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APPENDICES

Α



Appendix A

Complex Laplace-domain transfer functions from state-space equations

Consider the state and output equations of an LTI system (equations 2.5.1.19):

$$\dot{x} = Ax + Bu \tag{A.1a}$$

$$y = Cx + Du (A.1b)$$

The Laplace-transform of equation A.1a is:

$$sX(s) - x(0) = AX(s) + BU(s)$$
 (A.2)

For x(0) = 0, equation A.2 becomes:

$$sX(s) = AX(s) + BU(s)$$
(A.3)

Solving equation A.3 for X(s) gives:

$$X(s) = [sI - A]^{-1}BU(s)$$
(A.4)

The Laplace-transform of equation A.1b is:

$$Y(s) = CX(s) + DU(s)$$
(A.5)

Substitution of equation A.4 into equation A.5 gives:

$$Y(s) = (C[sI - A]^{-1}B + D)U(s)$$
(A.6)

The transfer function G(s) is the ratio of the output Y(s) to the input U(s):

$$G(s) = \frac{Y(s)}{U(s)} = C[sI - A]^{-1}B + D$$
 (A.7)



Appendix B

Exact solution methods for equations of motion of nonlinear systems

Consider the equation of motion for natural behaviour of an undamped SDOF system with a general, nonlinear restoring force g(y), as governed by the DE:

$$m\ddot{y} + \kappa g(y) = 0 \tag{B.1}$$

where m is the mass, κ is a constant, y and \ddot{y} are the mass displacement and acceleration respectively.

Equation B.1 can be written as:

$$\frac{d(\dot{y}^2)}{dy} + 2\frac{\kappa}{m}g(y) = 0$$
 (B.2)

The time versus displacement relationship is obtained by integrating equation B.2 twice:

$$t - t_0 = \frac{1}{\sqrt{2\frac{\kappa}{m}}} \int_0^y \frac{d\chi_2}{\sqrt{\int_{\chi_2}^y g(\chi_1) d\chi_1}}$$
 (B.3)

where t is time, t_0 corresponds to the time when y = 0, Y is the displacement when $\dot{y} = 0$ and χ_1 and χ_2 are integration variables. The displacement versus time relationship may be obtained by inverting equation B.3.

For an odd nonlinear restoring force, i.e. g(-y) = -g(y), the natural period of vibration τ_n is given by:

$$\tau_n = \frac{4}{\sqrt{2\frac{\kappa}{m}}} \int_0^{\gamma} \frac{d\chi_2}{\sqrt{\int_{\chi_2}^{\gamma} g(\chi_1) d\chi_1}}$$
 (B.4)

Exact solutions can be obtained in all cases where the integrals in equation B.4 can be explicitly expressed in terms of Y. For pure powers of displacement g(y) = y'', τ_{ii} is given by:

$$\tau_{n} = \frac{4}{\sqrt{\frac{\kappa}{m}Y^{n-1}}}\psi(n) \tag{B.5}$$

where $\psi(n)$ is a function whose values are given in table B.1, for n ranging from 0 to 7.



Table B.1: Values of the function $\psi(n)$ [Harris, 1988]

[HAITIS, 1700]			
n	$\psi(n)$		
0	1,4142		
1	1,5708		
2	1,7157		
3	1,8541		
4	1,9818		
5	2,1035		
6	2,2186		
7	2,3282		

Solutions for polynomials of displacement and velocity-squared damping are given by Harris [1988]. Exact solutions for forced vibration of nonlinear systems are virtually nonexistent, except if the system can be represented in a stepwise linear manner [Harris, 1988].



Appendix C The perturbation method

Consider the following equation of motion for natural behaviour of a weakly nonlinear SDOF system:

$$\underbrace{\ddot{y} + \omega_0^2 y}_{linear} + \underbrace{\frac{\kappa}{m} g(\dot{y}, y)}_{nonlinear} = 0$$
 (C.1)

where ω_0 is the linear system natural frequency, κ is a small parameter and g is a nonlinear function of speed \dot{y} and displacement y. The first two terms on the left hand side of equation C.1 describe the behaviour of the linear part of the system, while the third, or perturbation term, accounts for the effects of the weak nonlinearity.

A solution of the following form is assumed for y [Gelb & VanderVelde, 1968]:

$$y(t) = y_0(t) + \frac{\kappa}{m} y_1(t) + \left(\frac{\kappa}{m}\right)^2 y_2(t) + \dots +$$
 (C.2)

where y_0, y_1, y_2, \dots , are displacement functions of time. These functions are obtained by solving the following system of linear, non-homogeneous, 2^{nd} order DE's:

$$\ddot{y}_0 + \omega_0^2 y_0 = 0 ag{C.3a}$$

$$\ddot{y}_1 + \omega_0^2 y_1 = -g(\dot{y}_0, y_0)$$
 (C.3b)

$$\ddot{y}_2 + \omega_0^2 y_2 = -\frac{\partial g}{\partial y} (\dot{y}_0, y_0) y_1 - \frac{\partial g}{\partial \dot{y}} (\dot{y}_0, y_0) \dot{y}_1$$

$$\vdots \qquad \vdots \qquad \vdots \qquad \vdots \qquad \vdots$$
(C.3c)

Equations C.3a to C.3b are solved recursively. The first equation (C.3a) is solved for y_0 and \dot{y}_0 . With y_0 and \dot{y}_0 known, g, $\partial g/\partial y$ and $\partial g/\partial \dot{y}$ are obtained. Equation C.3b is subsequently solved for y_1 and \dot{y}_1 , whereafter equation C.3c is solved for y_2 and \dot{y}_2 . The solutions of y_0 , y_1 and y_2 are substituted into equation C.2 to obtain y(t). The procedure can be repeated as many times as the number of terms required in the series expansion in equation C.2.



Appendix D The method of slowly varying amplitude and phase

Consider the following equation of motion for natural behaviour of a weakly nonlinear SDOF system:

$$\underbrace{\ddot{y} + \omega_0^2 y}_{linear} + \underbrace{\frac{\kappa}{m} g(\dot{y}, y)}_{nonlinear} = 0$$
 (D.1)

where ω_0 is the linear system natural frequency, κ is a small parameter and g is a nonlinear function of speed \dot{y} displacement y.

A solution of the following form is assumed for y [Gelb & VanderVelde, 1968]:

$$y(t) = A\sin\psi \tag{D.2}$$

where A and ψ are functions of time, given by:

$$A = A(t)$$

$$\psi = \omega_0 t + \theta(t)$$
(D.3a)
(D.3b)

$$\psi = \omega_0 t + \theta(t) \tag{D.3b}$$

The following set of approximate DE's in A and θ can be derived [Gelb & VanderVelde, 1968]:

$$\dot{A} \approx -\frac{1}{2\pi\omega_0} \int_0^{2\pi} \mu g(A\sin\psi, A\omega_0\cos\psi)\cos\psi d\psi$$
 (D.4a)

$$\dot{\theta} \approx \frac{1}{2\pi\omega_0} \int_0^{2\pi} \frac{1}{A} \mu g(A\sin\psi, A\omega_0\cos\psi)\cos\psi d\psi$$
 (D.4b)

Equations D.4 are solved as follows: The right hand side of equation D.4 a is integrated with respect to ψ , resulting in a 1st order separable DE in A. This DE is solved and its solution $A = \underline{A}$ is obtained. Subsequently, the right hand side of equation D.4b is integrated with respect to ψ and another 1st order DE is obtained. Replacing A by \underline{A} , the latter DE is solved to obtain its solution $\theta = \theta$.

The solution of equation D.1 is then:

$$y(t) = \underline{A}(t)\sin[\omega_0 t + \underline{\theta}(t)]$$
 (D.5)



Appendix E Statistical linearization

Consider a general nonlinear SDOF system with an equation of motion of the form

$$g(\ddot{y},\dot{y},y) = f(t) \tag{E.1}$$

where g is the total internal force and f is the stationary gaussian random excitation force with zero mean. It is assumed that a stationary solution to equation E.1 exists.

The linear equivalent of equation E.1 is:

$$m\ddot{y} + c\dot{y} + ky = f(t) \tag{E.2}$$

where m, c and k are the mass, damping coefficient and spring stiffness to be determined, such that the solution of equation E.2 will give an approximate solution to equation E.1.

In approximating a nonlinear system by a linear system, an error will be made. The error may be defined as the difference between the solutions of the two systems, an can be written as:

$$e = g(\ddot{y}, \dot{y}, y) - m\ddot{y} - c\dot{y} - ky$$
(E.3)

The mass, damping coefficient and stiffness are determined by setting the mean square (MS) error equal to zero, resulting in:

$$m = E\left[\frac{\partial g(\ddot{y}, \dot{y}, y)}{\partial \ddot{y}}\right]$$
 (E.4a)

$$c = E\left[\frac{\partial g(\ddot{y}, \dot{y}, y)}{\partial \dot{y}}\right]$$
 (E.4b)

$$k = E\left[\frac{\partial g(\ddot{y}, \dot{y}, y)}{\partial y}\right]$$
 (E.4c)

where $E[\]$ is the expected value of the term in brackets.

The method of equivalent linearization can be applied to a large variety of nonlinear system types. Examples are systems with cubic stiffening characteristics [Atalik & Utku, 1976], nonlinear damping characteristics [Atalik & Utku, 1976], hysteretic restoring forces [Wen, 1980], the Duffing oscillator [Budgor et al, 1976] and equations of motion of articulated vehicles [ElMadany & Dokainish, 1980]. Application of the method to a system with hysteretic restoring forces, as described by Wen [1980], is discussed next.



Statistical linearization applied to a system with hysteretic restoring forces

Consider a system whose equation of motion is of the form

$$m\ddot{\overline{y}} + g(\dot{\overline{y}}, \overline{y}) + z(\overline{y}) = \overline{f}(t)$$
 (E.5)

where \overline{y} is the dimensionless displacement and \overline{f} is the dimensionless force:

$$\overline{y} = \frac{y}{Y}$$
 (E.6a)

$$\bar{f} = \frac{f}{F_{\gamma}} \tag{E.6b}$$

In equation E.5, g is the nonhysteretic component of the restoring force, while z is the hysteretic component.

The relationship between z and y is given by the following 1^{st} order nonlinear DE:

$$\dot{z} = -\gamma \left| \dot{y} \right| z \cdot \left| z \right|^{n-1} - \beta \dot{y} \left| z \right|^{n} + G \dot{y}$$
(E.7)

where γ and β are parameters which control the shape of the hysteresis loop, G is the restoring force amplitude and n is an exponent which describes the smoothness of transition from elastic to plastic response.

From Wen [1980], equations E.5 and E.7 can be written in linear state-space form as

$$\dot{x} = Ax + Bu \tag{E.8}$$

where x and u are the state vector and input respectively:

$$x = \{\overline{y}, \quad z, \quad \dot{\overline{y}}\}^T$$

$$u = \overline{f}(t)$$
(E.9a)
(E.9b)

$$u = \bar{f}(t) \tag{E.9b}$$

A and B are the coefficient and driving matrices respectively, given by:

$$A = \begin{bmatrix} 0 & 0 & 0 \\ 0 & -k_h & -c_h \\ -\alpha\omega_0^2 & -(1-\alpha)\omega_0^2 & -2\zeta_0\omega_0 \end{bmatrix}$$
 (E.10a)

$$B = [0, 0, 1/m]^T$$
 (E.10b)

In equation E.10a, α is the post-to pre-yielding stiffness ratio, ζ_0 is the viscous damping ratio



and ω_0 is the pre-yielding natural frequency:

$$\omega_0 = \sqrt{\frac{F_Y}{mY}}$$
 (E.11)

 k_h and c_h are the linearized hysteresis stiffness and damping coefficient. Both k_h and c_h depend on n. For the special case of n = 1, k_h and k_h are given by:

$$k_h = \gamma E[|\dot{y}|] + \beta E[\dot{y}\frac{\partial |z|}{\partial z}]$$
 (E.12a)

$$c_h = \gamma E \left[z \frac{\partial |\dot{y}|}{\partial y} \right] + \beta E \left[|z| \right] - G$$
 (E.12b)

It can be seen that the highly nonlinear system, as described by equations E.5 and E.7, is replaced by the equivalent linear system, as represented by equations E.8 to E.12. By comparison, the linear equations are considerably easier to solve than the nonlinear equations. Linear solutions of the above hysteretic system to gaussian inputs are given by Wen [1980].



Appendix F The Describing Function (DF) method

Consider a nonlinear system driven by a harmonic input of the form

$$u(t) = U \sin \omega t \tag{F.1}$$

where u is the input as a function of time and U is the input amplitude.

