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Brincker, Rune; Andersen, Palle

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Ambient Response Analysis
Modal Analysis for Large Structures

Rune Brincker
Department of Building Technology and Structural Engineering, Aalborg University
Sohngardsholmvej 57, DK-9000 Aalborg, Denmark

Palle Andersen
Structural Vibration Solutions ApS.
NOVI Science Park, Niels Jeruns Vej 16, DK 9220 Aalborg Ø, Denmark

ABSTRACT
In this paper an outline is given of the basic ideas in ambient response analysis, i.e. when modal analysis is performed on systems based on output only. Some of the most know techniques are briefly introduced, and the basic problems discussed. The introduced techniques are the frequency domain based peak-picking methods, the polyreference LSCE method, the stochastic subspace method for estimation of state space systems and the prediction error method for estimation of Auto-Regressive Moving Average Vector models. The techniques are illustrated on an example of ambient response measurements made on a highway bridge.

NOMECLATURE

- $y_k$: Vector of measured output.
- $R_k$: Correlation function between outputs.
- $\Delta t$: Sampling interval.
- $\lambda$: Discrete-time system pole.
- $\mu$: Continuous-time system pole.
- $\mathbb{E}[*]$: Expectation operator.
- $\psi$: Mode shape vector.
- $L$: Vector of multipliers.
- $A$: State (transition) matrix.
- $C$: Output (observation) matrix.
- $x_k$: State vector.
\( A \), Auto-Regressive coefficient matrix.

\( B \), Moving Average coefficient matrix.

1 INTRODUCTION

In traditional modal analysis the modal parameters are obtained from the Frequency Response Functions, i.e. the function relating the output (the response) and the input (the loading). Thus, in traditional modal analysis the input must be known. For smaller structures that can be tested in the laboratory, or for larger structures that can be excited artificially without significant problems, this approach should be preferred.

For larger structures however, the traditional approach is often difficult to implement. If the structure is large like a suspension bridge or a dam, it might be difficult to artificially load the structure to a level where the response from the background loading like wind or other types of non-controllable loading is small compared to the response from the artificial loading. Even in the cases where this is possible problems may arise from non-linearities introduced by exciting the structure to a higher response level. Applying artificial loading might also be expensive and involving a risk of damaging the structure.

Thus, for large structures, the natural solution is to measure the natural responses, also denoted the ambient responses, and then analyse them to obtain the same information that could be achieved from a traditional modal analysis test. Instead of loading the structure artificially and dealing with the natural loading as an unwanted noise source, the natural loading is used as the loading source.

The main advantages of this kind of testing are:

- Testing is cheap and fast, since big equipment for excitation is not needed,

- Testing does not interfere with the operation of the structure,

- The measured response is representative for the real operating conditions of the structure.

However, using this kind of testing, responses are small, and often partly covered in noise. Further, the loading is unknown, and, thus, the analysis becomes more difficult than in traditional modal analysis. The main drawbacks are:

- very sensitive equipment needed,

- careful data analysis needed.

These drawbacks are the main reasons that these kinds of techniques have not been used in large scale in the past. Now, however, the problems are vanishing: The last few years prices on high quality equipment has dropped significantly, analysis techniques are developing, and the necessary large scale computer analysis can now be performed on a PC.
The theory behind the four different methods is briefly reviewed in section 2. In section 3, the test structure, a Swiss highway bridge, as well as the test procedure and the practical aspects of using the four different methods, are presented. Section 4 presents the results.

In this paper four methods of ambient response analysis is briefly introduced and compared. The methods are the frequency domain peak-picking (PP) method, the polyreference LSCE (LSCE) method, a stochastic subspace identification technique for estimation of state space systems (SSI), and the prediction error method (PEM) applied to Auto-Regressive Moving Average Vector (ARMAV) models.

