE OF LIFTING ROTOR PARAMETER IDENTIFICATION

Using the simplest analytical model of a lifting rotor, a straight blade flapping about the rotor center, one has in a rotating frame of reference for the flapping angle \( \beta \) the following equation.\(^{10}\)

\[
\ddot{\beta} + \left(\frac{\gamma}{2}\right)C(t)\dot{\beta} + \left(\frac{\gamma}{2}\right)[P^2 + K(t)]\beta = \left(\frac{\gamma}{2}\right)[m_\theta \theta + m_\lambda \lambda] \\
(24)
\]

where: \( \gamma \) is the blade Lock number

\( P \) is the blade flapping natural frequency in the rotating system

\[ \text{REPRODUCIBILITY OF THE ORIGINAL PAGE IS POOR} \]
\( \theta \) is the instantaneous blade pitch angle
\( \lambda \) is the non-dimensional normal inflow

One rotor revolution corresponds to \( t = 2\pi \). For neglected reversed flow effects, zero root cut-out and with tip loss factor \( B \), the functions \( C(t) \), \( K(t) \), \( m_\theta(t) \), \( m_\lambda(t) \) in terms of rotor advance ratio \( \mu \) are:\(^{10}\)

\[
C(t) = \frac{1}{4}B^4 + \frac{1}{3}B^3 \mu \sin(t) \\
K(t) = \frac{1}{3}B^3 \mu \cos(t) + \frac{1}{4}B^2 \mu^2 \sin(2t) \\
m_\theta(t) = \frac{1}{4}(B^4 + B^2 \mu^2) + \frac{2}{3}B^3 \mu \sin(t) \quad (27)
- \frac{1}{4}B \mu^2 \cos(2t) \\
m_\lambda(t) = \frac{1}{3}B^3 + \frac{1}{2}B^2 \mu \sin(t) \quad (28)
\]

In the numerical analysis, we use \( B = 0.97 \). A simple improvement of this analytical model that takes into account blade bending flexibility is possible.\(^{11}\)

In transient conditions the inflow \( \lambda \) includes the dynamic rotor wake in a complicated form. As a first approximation of dynamic rotor wake effects one can use, instead of the actual blade Lock number an equivalent smaller value of \( \gamma \).\(^{12}\) Such an approximation can be expected to be satisfactory if the transient is relatively slow. For transients with high frequency contents, this approximation is invalid.\(^{13}\)
Due to rotor induced cross flow in a wind tunnel, the inflow parameter $\lambda$ will usually not be well known. In addition, the aerodynamic pitch angle $\theta_0$, due to airfoil inaccuracies and pitch setting errors, will also not be well known. For the wind tunnel tests considered here, we assume $\lambda = 0$ and use the collective pitch setting $\theta_0$ as an unknown parameter to be determined from the blade flapping measurements. In addition we have a transient blade pitch input $\theta$ assumed to be known. The problem then is to determine from blade flapping transients caused by blade pitch inputs, the equivalent Lock number $\gamma$ and the equivalent collective pitch setting $\theta_0$.

The lifting rotor wind tunnel model described in reference 14 allows excitation of progressing and regressing flapping modes at various frequencies. By a minor modification of this model, progressing or regressing transients can be excited. One can describe such inputs as pitch stirring transients. In a helicopter, this would amount to cyclic stick stirring, whereby the amplitude of the cyclic pitch would remain constant while the frequency of the stirring motion changes. The blade pitch input for such a stirring transient is selected to satisfy the equation:

$$\theta = 1.5 \cos[\omega(t - t_0) + t] + \theta_0$$  \hspace{1cm} (29)
where \( \omega \) is the angular stirring speed in the sense of blade rotation

\[
\begin{align*}
\omega &= \begin{cases} 
12 & \text{if } 0 < t < t_0 \\
-(0.1/\pi)(t - t_0) & \text{if } t_0 < t < T
\end{cases}
\end{align*}
\]

The meaning of these input equations is the following. At time \( t = 0 \), a step lateral cyclic pitch input of 1.5 degrees is imposed. At time \( t = 12 \), the response to this input is approximately stabilized. At this time the pitch stirring acceleration of the progress leads to a progressing flapping excitation. The identification starts at \( t = 12 \) with the pitch stirring transient.

Test results with transient pitch stirring inputs will be presented at a later time when they become available. Here we are concerned with the problem of designing the tests in such a way that the test data will be sufficient to determine the two unknown parameters \( \gamma \) and \( \theta_0 \) with good accuracy, i.e., to determine a suitable value of \( T \) that allows an accurate identification of parameters.

The simulated identification analysis was performed under the assumption of a random zero mean white noise sequence superimposed on the analytical flapping transient.
This transient was obtained for $\theta_0 = 20^\circ$, $\mu = 0.4$ and $\gamma = 5.0$. For convenience, the parameters $\delta = \gamma \theta_0$ and $\gamma$ instead of $\theta_0$ and $\gamma$ were identified.