The output y(t), which is generally non-sinusoidal, can be expressed as follows by means of a Fourier series:

$$y(t) = \frac{a_0}{2} + \sum_{i=1}^{\infty} \left[a_i \cos(i\omega t) + b_i \sin(i\omega t) \right]$$
 (F.2)

where the Fourier coefficients are a_i and b_i are functions of U and ω , determined by:

$$a_0 = \frac{1}{\pi} \int_{\pi}^{\pi} y(t) d(\omega t)$$
 (F.3a)

$$a_i = \frac{1}{\pi} \int_{-\pi}^{\pi} y(t) \cos(i\omega t) d(\omega t)$$
 (F.3b)

$$b_i = \frac{1}{\pi} \int_{-\pi}^{\pi} y(t) \sin(i\omega t) d(\omega t)$$
 (F.3c)

For an odd nonlinearity, the coefficient a_0 is zero:

$$a_0 = 0 ag{F.4}$$

In harmonic DF analysis, only the fundamental frequency is considered. Therefore, y(t) can be approximated by:

$$y(t) \approx y_1(t) = a_1 \cos \omega t + b_1 \sin \omega t$$
 (F.5)

The output y(t) is written in terms of the output amplitude and phase of the fundamental frequency as:

$$y_1(t) = Y \sin(\omega t + \phi)$$
 (F.6)

where Y are the output amplitude and phase respectively, as functions of input amplitude and frequency:

$$Y = Y(U, \omega) \tag{F.7a}$$

$$\phi = \phi(U, \omega) \tag{F.7b}$$



In terms of the fundamental Fourier coefficients, Y and ϕ are:

$$Y = \sqrt{a_1^2 + b_1^2}$$
 (F.8a)

$$\phi = \tan^{-1} \left(\frac{a_1}{b_1} \right) \tag{F.8b}$$

In complex representation, y_1 can be written as:

$$y_1 = Ye^{j(\omega t + \phi)} \tag{F.9}$$

The describing function N of a nonlinear system is the ratio of the fundamental component of the output, to the input:

$$N = \frac{Ye^{j(\omega t + \phi)}}{Ue^{j\omega t}}$$
 (F.10)

where N is a function of the input amplitude and frequency:

$$N = N(U, \omega) \tag{F.11}$$

In rectangular coordinates, N can be written as:

$$N(U,\omega) = \frac{1}{U}(b_1 + ja_1)$$
 (F.12)

The above derivation is only applicable to harmonically excited nonlinear systems. The DF method can also be extended to analyze nonlinear system behaviour to non-harmonic inputs, like two-sinusoid inputs, dual-inputs (e.g. DC plus sinusoid), transient and random inputs. N is tabled for a vast range of nonlinearities, for each of the above input types, by Gelb & Vander Velde [1968]. Examples of nonlinearities applicable to the analysis of Terfenol-D characteristics, are saturation and hysteresis. The DF's of these nonlinearities can be found in Gelb & Vander Velde [1968].



Appendix G Fifth order Runge-Kutta (R-K) method

In the R-K methods, the states and outputs are obtained in the time domain by direct integration of the state equations. The solution of each state equation is approximated by a polynomial. The order of the polynomial can vary from 1st order to 5th order and higher. The higher the order, the more accurate the solution for the same computational effort [Chapra & Canale, 1985]. For this reason, higher order methods are often preferred to lower order methods.

The R-K methods are discussed in detail by Burden & Faires [1985], Chapra & Canale [1985], Conte & de Boor [1972], Gerald & Wheatley [1984] and Press et al [1992]. A short description of the 5th order method, also known as Butcher's method, is given below.

Consider a nonlinear time-invariant system, whose state and output equations are of the following form (see equations 2.5.1.2a and 2.5.1.2b):

$$\dot{x} = f(x, u) \tag{G.1a}$$

$$y = g(x, u) \tag{G.1b}$$

Equations G.1a and G.1b are solved by dividing time into finite intervals, and by recursively calculating the states and outputs for successive intervals using simple algebra.

Time at the beginning of the *i*-th time step is denoted by t_i . The state and output equations corresponding to t_i are:

$$\dot{x}_i = f(x_i, u_i) \tag{G.2a}$$

$$y_i = g(x_i, u_i)$$
 (G.2b)

where x_i , u_i and y_i respectively represent the state vector, input and output at the beginning of the *i*-th time step.

The state vector x_{i+1} , at the beginning of the i+1-th time step, is obtained by the following equation:

$$x_{i+1} = x_i + \frac{\Delta t}{90} \left(7k_1 + 32k_3 + 12k_4 + 32k_5 + 7k_6 \right)$$
 (G.3)

where Δt is the time step, given by:

$$\Delta t = t_{i+1} - t_i \tag{G.4}$$

 Δt may be constant, or may be adjusted during simulation to accommodate changing stiffness of the system. Generally, the smaller Δt , the more accurate the solutions.

Variables k_1 to k_6 in equation G.3 represent the derivatives of the state vector at intermediate



intervals between t_i and t_{i+1} . Note that k_2 does not appear in equation G.3. It is a "dummy" variable, which is only used to determine k_3 , k_4 and k_6 .

Variables k_1 to k_6 are determined recursively by means of the following equations:

$$k_1 = f(t_i, x_i) \tag{G.5a}$$

$$k_2 = f\left(t_i + \frac{\Delta t}{4}, \quad x_i + \frac{\Delta t}{4}k_1\right)$$
 (G.5b)

$$k_3 = f\left(t_i + \frac{\Delta t}{4}, \quad x_i + \frac{\Delta t}{8}k_1 + \frac{\Delta t}{8}k_2\right)$$
 (G.5c)

$$k_4 = f\left(t_i + \frac{\Delta t}{2}, \quad x_i - \frac{\Delta t}{2}k_2 + (\Delta t)k_3\right)$$
 (G.5d)

$$k_5 = f\left(t_i + \frac{3\Delta t}{4}, \quad x_i + \frac{3\Delta t}{16}k_1 + \frac{9\Delta t}{16}k_4\right)$$
 (G.5e)

$$k_6 = f\left(t_i + \Delta t, \quad x_i - \frac{3\Delta t}{7}k_1 + \frac{2\Delta t}{7}k_2 + \frac{12\Delta t}{7}k_3 - \frac{12\Delta t}{7}k_4 + \frac{8\Delta t}{7}k_5\right)$$
(G.5f)

The output y_{i+1} for the i+1-th time step is given by:

$$y_{i+1} = g(x_{i+1}, u_{i+1})$$
 (G.6)



Appendix H Actuator nonlinear dehysterized transfer functions

In order to obtain the nonlinear, dehysterized TF's, three sets of simulations are done, one for each of the input RMS values, i.e. 2,18 V, 1,63 V and 1,09 V. The nonlinear state equations (equations 2.5.1.11 and 2.5.1.12) are solved in the time-domain by means of the 5^{th} order Runge-Kutta method. The states are the actuator displacement x_a , speed \dot{x}_a and coil current I (see equation 2.5.1.4). The input is the coil voltage V. The input with the highest RMS value, i.e. 2,18 V, is applied first. The initial values of the states are set equal to zero.

For each time-step, stress and field are calculated from the states, using equations 2.4.4 and 2.4.10. The magnetostrictive and magnetization parameters, i.e. permeability μ^{σ} , Young's modulus E, piezomagnetic cross-coupling constant d^{σ} and strain constant d^{H} , are calculated from stress and field, using equations 2.5.1.10a to 2.5.1.10d. The state equations are integrated using equations G.3 and G.5a to G.5f (see appendix G), thereby obtaining the state vector at the end of each time step. The outputs are calculated in terms of the states at the end of each time step, using equation G.6. At the beginning of a new time-step, input voltage is updated and the procedure as set out in this paragraph, is repeated. At the end of the simulation, outputs for all the time steps are stored. The procedure is repeated for the 1,63 V and 1,09 V RMS inputs.

The nonlinear dehysterized transfer functions for the three coil input voltages, in terms of the inputs and outputs, are obtained by means of the following equation:

$$G(j\omega) = \frac{P_{xy}}{P_{yy}}$$
 (H.1)

where $G(j\omega)$ is the transfer function between the coil input voltage and actuator displacement output, P_{xy} is the cross-spectral density (CSD) between the input and output and P_{xx} is the power spectral density (PSD) of the input. Since the displacement signal is a simulation output, no measurement noise is present, resulting in high coherence.

 $G(j\omega)$ in equation H.1 is complex, with a frequency-dependent magnitude and phase angle. The transfer function magnitude and phase, for a 2,18 V RMS coil input voltage, are shown in figure H.1.

The DC magnitude is $13,57 \mu m/V$. The resonance frequency is 375 Hz. The maximum magnitude, at resonance, is $21,88 \mu m/V$. The ratio between the maximum and DC magnitudes is 1,61. The -3dB bandwidth, i.e. where the magnitude is 70,7% of the DC value, is 482 Hz. The TF phase decreases linearly from 0 Hz to approximately 300 Hz. This reduction in phase can be attributed to coil inductance. At 375 Hz, phase decreases more rapidly with frequency. This phase decrease is due to resonance. Above 482 Hz, the actuator is operated in its filtering range, where the slope of the phase is approximately equal to that for frequencies below 300 Hz.



Similar *TF* spectra to figure H.1, exist for the 1,63 V and 1,09 V RMS coil input voltages. These spectra will however not be shown because they correspond almost exactly with the *TF* shown in figure H.1. Instead, the maximum and static magnitudes and their ratio, resonance frequency and -3 dB bandwidth for the three input voltages, are given in table H.1.

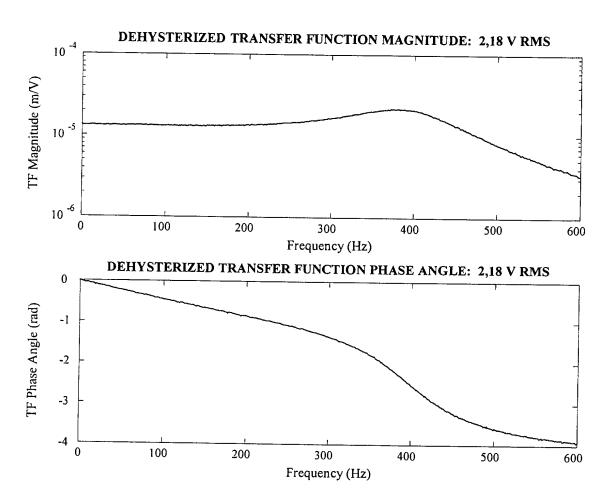


Figure H.1: Dehysterized transfer function for a 2,18 V RMS coil input voltage

Table H.1: Nonlinear dehysterized static & dynamic transfer function parameters

CHICAGO CONTRACTOR CON	a difference of a gradule transfer function parameters			
Parameter	2,18 V	1,63 V	1.09 V	
Static magnitude (TF _S):	13,57 μm/V	13,68 μm/V	13,76 μm/V	
Maximum magnitude (TF_R) :	21,88 μm/V	21,95 μm/V	21,99 µm/V	
TF_R/TF_S :	1,61	1,60	1,60	
Resonance frequency:	375 Hz	375 Hz	375 Hz	
-3 dB bandwidth:	483 Hz	482 Hz	481 Hz	

It can be seen from table H.1 that the static and dynamic parameters are, for all practical purposes, independent of input voltage. The dehysterized characteristics can therefore be considered linear and can be expressed in terms of constant magnetostrictive and magnetization parameters, i.e. μ^{σ} , E, d^{σ} and d^{H} . The conclusion can be drawn that, for the



voltage inputs used in the simulations, the actuator is excited inside its linear range, i.e. outside its saturation range (see figure 2.2.1.6). The linear parameters are calculated in appendix J and their effects on the linear TF are discussed.



Appendix J Actuator linear dehysterized characteristics

An equivalent linear TF of the actuator, using constant magnetostrictive and magnetization parameters, is obtained in this section. The linear TF magnitude and phase are calculated from the average parameters and compared with the nonlinear TF magnitude and phase. It is shown that the linear phase spectra closely match the nonlinear spectra, but that the magnitude spectra differ, especially for frequencies above the resonance frequency.

The average values of μ^{σ} , E, d^{σ} and d^{μ} are calculated as follows from the nonlinear time-domain simulation outputs:

$$\mu^{\sigma} = \mu_m^{\sigma} = \frac{1}{T} \int_{t=0}^{t=T} \mu^{\sigma}(t) dt$$
 (J.1a)

$$E = E_m = \frac{1}{T} \int_{t=0}^{t=T} E(t) dt$$
 (J.1b)

$$d^{\sigma} = d_m^{\sigma} = \frac{1}{T} \int_{t=0}^{t=T} d^{\sigma}(t) dt$$
 (J.1c)

$$d^{H} = d_{m}^{H} = \frac{1}{T} \int_{t=0}^{t=T} d^{H}(t) dt$$
 (J.1d)

where t is time, T is the maximum time and the subscript m denotes the average value of the particular parameter.

The average parameters for the three coil input voltages, using equations J.1a to J.1d, with T = 40 s, are given in table J.1.

Table J.1: Average linear dehysterized magnetostrictive & magnetization parameters

Parameter	2,18 V RMS	1,63 V RMS	1,09 V RMS
Permeability μ^{σ} :	6,962 μTm/A	6,969 μTm/A	6,977 μTm/A
Young's modulus <i>E</i> :	23,46 GPa	23,36 GPa	23,28 GPa
Piezomagnetic cross-coupling constant d^{σ} :	1,088.10 ⁻⁸ m/A	1,089.10 ⁻⁸ m/A	1,09.10 ⁻⁸ m/A
Strain constant d^{H} :	1,204.10 ⁻⁸ m/A	1,208.10 ⁻⁸ m/A	1,21.10 ⁻⁸ m/A

The parameters tabled above and the actuator parameters (m_a , G, R_c , N, l_T and A_T , from table 2.7.3.1), are used to calculate k_a and ω_n (equations 2.4.22 and 2.4.23) and L_f , cf and L_0 (equations 2.4.45 to 2.4.47). For the dehysterized characteristics, zero damping is used. The Laplace-domain transfer functions are calculated from equations 2.5.2.1 to 2.5.2.3. The frequency-domain transfer function magnitude and phase angle are calculated from equation 2.5.3.6a and 2.5.3.6b respectively.