2 APPLIED IDENTIFICATION TECHNIQUES

2.1 Peak-Picking

A fast method to estimate the modal parameters of a structure based on output-only measurements is the rather simple peak-picking method. The method is widely used and one practical implementation of the method was realized by Felber, see Felber [1]. In this implementation the natural frequencies are determined as the peaks of the Averaged Normalized Power-Spectral Densities (ANPSDs). The ANPSDs are basically obtained by converting the measured data to the frequency domain by a Discrete Fourier Transform (DFT). The coherence function computed for two simultaneously recorded output signals has values close to one at the natural frequencies, see Bendat et al. [2]. This fact also helps to decide which frequencies can be considered natural. The components of the mode shapes are determined by the values of the transfer functions at the natural frequencies. Note that in the context of ambient testing, transfer function does not mean the ratio of response over force, but rather the ratio of response measured by a roving sensor over response measured by a reference sensor. So every transfer function yields a mode shape component relative to the reference sensor. It is assumed that the dynamic response at resonance is only determined by one mode. The validity of this assumption increases as the modes are better separated and as the damping is lower. The method has been used successfully for a large amount of structures, see Felber et al. [3].

2.2 Polyreference LSCE applied to Auto- and Cross-Correlation Functions

On the assumption that the system is excited by stationary white noise it has been shown that correlation functions between the response signals can be expressed as a sum of decaying sinusoids, see James et al. [4]. Each decaying sinusoid has a damped natural frequency and damping ratio that is identical to that of a corresponding structural mode. Consequently, the classical modal parameter estimation techniques using impulse response functions as input like Polyreference LSCE, Eigensystem Realization Algorithm (ERA) and Ibrahim Time Domain are also appropriate for extracting the modal parameters from response-only data measured under operational conditions. This technique is also referred to as NExT, standing for Natural Excitation Technique, see James et al. [4]. In this paper the discussion will be limited to polyreference LSCE.

The correlation functions between the outputs and a set of outputs serving as references are defined as:

$$R_y = E[y_{ref},y_{ref}^T] \in \mathbb{R}^{l_{ref} \times l}$$  (1)

$y \in \mathbb{R}^l$ is the output vector containing $l$ channels, $y_{ref} \in \mathbb{R}^{l_{ref}}$ is a subset of $y$ containing only the $l_{ref}$ references, and $E[\cdot]$ denotes the statistical expectation. The correlation functions can be estimated by replacing the expected value operator in (1) by a summation over the available
measurements. Using the unbiased DFT this calculation can be performed significantly faster. If the unbiased DFT is not used, the results will be biased by leakage, and the damping will be over-estimated.

The polyreference LSCE yields global estimates of the poles and the modal reference factors [5]. Mathematically, the polyreference LSCE decomposes the correlation functions in a sum of decaying sinusoids:

\[ R_k = \sum_{i=1}^{n_p} \{ \psi_i, \lambda_i^k L_i^T + \psi_i, \lambda_i^{-k} L_i^{-T} \} \]  

(2)

where \( n_p \) is the number of poles; \( \psi_i \in \mathbb{C}^{n} \) is the \( r \) mode shape; \( \lambda_i = e^{i \omega i \Delta t} \) is the \( r \) complex discrete system pole (related to the continuous system pole \( \mu_i \) and the sample time \( \Delta t \)); \( L_i \in \mathbb{C}^{n \times r} \) is a vector of multipliers which are constant for all response stations for the \( r \) mode. Note that in conventional modal analysis, these constant multipliers are the modal participation factors. In case of output-only modal analysis, they will be further referred to as the modal reference factors. It can be proved that if the correlation data can be described by (2), it can also be described by the following model:

\[ R_k I + R_{kj}F_i + \ldots + R_{kj}F_i = 0 \]  

(3)

if the following conditions are fulfilled:

\[ L_i^T (\lambda_i^k + \lambda_i^{-k} F_i + \ldots + \lambda_i^{-k} F_i) = 0 \]  

(4)

\[ \lambda_i^{kr} \geq 2n_p \]  

(5)

Equation (3) represents a coupled set of finite difference equations with constant matrix coefficients \( F_i, \ldots, F_i \in \mathbb{R}^{n \times p} \). The condition expressed by (4) states that the terms \( \lambda_i L_i^T \) are characteristic solutions of this system of finite difference equations. As (3) is a superposition of \( 2n_p \) such terms, it is essential that the condition given by (5) is fulfilled.