System and measurement equations corresponding to equations (5) and (6) are:

$$
\begin{bmatrix}
  \dot{x}_1 \\
  \dot{x}_2
\end{bmatrix} = 
\begin{bmatrix}
  0 & 1 \\
  -(\gamma/2)(P^2 + K(t)) & -(\gamma/2)C(t)
\end{bmatrix}
\begin{bmatrix}
  x_1 \\
  x_2
\end{bmatrix}
+ 
\begin{bmatrix}
  (\gamma/2)m_{\theta_0}
\end{bmatrix}
$$

(30)

$$
y_m = [1 \ 0] \begin{bmatrix} x_1 \\ x_2 \end{bmatrix} + v(t)
$$

(31)

where

$$
E\{v(t)\} = 0 \quad E[v(t)v(t)] = 0.2 \delta(t - \tau)
$$

and $$[x_1 \ x_2] = [\beta \ \delta]$$

**ANALYSIS OF RESULTS**

We first show in Table 1 the effect of data length on the parameters and their associated $M^{-1}(i,i)$ values. The iteration of the maximum likelihood method was begun with a 20 percent error in the parameter values. It is seen that a data length of $t = 12 - 14$ is quite inadequate, a data length of $t = 12 - 18$ gives reasonably good parameters, while a data length of $t = 12 - 24$ is much better and leads to very small
lower bounds of the parameter covariance matrix.

Fig. 1 shows the correct flapping response together with the simulated measurement data. Pitch stirring is initiated at \( t = 12 \). Figs. 2 and 3 show \( M^{-1}(\gamma) \) and \( M^{-1}(\delta) \) from Eq. (22) between \( t = 16 \) and \( t = 24 \).

Two curves are plotted, one for the initial crude estimate of the parameters \( (\gamma = 4, \delta = 8) \), and one for the final estimate of the parameters for \( t = 24 \), \( (\gamma = 4.91, \delta = 9.83) \). The two curves are in this case not much different. Note the steep descent of the curves to about \( t = 17.5 \). It would, therefore, be not acceptable to use the data up to less than the time \( t = 17.5 \). However, there is another descent to \( t = 23.0 \), causing the improvement shown in Table 1.

From Figs. 2 and 3 it is clear that the selection of \( T = 24.0 \) is a good one, that the use of fewer data would result in substantial decrease in parameter accuracy, and that the use of additional data is unnecessary.
REFERENCES


FIGURE CAPTIONS

Fig. 1 Simulated measurements of $\beta$, $\sigma_\beta = .2$

Fig. 2 Cramer-Rao Lower Bound $M^{-1}(\gamma)$ vs. time

Fig. 3 Cramer-Rao Lower Bound $M^{-1}(\delta)$ vs. time

Table 1 Iteration vs. Time Variation of Parameters and Their Cramer-Rao Lower Bounds.
**FIGURE 1**

*SIMULATED MEASUREMENTS*

---

*MODEL RESPONSE*

---

Axis labels:
- Y-axis: \( \beta \)
- X-axis: \( \text{TIME} \)
\[ M^{-1}(\gamma) \]

\[ \text{FIGURE 2} \]

- \( \gamma = 4.0 \) \( \delta = 8.0 \)
- \( \gamma = 4.91 \) \( \delta = 9.83 \)
\[ M^{-1}(\delta) \]

\( (\gamma = 4.0; \delta = 8.0) \)

\( (\gamma = 4.91; \delta = 9.83) \)

**Figure 3**
<table>
<thead>
<tr>
<th>Parameter</th>
<th>$t = 12 - 14$</th>
<th>$t = 12 - 18$</th>
<th>$t = 12 - 24$</th>
</tr>
</thead>
<tbody>
<tr>
<td>True value</td>
<td>$\gamma$ 5.00 $\delta$ 10.00</td>
<td>$\gamma$ 5.00 $\delta$ 10.00</td>
<td>$\gamma$ 5.00 $\delta$ 10.00</td>
</tr>
<tr>
<td>Iteration 0</td>
<td>$\gamma$ 4.00 $\delta$ 8.00</td>
<td>$\gamma$ 4.00 $\delta$ 8.00</td>
<td>$\gamma$ 4.00 $\delta$ 8.00</td>
</tr>
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<td>$\gamma$ 4.29 $\delta$ 9.73</td>
<td>$\gamma$ 5.36 $\delta$ 9.67</td>
<td>$\gamma$ 4.94 $\delta$ 9.69</td>
</tr>
<tr>
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<td>$\gamma$ 5.23 $\delta$ 9.73</td>
<td>$\gamma$ 4.91 $\delta$ 9.85</td>
</tr>
<tr>
<td>3</td>
<td>$\gamma$ 4.10 $\delta$ 9.67</td>
<td>$\gamma$ 5.23 $\delta$ 9.73</td>
<td>$\gamma$ 4.91 $\delta$ 9.83</td>
</tr>
</tbody>
</table>

**TABLE 1**