The linear dehysterized *TF* magnitude and phase, for a 2,18 V RMS coil input voltage, are shown in figure J.1. For comparison purposes, the nonlinear *TF* magnitude and phase are also shown. It can be seen from figure J.1 that the *TF* phase obtained from the linear average parameters is virtually identical to that obtained from the nonlinear simulations. The linear and nonlinear *TF* magnitudes compare well for frequencies below approximately 300 Hz. However, for frequencies above 300 Hz, the magnitudes differ. The difference increases with an increase in frequency.

The difference in magnitudes can be attributed to the fact that the *TF* calculated from average parameters does not include the effects of superharmonics. The result would be the same if the harmonic balance technique were used. The nonlinear *TF*, on the other hand, was obtained using the 5th order Runge-Kutta method. Using this method, at least one superharmonic for all frequencies, up to 50% of the Nyquist frequency, is included in the output (see section 2.7.1).

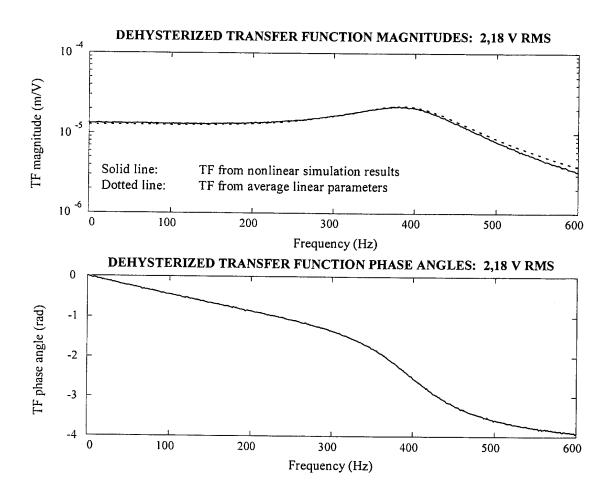


Figure J.1: Linear and nonlinear dehysterized TF's for a 2,18 V RMS input

The linear, dehysterized TF parameters for an input voltage of 2,18 V RMS, using the average values of μ^{σ} , E, d^{σ} and d^{H} in table J.1, are given in table J.2. For comparison purposes, the nonlinear TF parameters, from table H.1, are also given.



Table J.2: Linear and nonlinear TF characteristics for a 2,18 V RMS input

Nonlinear	Linear	
13,57 μm/V	13,00 μm/V	
21,88 μm/V	23,52 μm/V	
1,61	1,81	
375 Hz	385 Hz	
483 Hz	500 Hz	
	Nonlinear 13,57 μm/V 21,88 μm/V 1,61 375 Hz	

A comprehensive list of the linear characteristics, for each input voltage, is given in table J.3.

Table J.3: Linear characteristics for 2,18 V, 1,63 V and 1,09 V RMS input voltages

Parameter	2,18 V RMS	1,63 V RMS	1,09 V RMS
p_0	1,400.10 ⁵	1,397.105	1,392.10 ⁵
q_0	$1,077.10^{10}$	$1,071.10^{10}$	$1,065.10^{10}$
q_1	7,754.10 ⁶	$7,715.10^6$	$7,683.10^6$
q_2	2486	2482	2476
q_3	1	1	1
$\stackrel{\mu^{\sigma}}{E}$	6,962.10 ⁻⁶ Tm/A	6,969.10 ⁻⁶ Tm/A	6,977.10 ⁻⁶ Tm/A
\boldsymbol{E}	23,46 GPa	23,36 GPa	23,28 GPa
d^{σ}	1,088.10 ⁻⁸ A/m	$1,089.10^{-8} \text{ A/m}$	$1,090.10^{-8}$ A/m
d^H	1,204.10 ⁻⁸ A/m	$1,208.10^{-8}$ A/m	$1,210.10^{-8}$ A/m
f_n	331,3 Hz	330,6 Hz	330,0 Hz
f _{deff}	397,6 Hz	396,6 Hz	395,8 Hz
Seff	15,9%	15,9%	15,8%
Seff L_f	2,304 mH	2,306 mH	2,309 mH
cf	0,6643	0,6640	0,6635
L_o	1,287 mH	1,289 mH	1,292 mH
R_c/L_o	395,7 Hz	395,0 Hz	394,12 Hz
F_I	27,03 N/A	27,01 N/A	26,98 N/A

Note that the effective damped natural frequency of the actuator *exceeds* the undamped natural frequency. The reason is that f_n is the uncoupled (mechanical) natural frequency of the actuator, as given by equation 2.4.22, while f_{deff} is the imaginary part of the complex pole of the coupled TF (see equations 2.6.11 and 2.6.12a).

Table J.3 shows that the dehysterized parameters vary slightly for different RMS inputs. This is however not the case if hysteresis effects are taken into account (see appendix K).



Appendix K Actuator linear characteristics, including hysteresis effects

Actuator linear characteristics, including hysteresis effects, are calculated using the iterative harmonic balance technique. The technique is explained in more detail in section 2.7.2. The permeability μ^{σ} , strain constant d^{μ} and damping coefficient c, as well as the static and dynamic parameters, are input voltage- and frequency-dependent. The effects of hysteresis on E and d^{σ} are insignificant in comparison with its effect on μ^{σ} and d^{μ} (see section 2.3) and are therefore neglected.

Harmonic voltages, whose RMS-values correspond with those of the random voltage inputs used in the time-domain simulations, are used as inputs. This is done to allow a direct comparison between the dehysterized and hysteresis characteristics. For each voltage amplitude and frequency, the field amplitude H_A , permeability μ^{σ} , strain constant d^H and damping coefficient c, are calculated. The values of E and d^{σ} , for the appropriate voltage input, are obtained from table J.3 (see appendix J).

The frequency band is 0,01 Hz to 600 Hz. Frequency resolution varies according to the slope of the magnitude curve with respect to frequency. At low frequency, where the slope is low, a low resolution is used. As the slope increases, resolution is increased.

 μ^{σ} , E, d^{σ} and d^{H} , together with m_a , G, R_c , N, l_T and A_T (from table 2.7.3.1), are used to calculate k_a and ω_n (equations 2.4.22 and 2.4.23), L_f , cf and L_0 (equations 2.4.45 to 2.4.47). The dimensionless damping factor ζ is obtained from equation 2.4.25. The TF numerator and denominator polynomial coefficients are calculated by means of equation 2.5.2.4, table 2.5.2.1 and equations 2.5.2.5. The transfer function magnitude and phase for each frequency are calculated from equation 2.5.3.6a and 2.5.3.6b respectively.

The field strength frequency spectrum for a 2,18 V RMS input is shown in figure K.1. The field amplitude at 0,01 Hz is 17,62 kA/m. From 0,01 Hz to 328 Hz, field amplitude decreases. At 328 Hz, a gap appears in the spectrum, where the iterative harmonic balance technique fails to give convergence. The frequency range of non-convergence is 328 Hz to 332 Hz. The gap is situated at the minimum field amplitude (4156 A/m). For frequencies above 332 Hz, field increases until a local peak is reached at 440 Hz. From 440 Hz to 600 Hz, field amplitude decreases with an increase in frequency.

Similar spectra can be shown for the 1,63 V RMS and 1,09 V RMS inputs. The ranges of non-convergence for these inputs are 327 Hz to 331 Hz and 324 Hz to 331 Hz respectively.



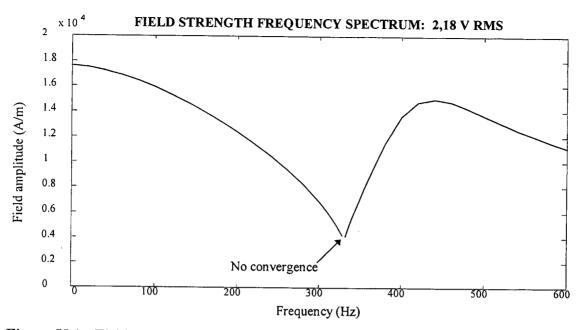


Figure K.1: Field strength frequency spectrum, including hysteresis: 2,18 V RMS

The frequency spectrum of permeability, including hysteresis, for a 2,18 V RMS input voltage, is shown in figure K.2. The permeability at 0,01 Hz is 5,57 μ Tm/A. Permeability decreases from 0,01 Hz to 328 Hz, where μ^{σ} is 0,62 μ Tm/A. For frequencies from 328 Hz to 332 Hz, permeability cannot be calculated, due to the convergence problem explained above. Above 332 Hz, permeability increases with frequency until a local peak is reached at 440 Hz. For frequencies above 440 Hz, permeability decreases with an increase in frequency.

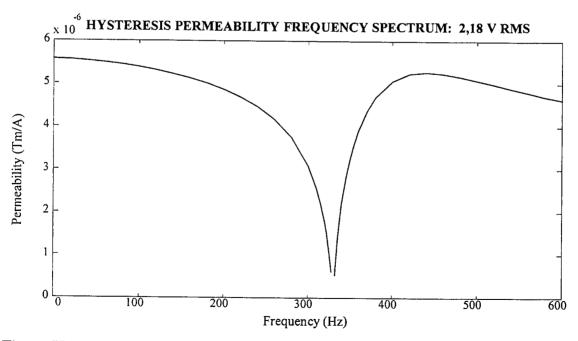


Figure K.2: Permeability frequency spectrum, including hysteresis: 2,18 V RMS



The frequency spectrum of the strain constant, including hysteresis, for a 2,18 V RMS input voltage, is shown in figure K.3. The strain constant at 0,01 Hz is $9,52.10^{-9}$ m/A. Strain constant decreases from 0,01 Hz to 328 Hz, where d^H is $1,27.10^{-9}$ m/A. Above 332 Hz, strain constant increases until a local peak is reached at 440 Hz. For frequencies above 440 Hz, strain constant decreases with an increase in frequency.

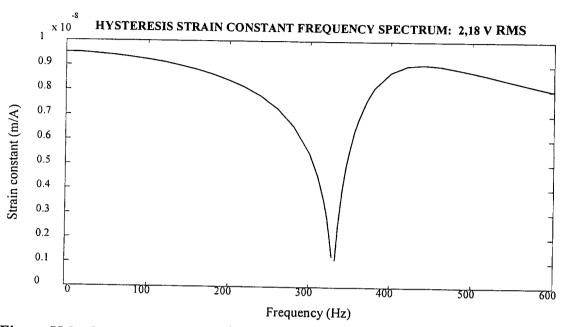


Figure K.3: Strain constant frequency spectrum, including hysteresis: 2,18 V RMS

For comparison purposes, the spectra of the strain constants for 1,63 V RMS and 1,09 V RMS inputs are shown together with that of the 2,18 V RMS input (see figure K.4). It can be seen that strain constant increases with an increase in input voltage.

The frequency spectrum of the damping coefficient, for a 2,18 V RMS input voltage, is shown in figure K.5. The damping coefficient at 0,01 Hz is 1,31.10⁶ Ns/m. A sharp notch appears between 328 Hz and 332 Hz. At the latter frequency, the damping coefficient is 4,59 Ns/m. For a frequency range of 0,01 Hz to approximately 200 Hz, damping coefficient varies hyperbolically with frequency. In this range, the mathematical relationship between frequency and damping coefficient is:

$$c(f) = \frac{13076}{f}$$
 (K.1)

where f is frequency and c is damping coefficient.



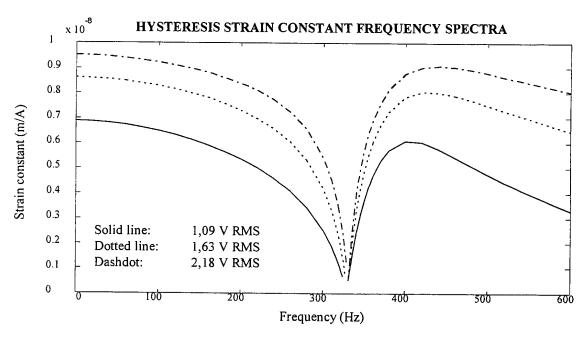


Figure K.4: Strain constant frequency spectra, including hysteresis, for 2,18 V, 1,63 V and 1,09 V RMS coil input voltages

The hyperbolic relationship between damping coefficient and frequency was discussed in section 2.3, where hysteresis models of Terfenol-D were derived (see also equation 2.3.15). On a logarithmic scale, the damping coefficient characteristic between 0,01 Hz and 200 Hz is a straight line, as shown in figure K.5.

The magnitude and phase of the transfer function for a 2,18 V RMS input, including hysteresis, are shown in figure K.6, together with the dehysterized magnitude and phase.