Polyreference LSCE essentially comes down to estimating the matrix coefficient \( F_i, \ldots, F_i \). Once these are known, (4) can be reformulated into a generalized eigenvalue problem resulting into \( \lambda_i \), yielding estimates for the system poles \( \mu_i \) and the corresponding left eigenvectors \( L_i^T \). Equations similar to (3) can be formulated for all possible correlations \( R_k \). The obtained overdetermined set of equations can than be solved in least squares sense to yield the matrix coefficients \( F_i, \ldots, F_i \). The order \( i \) of the finite difference equation is related to the number of modes in the data.

Contrary to the stochastic subspace and ARMAV methods (cf. the next two sections), the polyreference LSCE does not yield the mode shapes. So, a second step is needed to extract the mode shapes using the identified modal frequencies and modal damping ratios. This can be done either by fitting the correlation functions in the time domain or by fitting the power- and cross-spectral densities in the frequency domain, see Hermans et al. [6].

2.3 Stochastic Subspace Identification

Unlike the two previous methods the stochastic subspace identification method directly works with the recorded time signals. The peak-picking method requires frequency domain data while the polyreference LSCE method needs the correlation functions between time signals. It is beyond the scope of this paper to explain in full detail the stochastic subspace identification method. The interested reader is referred to Van Overschee et al. [7,8], Kirkegaard et al. [9] and Peeters et al.
for the theoretical background and applications in civil engineering. Here only the main ideas behind the method are given. The method assumes that the dynamic behaviour of a structure excited by white noise can be described by a stochastic state space model:

\[ x_{k+1} = Ax_k + w_k \]
\[ y_k = Cx_k + v_k \]  \hspace{1cm} (6)

where \( x_k \in \mathbb{R}^{n_p} \) is the internal state vector; \( n_p \) is the number of poles; \( y_k \in \mathbb{R}^{n_y} \) is the measurement vector and \( w_k, v_k \) are white noise terms representing process noise and measurement noise together with the unknown inputs; \( A \in \mathbb{R}^{n_p \times n_p} \) is the state matrix containing the dynamics of the system and \( C \in \mathbb{R}^{n_y \times n_p} \) is the output matrix, translating the internal state of the system into observations.

The subspace method then identifies the state space matrices based on the measurements and by using robust numerical techniques such as QR-factorization, Singular Value Decomposition (SVD) and least squares. Roughly, the QR results in a significant data reduction, whereas the SVD is used to reject the noise (assumed to be represented by the higher singular values). Once the mathematical description of the structure (the state space model) is found, it is straightforward to determine the modal parameters (by an eigenvalue decomposition): natural frequencies, damping ratios and mode shapes.

2.3 ARMAV Estimation using a Prediction Error Method

Just like the stochastic subspace identification method the Prediction Error Method for estimation of ARMAV models works directly with the recorded time signals. A detailed description of the Prediction Error Method is provided in Ljung [12] and Söderström et al. [13], and a comprehensive description of the use of ARMAV models in relation to civil engineering and mechanical applications is found in Andersen [14] and Pandit [15]. It can be shown that the ARMAV model can model the dynamics of a structure subjected to filtered white noise, see Andersen [14]. In other words, the only restrictions are that the structure behaves linearly and is time-invariant, and that the unknown input force can be modelled by a white noise filtered through a linear and time-invariant shaping filter. The definition of the ARMAV model is:

\[ y_k = A_1 y_{k-1} + \ldots + A_n y_{k-n} + B_0 e_k + B_1 e_{k-1} + \ldots + B_m e_{k-m} \] \hspace{1cm} (7)

where \( y_k \in \mathbb{R}^{n_y} \) is the measurement vector and \( e_k \in \mathbb{R}^{n_y} \) is a zero-mean white noise vector process. The auto-regressive matrix polynomial is described by the coefficient matrices \( A_i \in \mathbb{R}^{n_x} \). This polynomial models the dynamics of the combined system, i.e. the modes of the structural system combined with the noise modes. The moving average matrix polynomial is described by the coefficient matrices \( B_i \in \mathbb{R}^{n_y} \). This polynomial ensures that the statistical description of the data is optimal. It can be shown that by adding this moving average the covariance function of the predicted output \( \hat{y}_k \) of the ARMAV model will be equivalent to the covariance function of \( y_k \), see Andersen et al. [16]. The model order \( n \) depends on the number of modes as well as on the dimension of the measurement vector.

The ARMAV model is calibrated to the measured time signals by minimizing the prediction error \( y_k - \hat{y}_k \), i.e. the difference between the measured time signals and the predicted output of the ARMAV model. The criterion function \( V \) that is minimized is defined as, see Ljung [12] and Andersen et al. [17]:

\[ V = \det \left( \frac{1}{K} \sum_{k=1}^{K} [y_k - \hat{y}_k] [y_k - \hat{y}_k]^T \right) \] \hspace{1cm} (8)
This criterion function can be shown to correspond to a maximum likelihood if the prediction errors are Gaussian white noise, see Söderström et al. [13]. In this case the criterion provides maximum accuracy [Söderström]. The presence of the moving average makes it necessary to apply a non-linear optimization scheme. This minimization is started by providing an initial ARMAV model. In the present case this model is obtained by a stochastic subspace method, see Andersen [14]. Again, once the optimal ARMAV model is determined by a stabilization diagram, it is straightforward to determine the modal parameters by a modal decomposition.

3 HIGHWAY BRIDGE TEST CASE

The bridge, used as test object to illustrate the above mentioned system identification methods, is the Z24-bridge overpassing the national highway A1 between Bern and Zürich in Switzerland. It is a prestressed concrete box girder bridge with a main span of 30 m and two side spans of 14 m. The bridge is supported by 4 piers clamped into the girders. The two piers at the abutments are completely embedded in the ramps.

3.1 Description of the Acquired Data

In total, 145 points were measured, mainly in the vertical and transverse direction. This amounted to 172 degrees of freedom (DOFs). The same number of channels would have been necessary to measure all the DOFs at the same time. As a maximum of 23 channels were available the testing were divided into 9 setups. In each setup 19 different DOFs on the bridge were measured, along with 4 extra DOFs serving as references (3 in vertical direction and 1 in the transverse direction). These reference stations were measured again in each setup. The data were sampled at a rate of 80 Hz and the analogue anti-aliasing filter had a cut-off frequency at 20 Hz. A total of 65536 samples (13 min, 39.2 sec) was acquired for each channel and each setup. The ambient excitation sources of the bridge were wind and traffic on the highway. All setups were measured between 9 PM and midnight. More details concerning the bridge test can be found in Krämer et al. [18].

3.2 Practical Aspects of Using the Different Identification Techniques

For the peak-picking method, 16 segments of 4096 data points were transformed to the frequency domain and averaged to estimate the power-spectral densities. So all measured data were used in this method. By applying the procedures described in paragraph 2.1, estimates of the natural frequencies and mode shape parts were obtained. Every setup with 23 simultaneously recorded signals yields the mode shape at the corresponding 23 DOFs. The different parts were glued together using one of the reference sensors (the choice of the reference sensor depends on the nature of the mode shape).