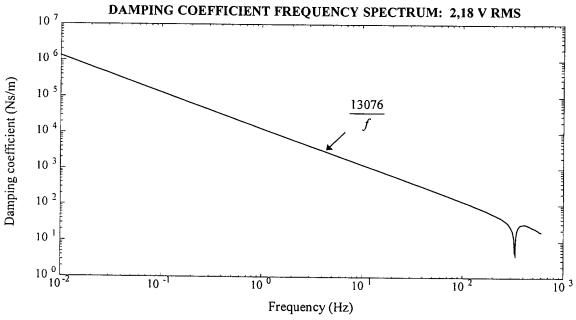


Figure K.5: Damping coefficient frequency spectrum for 2,18 V RMS input



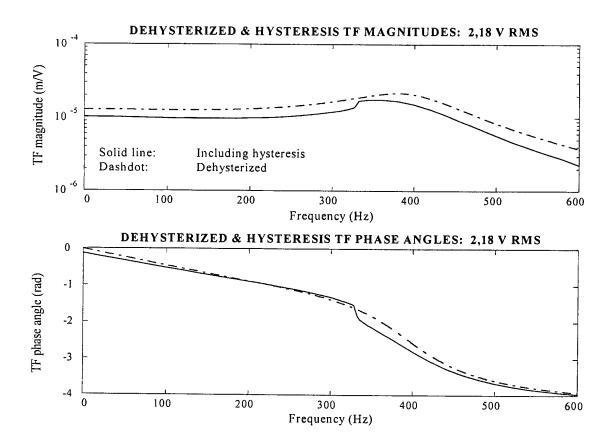


Figure K.6: Dehysterized and hysteresis transfer functions for a 2,18 V RMS input

At 0,01 Hz, the magnitude for hysteresis behaviour is 10,2 μ m/V, compared with 13,57 μ m/V for dehysterized behaviour (see table H.1). The resonance frequency and maximum magnitude for the hysteresis TF are 355 Hz and 17,7 μ m/V respectively, compared with 375 Hz and 21,88 μ m/V for the dehysterized TF. The inclusion of hysteresis effects therefore reduces magnitude at 0,01 Hz by 24,8%, resonance frequency by 5,3% and maximum magnitude by 19,1%.

Magnitude decreases slightly from 0,01 Hz to approximately 190 Hz, which can be attributed to coil inductance. At approximately 250 Hz, magnitude gradually increases with frequency. Between 328 Hz and 332 Hz, magnitude rises sharply with frequency. From 328 Hz to 370 Hz, magnitude is almost constant. Above 370 Hz, a gradual reduction in magnitude sets in. The -3 dB bandwidth is 475 Hz, which is 1,5% lower than in the dehysterized case.

The phase angle displays a peculiar characteristic, especially at low frequencies. At 0,01 Hz, phase is considerably lower that in the dehysterized case. The reason is the extremely high damping coefficient, i.e. 1,31.10⁶ Ns/m, at 0,01 Hz (see also figure K.5). This phase characteristic is typical of hysteretic systems.

For frequencies from 0,01 Hz to 328 Hz, the slope of the phase, with respect to frequency, is lower than in the dehysterized case. Between 328 Hz and 332 Hz, phase decreases sharply. At 332 Hz, the phase is considerably lower than in the dehysterized case. Above 332 Hz, a



gradual phase reduction sets in, until 600 Hz, where the phases of the dehysterized and hysteresis TF's are approximately equal.

A summary of the most important hysteresis characteristics at 0,01 Hz and resonance, for a 2,18 V RMS input, is given in table K.1.

Table K.1: Hysteresis characteristics at 0,01 Hz and resonance for a 2,18 V RMS input

Parameter	0,01 Hz	Resonance (355 Hz)
Field amplitude H_A	17618 A/m	7946 A/m
Permeability μ^{σ}	5,57.10 ⁻⁶ Tm/A	$3,63.10^{-6}$ Tm/A
Young's modulus E	23,46 GPa	23,46 GPa
Piezomagnetic cross-coupling constant d^{σ}	1,088.10 ⁻⁸ m/A	1,088.10 ⁻⁸ m/A
Strain constant d^H	9,522.10 ⁻⁹ m/A	6,338.10 ⁻⁹ m/A
Damping coefficient c	1,31.10 ⁶ Ns/m	24,31 Ns/m
TF magnitude	10,2 μm/V	$17.7 \ \mu \text{m/V}$
TF phase	-0,1263 rad	-2,205 rad
Stroke length (17,5 V p-p)	178,5 μm	309,8 μm

The transfer function spectra for 1,63 V and 1,09 V RMS inputs are shown in figures K.7 and K.8 respectively. For a 1,63 V RMS input, the magnitude at 0,01 Hz is 9,27 μ m/V, which is 32,2% lower than in the dehysterized case. The maximum magnitude of 17,9 μ m/V occurs at 345 Hz, a reduction of 18,5% in maximum magnitude and 8% in resonance frequency, in comparison with the dehysterized case. The -3 dB bandwidth is 470 Hz, which is 2,5% lower than in the dehysterized case.

For a 1,09 V RMS input, the magnitude at 0,01 Hz is 7,42 μ m/V, which is 46% lower than in the dehysterized case. The maximum magnitude of 18,1 μ m/V occurs at 340 Hz, a reduction of 17,7% in maximum magnitude and 9,3% in resonance frequency, in comparison with the dehysterized case. The -3 dB bandwidth is 445 Hz, which is 7,5% lower than in the dehysterized case.

A graphical comparison between the hysteresis transfer functions for the 2,18 V, 1,63 V and 1,09 V RMS inputs, is shown in figure K.9. The effect of input voltage on the *TF* spectra, as discussed above, is clearly visible. Furthermore, a typical nonlinear softening characteristic is displayed. This is due to the voltage dependence of the strain constant, as shown in figure K.4.



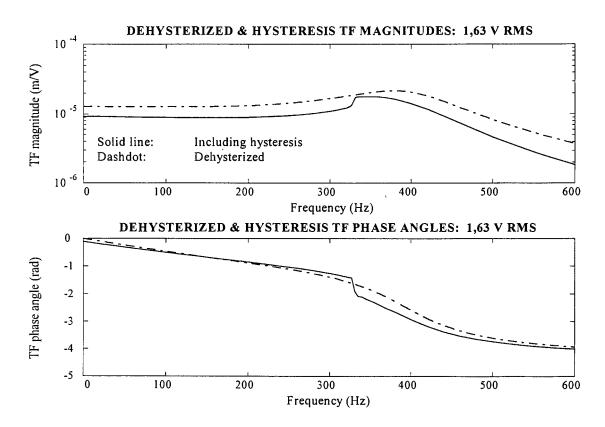


Figure K.7: Dehysterized and hysteresis transfer functions for a 1,63 V RMS input

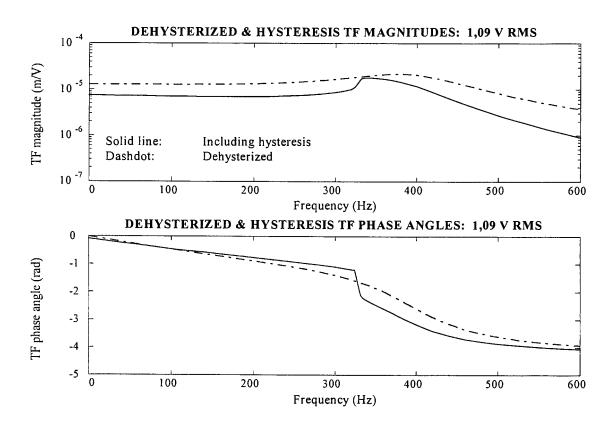


Figure K.8: Dehysterized and hysteresis transfer functions for a 1,09 V RMS input



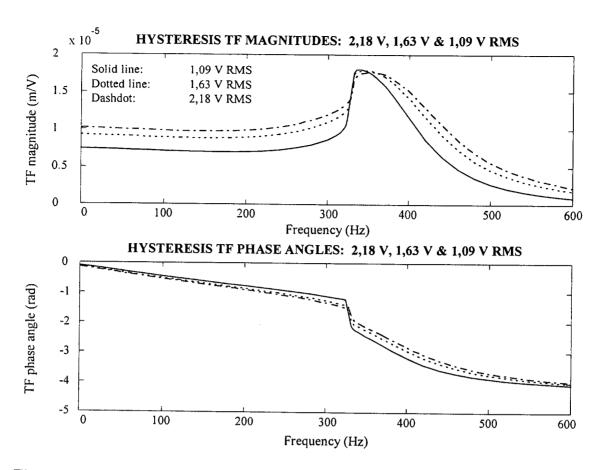


Figure K.9: Hysteresis transfer functions for 2,18 V, 1,63 V and 1,09 V RMS inputs



Appendix L Exact separation of variables method

Consider equation 2.8.2.1:

$$\underbrace{\rho_s A_s \frac{\partial^2 w}{\partial t^2}}_{mass \ term} + \underbrace{E_s I_s \frac{\partial^4 w}{\partial x^4}}_{stiffness \ term} = P$$
(L.1)

Using separation of variables, the exact solution to equation L.1 is expressed as:

$$w(x,t) = \sum_{i=1}^{\infty} \phi_i(x) q_i(t)$$
 (L.2)

where $\phi_i(x)$ is the *i*-th normal mode shape of the beam at a position x on the beam and $q_i(t)$ is the *i*-th modal amplitude at time t.

The equation of motion for natural behaviour of the beam, from equation L.1, is:

$$\underbrace{\rho_s A_s \frac{\partial^2 w}{\partial t^2}}_{mass term} + \underbrace{E_s I_s \frac{\partial^4 w}{\partial x^4}}_{stiffness term} = 0$$
(L.3)

For natural behaviour, q_i is a harmonic function of time:

$$q_i = \sin \omega_i t \tag{L.4}$$

where ω_i is the *i*-th angular natural frequency. The differential equation describing the *i*-th normal mode shape, is

$$\frac{d^4\Phi_i}{dx^4} - \beta_i^4\Phi_i = 0 \tag{L.5}$$

where β_i is the *i*-th eigenvalue, given by:

$$\beta_i = \left(\frac{\rho_s A_s}{E_s I_s} \omega_i^2\right)^{\frac{1}{4}} \tag{L.6}$$

The general solution of equation L.5 is:

$$\Phi_i(x) = A_i \cos \beta_i x + B_i \sin \beta_i x + C_i \cosh \beta_i x + D_i \sinh \beta_i x$$
 (L.7)

The values of β_i depend on the beam supports. Tabulated values of β_i for cantilever, simply-supported, clamped-clamped and clamped-free beams, are given by Harris [1988], Thomson [1993] and Tse et al [1978]. In most of these references, only the first four to six eigenvalues are given.



Appendix M Derivation of system state-space model

The forced modal equation of motion of the system (equation 2.8.5.1) is:

$$M^*\ddot{q} + C^*\dot{q} + K^*q = Q$$
 (M.1)

where M, C and K are normal mode mass, damping and stiffness matrices respectively and Q is the modal excitation force vector, given by:

$$Q = \int_{x=0}^{x=l} \Phi^{T}(x) F(x,t) dx$$
 (M.2)

F is the excitation force at any point x, at time t:

$$F = F(x,t) \tag{M.3}$$

In a more concise matrix form, equation M.2 can be written as:

$$Q = \Phi^T F \tag{M.4}$$

The coordinates x_{a1} and x_{a2} of the actuator attachment points are:

$$x_{a1} = 0 ag{M.5a}$$

$$x_{a2} = l ag{M.5b}$$

Substitution of equations M.5 into equation M.2 and writing the resulting equation in the form of equation M.4, gives:

$$Q = \left[\Phi^{T}(0) \quad \Phi^{T}(l) \right] \begin{cases} F_{a1} \\ F_{a2} \end{cases}$$
 (M.6)

 F_{a1} and F_{a2} are the actuator excitation forces, which (from section 2.4) can be expressed as:

$$F_{a1} = \frac{A_T E N d^H}{G l_T} I_1 \tag{M.7a}$$

$$F_{a2} = \frac{A_T E N d^H}{G l_T} I_2 \tag{M.7b}$$

where I_1 and I_2 are the two actuator coil currents.

The modal equation of motion is obtained by substitution of equations M.7a and M.7b into



equation M.6 and by subsequent substitution of the resulting equation into equation M.1:

$$\ddot{q} + 2Z\Omega\dot{q} + \Omega^2 q = \frac{A_T ENd^H}{Gl_T} M^{*-1} \left[\Phi^T(0) \quad \Phi^T(l) \right] \begin{Bmatrix} I_1 \\ I_2 \end{Bmatrix}$$
 (M.8)

where Z is the diagonal modal damping matrix, whose elements are given as follows in terms of the damping coefficient C_{ii}^{\bullet} , mass M_{ii}^{\bullet} and natural frequency Ω_{ii} of the *i*-th normal mode:

$$Z_{ii} = \frac{C_{ii}^{\bullet}}{2M_{ii}^{\bullet}\Omega_{ii}} \tag{M.9}$$

The coil current equations are:

$$\dot{I}_{1} = -\frac{A_{T}ENd^{\sigma}}{Gl_{T}L_{0}}\dot{w}(0,t) - \frac{R_{c}}{L_{0}}I_{1} + \frac{1}{L_{0}}V_{1}$$
(M.10a)

$$\dot{I}_{2} = -\frac{A_{T}ENd^{\sigma}}{Gl_{T}L_{0}}\dot{w}(l,t) - \frac{R_{c}}{L_{0}}I_{2} + \frac{1}{L_{0}}V_{2}$$
(M.10b)

where V_1 and V_2 are the coil input voltages and $\dot{w}(0,t)$ and $\dot{w}(l,t)$ are the vertical translational speeds at the actuator attachment points, given by:

$$\dot{w}(0,t) = \Phi(0)\dot{q}(t) \tag{M.11a}$$

$$\dot{w}(l,t) = \Phi(l)\dot{q}(t) \tag{M.11b}$$

Substitution of equations M.11 into equations M.10 gives:

$$\begin{cases}
\dot{I}_{1} \\
\dot{I}_{2}
\end{cases} = -\frac{A_{T}ENd^{\sigma}}{Gl_{T}L_{0}} \begin{bmatrix} \Phi(0) \\ \Phi(l) \end{bmatrix} \dot{q} - \frac{R_{c}}{L_{0}} \begin{bmatrix} 1 & 0 \\ 0 & 1 \end{bmatrix} \begin{bmatrix} I_{1} \\ I_{2} \end{bmatrix} + \frac{1}{L_{0}} \begin{bmatrix} 1 & 0 \\ 0 & 1 \end{bmatrix} \begin{bmatrix} V_{1} \\ V_{2} \end{bmatrix}$$
(M.12)

Combining equations M.8 and M.12 and writing the resulting system of equations in state-space form, gives:



The outputs can be chosen to be the translational and rotational displacements of the instrument. The rotational displacement, which is equal to the LOS angle θ , is of prime importance. The translational displacement is not required, since it does not determine LOS accuracy. Two other outputs, which are of importance, are the vertical displacements w_1 and w_2 at the instrument attachment points x_1 and x_2 on the support structure. The difference between these two outputs, divided by the instrument length, gives the LOS angle θ .

Displacement w_1 is given by:

$$w_1 = w(x_1, t) = \Phi(x_1)q(t)$$
 (M.14)

Displacement w_2 is given by:

$$w_2 = w(x_2, t) = \Phi(x_2)q(t)$$
 (M.15)

The angular displacement θ of the optical instrument is:

$$\theta(t) = \frac{1}{l} (w_2 - w_1) \tag{M.16}$$

Substitution of equations M.14 and M.15 into equation M.16 gives:

$$\theta(t) = \frac{1}{l} (\Phi(x_2) - \Phi(x_1)) q(t)$$
(M.17)

Equations M.14, M.15 and M.17 are combined as follows in one output equation:

$$\begin{cases}
w_1 \\
w_2 \\
\theta
\end{cases} = \begin{bmatrix}
\{\Phi(x_1)\} & \{0\} & \{0 & 0\} \\
\{\Phi(x_2)\} & \{0\} & \{0 & 0\} \\
\end{bmatrix} \begin{cases}
\{q\} \\
\{\dot{q}\} \\
\{\dot{q}\} \\
\end{bmatrix} \begin{cases}
\mathbf{M.18}$$



Appendix N Derivation of the complex Laplace-domain transfer function of a hydraulically-gained magnetostrictive actuator

A hydraulically-gained Terfenol-D actuator is schematically shown in figure N.1. Its equivalent mechanical model is shown in figure N.2.

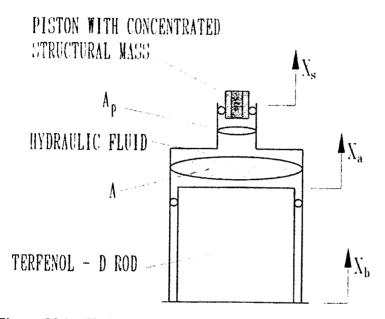


Figure N.1: Hydraulically-gained magnetostrictive actuator

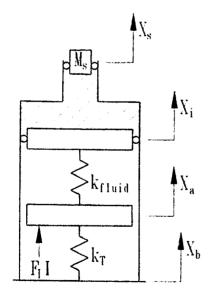


Figure N2: Equivalent mechanical model of the hydraulically-gained actuator

The derivation that follows, will be done for a static base, i.e. $x_b = 0$.



Bulk modulus and equivalent linear liquid stiffness

The bulk modulus K_B of a fluid is defined as the ratio of the change in pressure dP, to the change in volume dV per unit volume V:

$$K_B = \frac{dP}{\left(\frac{dV}{V}\right)} \tag{N.1}$$

which gives:

$$dP = \frac{K_B}{V} dV \tag{N.2}$$

Multiplication of equation N.2 with the Terfenol-D rod cross-sectional area A gives:

$$AdP = \frac{AK_B}{V}dV \tag{N.3}$$

The infinitesimal change in the force transmitted by the fluid dF_{fluid} is given by:

$$dF_{fluid} = AdP (N.4)$$

The infinitesimal change in fluid volume is given by:

$$dV = Ad(x_a - x_i) (N.5)$$

Substitution of equations N.4 and N.5 into equation N.3 and rewriting the resulting equation gives:

$$\frac{dF_{fluid}}{d(x_a - x_i)} = \frac{A^2 K_B}{V}$$
 (N.6)

It can be seen from equation N.6 that the rate of change of the force transmitted by the fluid, with respect to the deflection of the rod, is directly proportional to the rod area squared, directly proportional to the fluid bulk modulus and inversely proportional to the fluid volume. A and K_B are constant, while V is not. Equation N.6 is therefore nonlinear. In order to obtain a linear force-deflection characteristic, equation N.6 must be linearized. This is done below.

During excitation of the actuator, the fluid volume will contain a DC, or reference component V_{ref} and a fluctuating component ΔV :

$$V = V_{ref} + \Delta V \tag{N.7}$$



Substitution of equation N.7 into equation N.6 gives:

$$\frac{dF_{fluid}}{d(x_a - x_i)} = \frac{A^2 K_B}{V_{ref} + \Delta V}$$
 (N.8)

Dividing both the numerator and denominator of the right hand side of equation N.8 by V_{ref} gives:

$$\frac{dF_{fluid}}{d(x_a - x_i)} = \frac{\frac{A^2 K_B}{V_{ref}}}{1 + \frac{\Delta V}{V_{ref}}}$$
(N.9)

Most hydraulic fluids have high bulk moduli. $\Delta V/V_{ref}$ is therefore relatively small for most hydraulic fluids, so that the numerator on the right hand side of equation N.9 can be approximated by:

$$1 + \frac{\Delta V}{V_{ref}} \approx 1 \tag{N.10}$$

Substitution of equation N.10 into equation N.9 gives:

$$\frac{dF_{fluid}}{d(x_a - x_i)} = \frac{A^2 K_B}{V_{ref}}$$
 (N.11)

It can be seen from equation N.11 that the equivalent linear liquid stiffness k_{fluid} is given by:

$$k_{fhid} = \frac{A^2 K_B}{V_{ref}} \tag{N.12}$$

Deflection of the Terfenol-D rod

Consider figure N.2. The deflection x_a of the rod is determined by the force balance equation:

$$k_{fluid}(x_i - x_a) - k_T x_a + F_I I = 0$$
 (N.13)

where I is the coil current, F_I is the force per unit current and k_T is the rod stiffness. F_I and k_T are respectively given by:

$$F_{I} = \frac{AENd^{H}}{l_{T}}$$
 (N.14a)

$$k_T = \frac{AE}{l_T}$$
 (N.14b)



Rewriting equation N.13 gives:

$$x_a = \frac{k_{fluid}}{k_{fluid} + k_T} x_i + \frac{F_I}{k_{fluid} + k_T} I$$
 (N.15)

The displacement gain is given as follows in terms of x_s and x_i :

$$G = \frac{x_s}{x_i} \tag{N.16}$$

Substitution of equation N.16 into equation N.15 gives the deflection of the rod as follows in terms of x_x and I:

$$x_a = \frac{k_{fluid}}{k_{fluid} + k_T} \frac{x_s}{G} + \frac{F_I}{k_{fluid} + k_T} I$$
 (N.17)

Equation of motion

The equation of motion of the piston is:

$$\ddot{x}_s = \frac{PA_p}{m_s} + \ddot{x}_i \tag{N.18}$$

where P is the fluid pressure and A_p is the piston cross-sectional area.

The displacement gain factor, in terms of the piston and rod areas, is given by:

$$G = \frac{A}{A_p} \tag{N.19}$$

Substitution of equation N.19 into equation N.18 gives:

$$\ddot{x}_s = \frac{PA}{m_s G} + \ddot{x}_i \tag{N.20}$$

Substitution of equation N.16 into equation N.20 and simplification of the resulting equation gives:

$$\ddot{x}_s = \frac{PA}{m_s(G-1)} \tag{N.21}$$

The force exerted by the rod on the fluid, is given by:

$$PA = k_{fluid} \left(x_a - x_i \right) \tag{N.22}$$



Substitution of equation N.22 into equation N.21 gives:

$$\ddot{x}_s = \frac{k_{fluid}}{m_s(G-1)} (x_a - x_i)$$
 (N.23)

Substitution of equation N.16 into equation N.23 and simplification of the resulting equation gives:

$$\ddot{x}_s + \frac{k_{fluid}}{m_s G(G-1)} x_s = \frac{k_{fluid}}{m_s (G-1)} x_a$$
 (N.24)

Substitution of equation N.17 into equation N.24 gives:

$$\ddot{x}_{s} + \frac{k_{fluid}}{m_{s}G(G-1)}x_{s} = \frac{k_{fluid}}{m_{s}(G-1)} \left(\frac{k_{fluid}}{k_{fluid} + k_{T}} \frac{x_{s}}{G} + \frac{F_{I}}{k_{fluid} + k_{T}} I \right)$$
 (N.25)

Collecting terms containing x_s on the left hand side gives:

$$\ddot{x}_{s} + \frac{k_{fluid}}{m_{s}G(G-1)} \left(1 - \frac{k_{fluid}}{k_{fluid} + k_{T}} \right) x_{s} = \frac{F_{I}}{m_{s}(G-1)} \left(\frac{k_{fluid}}{k_{fluid} + k_{T}} \right) I$$
 (N.26)

Simplification of equation N.26 gives the equation of motion as follows in terms of x_s and I:

$$\ddot{x}_{s} + \frac{k_{T}}{m_{s}G(G-1)} \left(\frac{k_{fluid}}{k_{fluid} + k_{T}} \right) x_{s} = \frac{F_{I}}{m_{s}(G-1)} \left(\frac{k_{fluid}}{k_{fluid} + k_{T}} \right) I$$
 (N.27)

Coil current equation

The coil current equation of an ungained magnetostrictive actuator is given by:

$$\dot{I} = -\frac{AENd^{\sigma}}{l_{T}L_{0}}\dot{x}_{a} - \frac{R_{c}}{L_{0}}I + \frac{1}{L_{0}}V$$
(N.28)

Substitution of equation N.17 into equation N.28 gives:

$$\dot{I} = -\frac{AENd^{\sigma}}{l_{T}L_{0}} \left(\frac{k_{fluid}}{k_{fluid} + k_{T}} \frac{\dot{x}_{s}}{G} + \frac{F_{I}}{k_{fluid} + k_{T}} \dot{I} \right) - \frac{R_{c}}{L_{0}} I + \frac{1}{L_{0}} V$$
(N.29)



Collecting terms containing \dot{I} on the left hand side gives:

$$\left[1 + \left(\frac{k_T}{k_{fluid} + k_T}\right) \frac{F_I N d^{\sigma}}{L_0}\right] \dot{I} = -\left(\frac{k_{fluid}}{k_{fluid} + k_T}\right) \frac{k_T N d^{\sigma}}{GL_0} \dot{x}_s - \frac{R_c}{L_0} I + \frac{1}{L_0} V \tag{N.30}$$

The term $k_T/(k_{fluid} + k_T)$ is expressed as:

$$\frac{k_T}{k_{fluid} + k_T} = 1 - \frac{k_{fluid}}{k_{fluid} + k_T}$$
 (N.31)

Substitution of equation N.31 into equation N.30 gives the coil current equation as follows in terms of x_s and I:

$$\[1 + \left(1 - \frac{k_{fluid}}{k_{fluid} + k_{T}}\right) \frac{F_{I} N d^{\sigma}}{L_{0}} \right] \dot{I} = -\left(\frac{k_{fluid}}{k_{fluid} + k_{T}}\right) \frac{k_{T} N d^{\sigma}}{GL_{0}} \dot{x}_{s} - \frac{R_{c}}{L_{0}} I + \frac{1}{L_{0}} V \quad (N.32)$$

State-space equations

The equation of motion (N.27) and coil current equation (N.32) are simplified by defining the following two constant parameters α and β :

$$\alpha = \frac{k_{fluid}}{k_{fluid} + k_{T}} \tag{N.33a}$$

$$\beta = 1 + \left(1 - \alpha\right) \frac{F_I N d^{\sigma}}{L_0}$$
 (N.33b)

Substitution of equation N.33a into equation N.27 gives the equation of motion as:

$$\ddot{x}_s + \alpha \frac{k_T}{m_s G(G-1)} x_s = \alpha \frac{F_I}{m_s (G-1)} I$$
 (N.34)

Substitution of equations N.33 into equation N.32 and dividing both sides of the resulting equation by β gives the coil current equation as:

$$\dot{I} = -\frac{\alpha}{\beta} \frac{k_T N d^{\sigma}}{GL_0} \dot{x}_s - \frac{R_c}{\beta L_0} I + \frac{1}{\beta L_0} V$$
 (N.35)



The equation of motion (N.34) and coil current equation (N.35) are combined into a single set of state-space equations, in the following familiar form:

$$\dot{x} = Ax + Bu \tag{N.36a}$$

$$y = Cx + Du (N.36b)$$

where

$$x = \left\{ x_s, \quad \dot{x}_s, \quad I \right\}^T \tag{N.37}$$

is the state vector,

$$u = V \tag{N.38}$$

is the coil voltage input,

$$y = x_s (N.39)$$

is the actuator displacement output,

$$A = \begin{bmatrix} 0 & 1 & 0 \\ -\alpha \frac{k_T}{m_s G(G-1)} & 0 & \alpha \frac{F_I}{m_s (G-1)} \\ 0 & -\frac{\alpha}{\beta} \frac{k_T N d^{\sigma}}{G L_0} & -\frac{1}{\beta} \frac{R_c}{L_0} \end{bmatrix}$$
 (N.40)

is the coefficient matrix,

$$B = \begin{bmatrix} 0 \\ 0 \\ \frac{1}{\beta L_0} \end{bmatrix}^T$$
 (N.41)

is the driving matrix,

$$C = \begin{bmatrix} 1 & 0 & 0 \end{bmatrix}$$
 (N.42)

is the output matrix and

$$D=0 (N.43)$$

is the transmission matrix.