The polyreference LSCE method was applied to the auto- and cross-correlations of the responses. For each setup, the correlations between all responses and 3 responses in the vertical direction serving as references were calculated using equation (1). The number of estimated time lags equalled 256 which corresponds to a duration of 3.2 sec. The correlation functions were then fed to the LSCE method in order to extract the natural frequencies and damping ratios. As the correlation functions of the different setups were referenced to the same 3 reference stations, they could be combined into one global model, yielding global estimates for the frequency and damping. Stabilisation diagrams showing the stability of the poles as function of increasing model order were used to distinguish the spurious modes from the physical ones. Next to this global analysis, the modal parameters were also separately extracted for each setup and a comparison of the modal
estimates was made. As the LSCE method does not yield the mode shapes, an additional step was needed. This was done by fitting the power- and cross-spectral densities between the responses and the selected reference stations in a least squares sense. The power- and cross-spectral densities were estimated on the basis of the DFT and segment-based averaging. The segment size equalled 2048 time points and 50% overlap was used. A Hanning window was used to reduce the leakage effects.

As the excitation was different for each setup, the mode shapes were separately identified for each setup and glued together via the 4 reference stations. For setup 5, the power-spectral densities of a few DOFs were difficult to fit, which leads to some irregularities in the animation of the mode shape. Also, for most setups, the fit was poor for frequencies higher than 12 Hz and consequently, the shape of the fifth mode (cf. next section) could not be extracted with high confidence.

For the stochastic subspace method it was not possible to treat all 65536 samples x 23 channels of one setup at once. The computational time and memory needed can increase to an inadmissible level with an increasing number of samples and channels. Therefore the analysis for all setups was limited to a high-quality segment of 4096 samples. If such a segment did not give satisfactory results, another segment was chosen afterwards to perform a new analysis. The number of time lags used in the method were 20; since there were 23 channels, the maximum number of singular values was 460 (20 x 23). Consecutive state space models of dimension 2 to 60 in steps of 2 were identified. From all these state space models, the modal parameters were extracted. Stabilisation diagrams were then used to distinguish the spurious modes from the physical ones. For every setup, seven modes could be identified.

To apply the prediction error method for estimation of ARMA models, an accurate initial estimate was needed. By supplying an accurate initial estimate the number of iterations needed was kept at a minimum and convergence was ensured. To provide such initial estimates, a subspace technique returning ARMA models was applied, see Andersen [14]. The modal parameters of interest of the initial ARMA models were then refined, one mode at a time, by minimizing (8) in modal space, see Brincker et al. [19]. In the subspace estimation the number of time lags used were 30. All available data were used, i.e. up to 65536 samples x 23 channels per setup. The orders of the applied ARMA models were in the range from n=1 to n=5. Again, due to the differences of the excitation from setup to setup, the mode shapes were separately identified for each setup and glued together via the 4 reference stations.

4 COMPARISON OF MODAL RESULTS

The results of the comparison are presented in this section for the two first modes only. More results might be found in Andersen et al. [20]

In the comparison of the mode shapes only the sensors located at the girder are included. These sensors are placed in three rows along the girder, which means that each mode shape can be represented by three curves. These three curves are plotted in the same figure for all four techniques. Below the plots the Modal Assurance Criterion between the four techniques is listed in a table. Also listed are the estimated natural frequencies, damping ratios, and standard deviations. The 1st mode is a vertical bending mode. In the 2nd mode, the piers are bending in the transverse direction and the girder is submitted to torsion.

In general, all methods seems to agree very well on the natural frequency estimates of the shown modes. Also for the remaining modes, the different techniques agreed quite well on all modal estimates. Even the damping ratio estimates correspond fairly well for three of the methods. The damping ratios have not been estimated in the peak-picking method.
As it appears from the results, the mode shape estimates correlates quite well for all the techniques, however, for the 2nd mode some differences are observed. The uncertainty is large around the points where the reference points are situated. This problem illustrates the importance of making a good choice for placement of the reference sensors.