Coil voltage to displacement TF

The coil voltage input to displacement output TF is obtained from:

$$\frac{X_s(s)}{U(s)} = C[sI - A]^{-1}B + D$$
 (N.44)

Substitution of equations N.40 to N.43 into equation N.44, and using the relationships for k_r and F_t in equations N.14, gives the TF as:

$$\frac{X_{s}(s)}{V(s)} = \frac{\frac{\alpha}{\beta} \frac{AENd^{H}}{m_{s}(G-1)l_{T}L_{0}}}{s^{3} + \frac{1}{\beta} \frac{R_{c}}{L_{0}} s^{2} + \left[\alpha \frac{AE}{m_{s}G(G-1)l_{T}} + \frac{\alpha^{2}}{\beta} \left(\frac{AEN}{l_{T}}\right)^{2} \frac{d^{\sigma}d^{H}}{m_{s}G(G-1)L_{0}}\right] s + \frac{\alpha}{\beta} \frac{AE}{m_{s}G(G-1)l_{T}} \frac{R_{c}}{L_{0}}$$
(N.45)



Appendix P Displacement gain factor for elliptical gain mechanism

Consider figure 3.4.1.4. The undeformed lengths of the major and minor axes of the ellipse are given by b and a, respectively. The displacement gain factor G is the ratio of the change in length Δa of the minor axis, to the change in length Δb of the major axis:

$$G = \frac{\Delta a}{\Delta b} \tag{P.1}$$

It is however more convenient to obtain an equation for the gain factor in terms of the change in length of the major axis only, since this will make it possible to obtain the output displacement of the gain mechanism for a given input displacement. The derivation of this equation is done as follows. Exact and approximate equations for the perimeter of an ellipse are given and compared. It will be shown that the approximate equation is reasonably accurate for a/b ratios ranging from 0,1 to 1. The approximate equation is then used to derive an expression for Δa in terms of a, b and Δb . The latter equation is substituted into equation P.1 to give the gain factor.

The exact equation for the perimeter of an ellipse, from Spiegel [1968], is:

$$P_{e} = 4b \int_{0}^{\pi/2} \sqrt{1 - k^{2} \sin^{2} \theta} \ d\theta$$
 (P.2)

where P_e is the perimeter, θ is a dummy variable, which ranges from zero to the included angle between the major and minor axes, i.e. $\pi/2$, and k is a dimensionless constant, which depends on the lengths of the major and minor axes:

$$k = \frac{\sqrt{b^2 - a^2}}{b} \tag{P.3}$$

Division of equation P.2 by b, gives the ratio of the exact perimeter, to the length of the major axis:

$$\frac{P_e}{b} = 4 \int_{0}^{\pi/2} \sqrt{1 - k^2 \sin^2 \theta} \ d\theta$$
 (P.4)

The integral in equation P.4 is an incomplete elliptical integral of the second kind and does not have a closed-form solution. The solution can be obtained numerically.

As an alternative, the following approximate equation can be used [Spiegel, 1968]:

$$P_e \approx 2\pi \sqrt{\frac{1}{2}(a^2 + b^2)}$$
 (P.5)



From equation P.5, the ratio of the approximate perimeter to the length of the major axis, is obtained as:

$$\frac{P_e}{b} \approx 2\pi \sqrt{\frac{1}{2} \left[\left(\frac{a}{b} \right)^2 + 1 \right]}$$
 (P.6)

The approximate and exact P_e/b ratios, for a/b ratios from 0,1 to 1, together with the percentage error of the approximate P_e/b ratio, are given in table P.1.

Table P.1: Approximate and exact P_e/b ratios for different a/b ratios

a/b	Approximate <i>P_e/b</i> (Equation P.6)	Exact <i>P_e/b</i> (Equation P.4)	% Error
0,10	4,4650	4,0574	10,05
0,20	4,5309	4,2052	7,74
0,30	4,6385	4,3910	5,64
0,40	4,7851	4,5993	4,04
0,50	4,9673	4,8444	2,54
0,60	5,1812	5,1065	1,46
0,70	5,4232	5,3817	0,77
0,80	5,6897	5,6725	0,30
0,90	5,9773	5,9732	0,07
1,00 (circle)	6,2832	6,2832	0,00

It can be seen from table P.1 that the approximate P_c/b ratio, as obtained from equation P.6, is acceptably accurate for the given range of a/b ratios.

Equation P.6 is subsequently used to derive an equation for G in terms of a, b and Δb . It is assumed that the perimeter P_e stays constant, irrepective of Δb .

Replacing the approximate equation P.6 with an equation, squaring both sides, and rewriting gives:

$$\frac{P_e^2}{2\pi^2} = a^2 + b^2 \tag{P.7}$$

For a constant perimeter, b can be increased to $b+\Delta b$, and a can be simultaneously decreased to $a-\Delta a$, without affecting the left hand side of equation P.7. Replacing a with $a-\Delta a$ and b with $b+\Delta b$ in equation P.7, gives:

$$\frac{P_e^2}{2\pi^2} = (a - \Delta a)^2 + (b + \Delta b)^2$$
 (P.8)



Equations P.7 and P.8 are equal, therefore:

$$(a - \Delta a)^2 + (b + \Delta b)^2 = a^2 + b^2$$
 (P.9)

Simplification of equation P.9 gives the following quadratic equation in Δa :

$$(\Delta a)^2 - 2a(\Delta a) + \left[2b\Delta b + (\Delta b)^2\right] = 0$$
 (P.10)

The roots of equation P.10 are:

$$\Delta a = a \pm \sqrt{a^2 - 2b\Delta b + (\Delta b)^2}$$
 (P.11)

Rejection of the largest root in equation P.11 gives Δa as follows in terms of a, b and Δb :

$$\Delta a = a - \sqrt{a^2 - 2b\Delta b + (\Delta b)^2}$$
 (P.12)

Substitution of equation P.12 into equation P.1, gives the following expression for G in terms of a, b and Δb :

$$G = \frac{a - \sqrt{a^2 - 2b\Delta b + (\Delta b)^2}}{\Delta b}$$
 (P.13)



Appendix Q

Exact and approximate gain factors for octagonal flexural gain mechanism

Consider figure 3.4.1.5. The displacement gain factor is the ratio of the change in height of the octagonal gain mechanism to the elongation of the rod:

$$G = \frac{x_s}{x_a} \tag{Q.1}$$

where x_a is the product of the strain ε in the rod and the rod length l_T :

$$x_a = \varepsilon l_T \tag{Q.2}$$

and x_s is given by:

$$x_s = 2(h_0 - h) \tag{Q.3}$$

The initial height h_0 of the slanted beam is given by:

$$h_0 = r \sin \theta_0 \tag{Q.4}$$

and the final height h is given by:

$$h = \sqrt{r^2 - \left(r\cos\theta_0 + \frac{x_a}{2}\right)^2}$$
 (Q.5)

The final height is obtained in terms of the rod strain and length, by substitution of equation Q.2 into equation Q.5:

$$h = \sqrt{r^2 - \left(r\cos\theta_0 + \frac{\varepsilon l_T}{2}\right)^2}$$
 (Q.6)

Substitution of equations Q.4 and Q.6 into equation Q.3 gives x_s as follows:

$$x_s = 2\left(r\sin\theta_0 - \sqrt{r^2 - \left(r\cos\theta_0 + \frac{\epsilon l_T}{2}\right)^2}\right)$$
 (Q.7)

The exact gain factor is obtained by substitution of equations Q.7 and Q.2 into equation Q.1:

$$G = 2 \frac{r \sin \theta_0 - \sqrt{r^2 - \left(r \cos \theta_0 + \frac{\varepsilon l_T}{2}\right)^2}}{\varepsilon l_T}$$
 (Q.8)



Approximate gain factor

An approximate equation for the displacement gain, in terms of θ_0 , is subsequently derived.

From equation Q.8:

$$\frac{G\varepsilon l_T}{2} - r\sin\theta_0 = -\sqrt{r^2 - \left(r\cos\theta_0 + \frac{\varepsilon l_T}{2}\right)^2}$$
 (Q.9)

Squaring both sides of equation Q.9 gives:

$$\left(\frac{G\varepsilon l_T}{2} - r\sin\theta_0\right)^2 = r^2 - \left(r\cos\theta_0 + \frac{\varepsilon l_T}{2}\right)^2 \tag{Q.10}$$

Expanding terms on both sides of equation Q.10 gives:

$$\left(\frac{G\varepsilon l_T}{2}\right)^2 - G\varepsilon l_T r \sin\theta_0 + r^2 \sin^2\theta_0 = r^2 - r^2 \cos^2\theta_0 - \varepsilon l_T r \cos\theta_0 - \left(\frac{\varepsilon l_T}{2}\right)^2$$
 (Q.11)

Neglecting 2^{nd} order terms and noting that $\sin^2 \theta_0 + \cos^2 \theta_0 \equiv 1$, equation Q.11 reduces to:

$$-G\varepsilon l_T r \sin\theta_0 \approx -\varepsilon l_T r \cos\theta_0 \tag{Q.12}$$

From equation Q.12, the following approximate equation for G, in terms of θ_0 , results:

$$G \approx \cot \theta_0 \tag{Q.13}$$

A graphical comparison between the gain factors calculated with equations Q.8 and Q.13, for initial angles ranging from 5° to 30°, an active rod length of 35 mm, strain of 1000 $\mu\epsilon$ and slanted beam radius of 19 mm, is shown in figure Q.1.

The error of the approximate gain factor, relative to the exact gain factor, is shown in figure Q.2. It can be seen that the approximate gain factor is sufficiently accurate for initial angles ranging from 5° to 30°, i.e. for gains ranging from 1,7 to 11,4. The error is less than 5% for angles larger than 5,8° and less than 1° for angles larger than 12,8°.

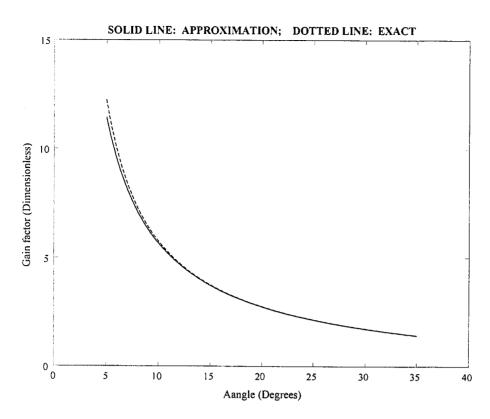


Figure Q.1: Approximate and exact gain factors versus initial angle of slanted beam

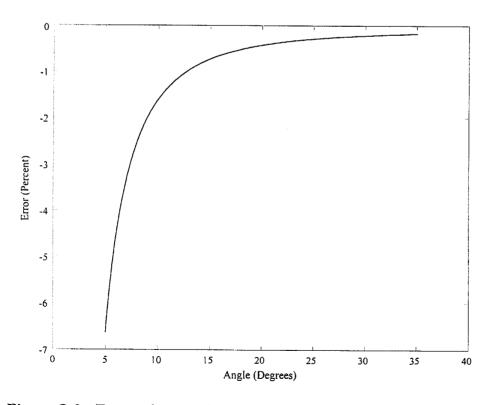


Figure Q.2: Error of approximate gain factor versus initial angle of slanted beam



Appendix R Volume of a permanent magnet

Consider figure 3.4.2.1. The fluxes in the magnet, Terfenol-D rod and air gap between the end caps and magnet, are respectively given by:

$$\phi_{m} = B_{m}A_{m} \tag{R.1a}$$

$$\phi_T = B_T A_T \tag{R.1b}$$

$$\phi_{g} = B_{g} A_{g} \tag{R.1c}$$

where ϕ_m , ϕ_T and ϕ_g are the fluxes in the magnet, rod and air gap respectively. B_m , B_T and B_g are the flux densities in the magnet, rod and air gap respectively. A_m , A_T and A_g are the cross-sectional areas of the magnet, rod and air gap respectively.

The flux in the magnet is equal to the flux in the rod and the flux in the air gap:

$$\phi_m = \phi_T = \phi_g \tag{R.2}$$

Substitution of equations R.1a to R.1c into equation R.2 gives:

$$B_g = \frac{B_T A_T}{A_g} \tag{R.3a}$$

$$B_m = \frac{B_T A_T}{A_m} \tag{R.3b}$$

The total magnetomotive force in the loop is zero:

$$H_m l_m + H_T l_T + H_g l_g = 0 (R.4)$$

where H_m is the demagnetizing force in the magnet and H_T and H_g are the field in the rod and the field in the gap respectively. l_m , l_T and l_g are the magnet, rod and gap lengths respectively.

In equation R.4, it is assumed that the permeability of the end caps is infinite. The assumption is justifiable, for two reasons. In the first place, the end caps are made of steel, which has a relatively high permeability. Secondly, the end caps are relatively thin in comparison with the rod length.

Equation R.4 can be written as:

$$H_m l_m = -H_T l_T - H_g l_g \tag{R.5}$$



The field in the gap is given by:

$$H_g = \frac{B_g}{\mu_0} \tag{R.6}$$

where $\mu_0 = 4\pi . 10^{-7} \, Tm / A$ is the permeability of free space.

Substitution of equation R.6 into equation R.3a gives:

$$H_g = \frac{B_T A_T}{\mu_0 A_g} \tag{R.7}$$

Subsequent substitution of equation R.7 into equation R.5 gives:

$$H_{m}l_{m} = -\frac{B_{T}A_{T}l_{g}}{\mu_{0}A_{g}} - H_{T}l_{T}$$
 (R.8)

 H_m is obtained as follows from equation R.8:

$$H_{m} = -\frac{1}{l_{m}} \left(\frac{B_{T} A_{T} l_{g}}{\mu_{0} A_{g}} - H_{T} l_{T} \right)$$
 (R.9)

Multiplying equation R.3b with equation R.9 gives:

$$B_{m}H_{m} = -\frac{B_{T}A_{T}}{A_{m}l_{m}} \left(\frac{B_{T}A_{T}l_{g}}{\mu_{0}A_{g}} - H_{T}l_{T} \right)$$
 (R.10)

The magnet volume V_m is given by:

$$V_m = A_m l_m \tag{R.11}$$

Substitution of equation R.11 into equation R.10 gives:

$$B_{m}H_{m} = -\frac{B_{T}A_{T}}{V_{m}} \left(\frac{B_{T}A_{T}l_{g}}{\mu_{0}A_{g}} - H_{T}l_{T} \right)$$
 (R.12)

Rewriting equation R.12 gives:

$$V_{m} = \frac{B_{T}A_{T}}{B_{m}H_{m}} \left(H_{T}l_{T} - \frac{B_{T}A_{T}l_{g}}{\mu_{0}A_{g}} \right)$$
 (R.13)



The flux density in the rod is a function of the field in the rod:

$$B_T = B_T(H_T) \tag{R.14}$$

The required field in the rod is equal to the bias field:

$$H_T = H_b \tag{R.15}$$

Substitution of equation R.15 into equation R.14 gives the flux density in the rod in terms of the bias field:

$$B_T = B_T(H_b) \tag{R.16}$$

Substitution of equations R.15 and R.16 into equation R.13 gives:

$$V_{m} = \frac{B_{T}(H_{b})A_{T}}{B_{m}H_{m}} \left(H_{b}l_{T} - \frac{B_{T}(H_{b})A_{T}l_{g}}{\mu_{0}A_{g}}\right)$$
 (R.17)



Appendix S Maximum coil power for electromagnetic field biasing

The coil current, in terms of the magnetic field, is given by (see equation 3.4.2.7):

$$I = \frac{l_T}{N}H\tag{S.1}$$

The coil voltage, in terms of the field, is given by (see equation 3.4.2.10):

$$V = \frac{l_T}{N} \sqrt{R_c^2 + \omega^2 L_0^2} H$$
 (S.2)

The coil power is the product of the coil voltage and coil current:

$$P = VI (S.3)$$

Substitution of equations S.1 and S.2 into equation S.3 gives the following equation for coil power in terms of magnetic field:

$$P = \left(\frac{l_T}{N}\right)^2 \sqrt{R_c^2 + \omega^2 L_0^2} H^2$$
 (S.4)

Parameters which are of particular importance in the analysis of the electromagnetic field biasing, are the maxima of the coil current, voltage and power. For excitation frequencies that are considerably lower than the coil $R_{\rm o}/L_0$ ratio, the phase angle of the coil can be ignored. It can therefore be assumed that the maximum voltage and current coincide with the maximum field. By sustitution of $H = H_{\rm max}$ into equations S.1, S.2 and S.4, the maximum current, voltage and power are given by:

$$I_{\text{max}} = \frac{l_T}{N} H_{\text{max}}$$
 (S.5a)

$$V_{\text{max}} = \frac{l_T}{N} \sqrt{R_c^2 + \omega^2 L_0^2} H_{\text{max}}$$
 (S.5b)

$$P_{\text{max}} = \left(\frac{l_T}{N}\right)^2 \sqrt{R_c^2 + \omega^2 L_0^2} H_{\text{max}}^2$$
 (S.5c)

In equations S.5, the maximum field is the sum of the bias field H_b and the amplitude H_A of the AC component of the field:

$$H_{\text{max}} = H_b + H_A \tag{S.6}$$



By substitution of equation S.6 into equations S.5, the maximum coil current, voltage and power are expressed as follows:

$$I_{\text{max}} = \frac{l_T}{N} (H_b + H_A)$$
 (S.7a)

$$V_{\text{max}} = \frac{l_T}{N} \sqrt{R_c^2 + \omega^2 L_0^2} (H_b + H_A)$$
 (S.7b)

$$P_{\text{max}} = \left(\frac{l_T}{N}\right)^2 \sqrt{R_c^2 + \omega^2 L_0^2} (H_b + H_A)^2$$
 (S.7c)



Appendix T

Fundamental natural frequency of a simply-supported beam with a concentrated mass attached at the centre and which is subjected to a compressive axial force

The fundamental natural frequency of a beam with a concentrated mass at the centre, and which is subjected to an axial force acting at the beam ends, is analyzed in the following steps. The fundamental natural frequency of a simply-supported, uniform beam is first analyzed in the absence of any concentrated mass or axial force. The effect of the concentrated mass is then added. For the purpose of simplifying the analysis, the concentrated mass is expressed in terms of an equivalent density, which is added to the beam material density to give the total distributed beam density.

The next step will be to write the fundamental natural frequency of a uniform beam, which is subjected to an axial force. The frequency is then adjusted by replacing the beam material density with the total distributed density, to include the effect of the concentrated mass.

The fundamental angular natural frequency ω_l of a simply-supported uniform beam, in the absence of any concentrated mass, can be obtained from the following equation:

$$\omega_1^2 = \left(\frac{\pi}{l_T}\right)^4 \frac{EI}{\rho A_T} \tag{T.1}$$

where E is Young's modulus of the rods, I is the second moment of area of the rod cross-section, ρ is the density of the rod material, A_T is the rod cross-sectional area and I_T is the rod length.

The relationship between the natural frequency f_1 and the angular natural frequency ω_1 is:

$$f_1 = \frac{1}{2\pi}\omega_1 \tag{T.2}$$

By taking the square root of the terms on both sides of equation T.1, and substitution of the resulting equation into equation T.2, the fundamental natural frequency is obtained as:

$$f_1 = \frac{1}{2\pi} \left(\frac{\pi}{l_T}\right)^2 \sqrt{\frac{EI}{\rho A_T}}$$
 (T.3)

An equation of the same form as equation T.3, for the fundamental natural frequency of a beam with a concentrated mass attached at the centre, is required. This will facilitate inclusion of the effect of the axial force. To this end, the concentrated mass is first expressed in terms of an equivalent density. The equivalent density is then added to the beam material density to give the distributed density of the beam.



The equivalent density ρ_{eq} is given by the following equation:

$$\rho_{eq} = \frac{2m_c}{A_T l_T} \tag{T.4}$$

where m_c is the concentrated mass.

The distributed density ρ_{distr} is the sum of the beam material density and the equivalent density:

$$\rho_{distr} = \rho + \rho_{ea} \tag{T.5}$$

Substitution of equation T.4 into equation T.5 gives the distributed density as follows in terms of the beam material density and concentrated mass:

$$\rho_{distr} = \rho \left(1 + \frac{2m_c}{\rho A_T l_T} \right) \tag{T.6}$$

The fundamental natural frequency of the beam with a mass attached at the centre, is expressed as follows in the form of equation T.3:

$$f_1 = \frac{1}{2\pi} \left(\frac{\pi}{l_T}\right)^2 \sqrt{\frac{EI}{\rho_{div_T} A_T}}$$
 (T.7)

where ρ_{distr} is given by equation T.6.

Substitution of equation T.6 into equation T.7 gives the fundamental natural frequency of a simply-supported beam with a concentrated mass at the centre, as:

$$f_{1} = \frac{1}{2\pi\sqrt{1 + 2m_{c}/(\rho A_{T}l_{T})}} \left(\frac{\pi}{l_{T}}\right)^{2} \sqrt{\frac{EI}{\rho A_{T}}}$$
 (T.8)

It can be seen from equation T.8 that the addition of a concentrated mass at the beam centre lowers the fundamental natural frequency.

The effect of the axial force on the fundamental natural frequency of the beam can now be included. The fundamental natural frequency of a uniform, simply-supported beam subjected to an axial compressive force F_b acting at the ends, can be obtained from the following equation [Tse et al, 1978]:

$$\omega_1^2 = \left(\frac{\pi}{l_T}\right)^4 \frac{EI}{\rho A_T} - \left(\frac{\pi}{l_T}\right)^2 \frac{F_b}{\rho A_T} \tag{T.9}$$

sents the natural frequency of the uniform beam, as given by equation that the second term is negative for a compressive force, implying that a compressive force lowers the beam natural frequency. By substitution of equation T.2 into equation T.9 and taking the square root on both sides of the resulting equation, the fundamental natural frequency of a uniform beam subjected to an axial force at the ends, is given by:

$$f_1 = \frac{1}{2\pi} \sqrt{\left(\frac{\pi}{l_T}\right)^4 \frac{EI}{\rho A_T} - \left(\frac{\pi}{l_T}\right)^2 \frac{F_b}{\rho A_T}}$$
 (T.10)

In order to include the effect of the concentrated mass on the natural frequency, the material density ρ in equation T.10 is replaced with the distributed density ρ_{distr} :

$$f_1 = \frac{1}{2\pi} \sqrt{\left(\frac{\pi}{l_T}\right)^4 \frac{EI}{\rho_{distr} A_T} - \left(\frac{\pi}{l_T}\right)^2 \frac{F_b}{\rho_{distr} A_T}}$$
 (T.11)

Finally, by substitution of equation T.6 into equation T.11, the fundamental natural frequency of a simply-supported beam with a concentrated mass attached at the centre and which is subjected to compressive axial force, is given by:

$$f_{1} = \frac{1}{2\pi} \sqrt{\left(\frac{\pi}{l_{T}}\right)^{4} \frac{EI}{\rho(1 + 2m_{c}/\rho A_{T}l_{T})A_{T}} - \left(\frac{\pi}{l_{T}}\right)^{2} \frac{F_{b}}{\rho(1 + 2m_{c}/\rho A_{T}l_{T})A_{T}}}$$
(T.12)



Appendix U Linear and nonlinear system identification models and techniques

U.1 Linear system identification models and techniques

The objective of linear identification is to obtain a system in transfer function or state-space form from measured data. Different models have been developed for this purpose, such as the following: AR (Auto Regressive), MA (Moving Average), ARX (Auto Regressive with eXogenous inputs), ARMA (Auto Regressive Moving Average), ARMAX (Auto Regressive Moving Average with eXogenous inputs) and ARIMA (Auto Regressive Integrated Moving Average). In mechanical system identification terminology, "exogenous inputs" refers to measurement noise.

The ARMA model is widely used for discrete time-domain identification of linear systems. The model is a combination of the AR and MA models. Output at the current time step is expressed in terms of previous outputs, previous inputs and the current input, using the discrete delay element z^{-1} . The current output is given by:

$$Y(z) = Y(z) \sum_{k=1}^{N} a_k z^{-k} + U(z) \sum_{k=0}^{M} b_k z^{-k}$$
 (U.1.1)

where Y(z) is the z-domain output, U(z) is the z-domain input, a_k is the k-th output delay gain and b_k is the k-th input delay gain. N is the number of output delay elements used in the model and M is the number of input delays. The z-domain TF is given by:

$$G(z) = \frac{Y(z)}{U(z)} = \frac{\sum_{k=0}^{M} b_k z^{-k}}{1 - \sum_{k=1}^{N} a_k z^{-k}}$$
 (U.1.2)

Regression methods are used to determine the gains. Regression methods will be discussed in short later.

Time-domain model from free-decay system response

The method used for obtaining system parameters from free-decay response is known as the complex exponential method. The output is written as a linear series of exponentially decaying functions in complex conjugate pairs. Each function is represented by a modal amplitude and an eigenvalue. A number of methods for determining the amplitudes and eigenvalues exist, such as the Prony and Ibrahim time-domain (ITD) methods [Ewins, 1991]. The Prony method is limited in its application by the fact that the number of poles contained in the response data must be known in advance and the data must be noise-free. Many techniques have been developed to improve the performance of the method, two of which are based on singular value decomposition (SVD) and oversized eigenmatrix methods.



Identification models for time-varying systems

Various model types may be used to represent characteristics and behaviour of time-dependent systems, such a discrete-time and state-space models. In both model types, characteristics vary with time. In discrete-time models, the a_k 's and b_k 's in equation U.1.2 are functions of time. Identification of time-varying systems is described in more detail by Bittanti et al [1992] and Kashyap & Rao [1976]. The difference between time-invariant and time-dependent state-space models was discussed in short in section 2.5 (see equations 2.5.1.1 to 2.5.1.3). In linear time-dependent state-space systems, state and output matrices are of the form A = A(t), B = B(t), C = C(t) and D = D(t). If a discrete-time model of the system is known, state and output matrices can be obtained from the coefficients a_k and b_k .