**Figure 1.** Comparison of the first mode shape - all four methods.

<table>
<thead>
<tr>
<th></th>
<th>PP</th>
<th>LSCE</th>
<th>SSI</th>
<th>PEM</th>
</tr>
</thead>
<tbody>
<tr>
<td>PP</td>
<td>1</td>
<td>0.999</td>
<td>0.999</td>
<td>0.999</td>
</tr>
<tr>
<td>LSCE</td>
<td>0.999</td>
<td>1</td>
<td>0.999</td>
<td>0.999</td>
</tr>
<tr>
<td>SSI</td>
<td>0.999</td>
<td>0.999</td>
<td>1</td>
<td>1</td>
</tr>
<tr>
<td>PEM</td>
<td>0.999</td>
<td>0.999</td>
<td>1</td>
<td>1</td>
</tr>
</tbody>
</table>

Table 1: Modal Assurance Criterion of the 1st mode shape.

<table>
<thead>
<tr>
<th></th>
<th>( f_i ) [Hz]</th>
<th>( \zeta_i ) [%]</th>
<th>( \nu_i ) [Hz]</th>
<th>( \varsigma_i ) [%]</th>
</tr>
</thead>
<tbody>
<tr>
<td>PP</td>
<td>3.96</td>
<td>-</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>LSCE</td>
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<td>1.0</td>
<td>0.01</td>
<td>0.2</td>
</tr>
<tr>
<td>SSI</td>
<td>3.93</td>
<td>1.1</td>
<td>0.02</td>
<td>0.5</td>
</tr>
<tr>
<td>PEM</td>
<td>3.95</td>
<td>1.1</td>
<td>0.01</td>
<td>0.3</td>
</tr>
</tbody>
</table>

Table 2: The natural frequencies and damping ratios of the 1st mode.

<table>
<thead>
<tr>
<th></th>
<th>( f_i ) [Hz]</th>
<th>( \zeta_i ) [%]</th>
<th>( \nu_i ) [Hz]</th>
<th>( \varsigma_i ) [%]</th>
</tr>
</thead>
<tbody>
<tr>
<td>PP</td>
<td>5.27</td>
<td>-</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>LSCE</td>
<td>5.23</td>
<td>1.8</td>
<td>0.02</td>
<td>0.3</td>
</tr>
<tr>
<td>SSI</td>
<td>5.22</td>
<td>1.4</td>
<td>0.02</td>
<td>0.3</td>
</tr>
<tr>
<td>PEM</td>
<td>5.24</td>
<td>1.7</td>
<td>0.02</td>
<td>0.5</td>
</tr>
</tbody>
</table>

Table 3: Modal Assurance Criterion of the 2nd mode shape.

**Figure 2.** Comparison of the second mode shape - all four methods.
The advantages of the peak-picking method are that it is easy to use and provides fast estimates. However, the damping has not been estimated, and since no parametric model is calibrated, the mode shapes are in fact only operational deflection shapes. In the present case, where all modes are well-separated, deflection shapes seem to approximate the mode shapes well. The advantage of the LSCE method is its ability to identify modal parameters globally, even when data is divided into multiple setups. In the present case, it is seen to provide sound estimates of the natural frequencies and damping ratios that are comparable with the two other time domain methods. Since the mode shapes are estimated in frequency domain they are more comparable with the peak-picking method than the two time domain methods. The time domain methods have the advantages of operating directly on the measured time signals. However, they are a bit more complicated to use, and more time consuming. Different model orders have to be evaluated in order to determine the optimal one. However, stabilisation diagrams and other model validation techniques can be of aid to the user. The SSI method solves the time and memory problem by reducing the amount of data used in the analysis. The PEM method and the initial subspace estimator of ARMA models both use all available data. However, due to the high quality data this does not seem to improve the modal parameter estimates significantly.

REFERENCES