Regression techniques

Regression techniques are used to determine system model coefficients. Least-squares (LS) models are generally used for off-line identification of parameters. This method of identification may however be time-consuming.

In order to speed up the process, real-time identification tools, such as exponentially-weighted recursive least-squares (RLS) algorithms, have been developed. RLS identification is described by, inter alia, Goodwin & Sin [1984], Kim [1997] and Schoukens & Pintelon [1991]. The least-square error is given by Cowan & Grant [1985] as:

$$E_{n+1} = \sum_{i=1}^{n+1} w^{n+1-i} |\hat{y} - y|^2$$
 (U.1.3)

where E is the identification model error and w is an exponential weighting function of time, such that $0 \le w \le 1$. \hat{y} is the modelled output and y is the measured output. E is a function of the coefficient, or parameter vector, C:

$$E = E(C) \tag{U.1.4}$$

where $C = \{a_1, a_2, \dots, a_N \mid b_0, b_1, \dots, b_M\}^T$. The parameter vector is determined in real-time.

For off-line identification, the weighting function is unity. The parameter vector is obtained by setting the partial derivatives of *E* with respect to each of the parameters, equal to zero:

$$\frac{\partial E}{\partial C_l} = 0, \quad l = 1, 2, \dots, N + M + 1$$
 (U.1.5)

where C_l the *l*-th element of the coefficient vector.

For linear-in-the-parameters models, the solution of equation U.1.5 can be obtained using orthogonal, linear LS solution techniques. For nonlinear-in-the-parameters models, the



Gauss-Newton and Newton-Raphson methods are generally used. The Gauss-Newton method is described in appendix V.

Laplace- and frequency domain TF models from discrete time-domain models

The complex Laplace-domain TF of a system whose time-domain TF is known, is obtained by substitution of z with e^{sT} in equation U.1.2:

$$z = e^{sT} (U.1.6)$$

where s is the complex Laplace-domain differential operator and T is sample time. The transfer function G(s) can be written in the form:

$$G(s) = \frac{\sum_{k=0}^{M} p_k s^k}{\sum_{k=0}^{N} q_i s'}$$
 (U.1.7)

where the p_k 's and q_l 's are the numerator and denominator polynomial coefficients respectively.

The complex ω -domain TF of a system whose time-domain TF is known, is obtained by substitution of z with $e^{j\omega T}$ in equation U.1.2:

$$z = e^{j\omega T} (U.1.8)$$

The transfer function $G(j\omega)$ can be written in the following form:

$$G(j\omega) = \frac{\sum_{k=0}^{M} p_k (j\omega)^k}{\sum_{l=0}^{N} q_l (j\omega)^l}$$
 (U.1.9)

where the p_k 's and q_l 's are the numerator and denominator polynomial coefficients respectively, which are equal to those in equation U.1.7.

The polynomial coefficients can be determined if the a_k 's and b_k 's in equation U.1.2 are known. If not, the TF model must be available in some other form, such as measured spectra.



Frequency-domain models from measured TF spectra

The objective is to obtain system TF spectra that match the experimentally measured spectra as closely as possible. A vast number of ω domain identification techniques exist. The technique used will mainly depend on system characteristics, e.g. linearity and number of degrees of freedom. Linear frequency-domain identification techniques can broadly be classified as SDOF and MDOF identification techniques.

Frequency-domain identification of an SDOF system can be done by means of the peak-amplitude method, circle fit method and the so-called inverse method [Ewins, 1991]. Receptance properties are used in the peak amplitude and circle fit methods to determine system characteristics, while inverse receptance properties are used in the "inverse method".

The receptance curve (X/F) of an SDOF system displays, inter alia, a peak amplitude and two frequencies where the amplitudes are 70,7% of the peak amplitude. The frequency where the peak occurs, is the natural frequency f_m . The two frequencies where the amplitudes are 70,7% of the peak amplitude, are the half-power points. The half-power bandwidth is denoted by Δf . The dimensionless damping factor ζ is given by the equation $\zeta = 2\Delta f/f_n$. This identification method is known as the peak amplitude method. Stiffness and mass are respectively given by:

$$k = [(X/F)|_{f=0}]^{-1}$$
 (U.1.10a)

$$m = k/(2\pi f_n)^2$$
 (U.1.10b)

The Nyquist plot of the receptance of an SDOF system is a circle in the complex plane. This circle is known as the modal circle. Natural frequency, damping factor and modal amplitude can be calculated from the circle properties. The procedure is not given here, but is described in full by Ewins [1991]. This method is known as the circle-fit method and is widely used for frequency-domain identification of SDOF systems.

An alternative to the circle-fit method, is the "inverse" method. In the complex plane, the inverse receptance curve (F/X) produces a horizontal straight line. The maximum real value of F/X is k, the spring constant. The intercept of the line with the imaginary axis gives the damping $c\omega$ directly. The frequency at the origin of the graph, i.e. where the real and imaginary axes cross, is the natural frequency.

Three types of MDOF frequency-domain identification techniques are discussed here. The first type is the circle-fit method, which is an extension of the SDOF circle-fit method. The method is described by Ewins [1991]. The method cannot be used if the modes are closely-coupled or if damping is extremely light. (Closely coupled modes occur in systems where the natural frequencies are closely-spaced or where damping is high, or both).

A more general approach to multi-mode curve-fitting is an LS data fit to measured *TF* frequency spectra in complex form. Depending on reliability of measured data and system characteristics, more accurate data fits are generally obtained than with the circle fit method. One disadvantage of the method is that the LS problem is nonlinear, which requires iterative solution techniques, such as the Newton-Raphson and Gauss-Newton methods. As is the



case with all iterative techniques, starting values in the range of convergence of the numerical system, are required. Furthermore, poles of the fitted *TF* may be unstable, even for globally stable plants. The LS method for frequency-domain identification is discussed in short by Ewins [1991].

The third method is identification using overdetermined systems, as described by Braun & Ram [1987]. Model order is assumed higher than that of the physical system. The method was developed to overcome the problem of specifying model order in the presence of noise. Singular value decomposition is used to solve the LS data fitting problem.

State-space identification of linear systems

State-space identification techniques can broadly be classified as direct and indirect techniques. Indirect techniques are applied to convert existing time-, frequency- or Laplace-domain models to state-space models. Direct models are used to obtain state-space models from measured spectra without intermediate steps.

Conversion of s-domain models to state-space models is described by Schwarzenbach & Gill [1986]. The TF(Y/U) is written as the product of two functions in s:

$$\frac{Y(s)}{U(s)} = \frac{Y(s)}{V(s)} \frac{V(s)}{U(s)}$$
 (U.1.11)

where Y(s)/V(s) is the numerator, divided by the highest power of s, and V(s)/U(s) is the denominator, divided by the highest power of s. V/U is written as a linear DE and expressed in canonical state-space form. The output is expressed in terms of the state vector and input, in the form of equation 2.5.1.3b.

A similar technique can be used to convert discrete-time models to discrete state-space form if the a_k and b_k coefficients in equation U.1.2 are known. The state and output equations of a linear discrete system are:

$$x(k+1) = \Phi x(k) + \Gamma u(k)$$
 (U.1.12a)

$$y(k) = Hx(k) + Ju(k)$$
 (U.1.12b)

where x(k), u(k) and y(k) are the k-th state vector, input and output respectively. Φ , Γ , H and J are the coefficient-, driving-, output- and transmission matrix, respectively.



The discrete state and output matrices are related to their respective continuous equivalents by means of the following equations [Franklin et al, 1990] and [Liu & Miller, 1995]:

$$\Phi = e^{AT} \tag{U.1.13a}$$

$$\Gamma = \int_{0}^{T} e^{A\eta} B d\eta \qquad (U.1.13b)$$

$$H = C \qquad (U.1.13c)$$

$$H = C \tag{U.1.13c}$$

$$J = D \tag{U.1.13d}$$

where T is the sample period and η is an integration variable.

State-space models can be directly obtained from measured spectra, using, for instance, the Frequency domain Observability Range Space Extraction (FORSE) algorithm. algorithm is explained in detail by Liu et al [1996]. It uses proper state-space parameterization and Singular Value Decomposition (SVD) to ensure good numerical properties for multivariable and high dimensional structural systems. It achieves high accuracy in the frequency domain by overparameterizing the model. The FORSE algorithm is an extension of the ORSE time-domain identification algorithm [Liu & Miller, 1995].

U.2 Nonlinear system identification models and techniques

Techniques for determining the parameters of nonlinear systems from measured data can broadly be classified as time-domain techniques, frequency-domain techniques, force-state component identification and identification using neural networks.

Time-domain identification techniques for nonlinear systems

Similar models to the linear time-invariant models have been developed for time-domain identification of nonlinear systems. Two of the most popular models are as the NARX (Nonlinear Auto Regressive with eXogenous inputs) and NARMAX (Nonlinear Auto Regressive Moving Average with eXogenous inputs) models. The NARX model is a special case of the NARMAX model. The NARMAX model is described by Chen et al [1989] and Worden et al [1994]. Othogonal LS data fit techniques, such as the classical Gram-Schmidt (CGS) and modified Gram-Schmidt (MGS) schemes, Householder transformation and Givens method can be used to determine model parameters [Chen et al, 1989].



Force-state component identification

Force-state component identification is done to obtain the restoring force f in a component or structure in terms of the state vector x and its derivative with respect to time. \dot{x} .

The equation of motion is written in the following form:

$$m\ddot{x} + f(x, \dot{x}) = p(t) \tag{U.2.1}$$

where p(t) is the load vector.

f can be determined for a large variety of nonlinearities, e.g. nonlinear spring and nonlinear damping characteristics, nonlinearities with memory and combinations of linear and nonlinear stiffness and damping.

Characteristics may be represented mathematically, i.e. by $f(x, \dot{x})$, or graphically, by a 3-dimensional force-state map. Force-state component identification is described in more detail by Masters et al [1996], Worden & Tomlinson [1988] and Wright & Al Haddid [1991].

Frequency-domain identification of nonlinear systems

Nonlinear frequency-domain identification can broadly be classified as SDOF identification, modal analysis (for MDOF systems), and identification using time-domain inputs. The SDOF and modal analysis identification techniques are, in essence, extensions of linear system identification methods. Two SDOF identification techniques are the circle fit method, as described by Ewins [1991] and the use of response-surface plots, as explained by Dimas & Pardoen [1988].

Modal analysis identification methods for MDOF systems include the Hilbert transform method, the use of functional series (Volterra method) and range-dependent linear identification. The Hilbert transform method is a well-known linearization technique in the fields of numerical signal processing of filters, acoustics and physics. It is extended to MDOF mechanical system identification by Vinh et al [1984] and Simon & Tomlinson [1984]. The frequency response function (*FRF*) of an MDOF system, for a sinusoidal input, is expressed in terms of an equivalent linear *TF*. The method can be applied to systems with strong nonlinearities and no a priori assumption of the nonlinearities is required. The disadvantage, however, is that the method is more difficult to apply to identification procedures where random excitation is used.

The Volterra system model consists of two parts, i.e. a linear system model and a nonlinear extension to the model, using higher order terms. In the ω -domain, the linear term is expressed by a classical Fourier transform of the TF. The nonlinear terms are written as multiple convolutions, using Volterra kernels. The ω -domain representation of the nonlinear part is obtained by taking the multi-dimensional Fourier transform of the higher order terms. Two difficulties associated with the method are the number of terms required for accurate



modelling and efficient methods of computing multi-dimensional Fourier transforms (MDFT's). The Volterra method is explained in more detail by Tomlinson [1986].

A third method is range-dependent linearization. The system is assumed linear for a given input energy level. Linear identification is done for each input energy level, using linear MDOF techniques. The effects of nonlinearities on the ω -domain FRF for a constant input energy level are ignored. This method is applicable if the system is weakly nonlinear or if the nonlinearity only has a significant influence outside the frequency band of interest. The disadvantage is that characterization tests must be done for each input energy level in order to determine the effect of the nonlinearity on the FRF.

Nonlinear frequency-domain identification is often done from measured time-domain inputs and responses, using methods such as the NARMAX method. Application of the NARMAX method to nonlinear frequency response estimation is described by Billings et al [1988]. The NARMAX method is a convenient method of expressing inputs and outputs as higher order series. The model is validated in the time domain and transformed to the frequency-domain. The procedure is as follows: The input is first assumed to be harmonic. The input is substituted in the NARMAX model and the output amplitude at the given frequency is calculated, using harmonic balance. A second harmonic is added to the input. The output amplitude is calculated as a function of the two input frequencies. The procedure is repeated until the amplitudes of all the desired frequencies have been obtained.

Identification using neural networks

Neural network identification makes use of parallel distributed processing to identify the internal forces of structure-unknown nonlinear dynamic systems typically encountered in the field of applied mechanics [Masri et al, 1992]. Identification of the restoring force of an SDOF system is described by Masri et al [1993]. The restoring force $g(y, \dot{y})$ is written in terms of the measured input u(t), acceleration $\ddot{y}(t)$ and mass m as:

$$g(y,\dot{y}) = u(t) - m\ddot{y}(t) \tag{U.2.2}$$

Measured signals are discretized and fed into the neural network. Network output is an approximate restoring force $\hat{g}(y,\dot{y})$. The identification procedure consists of two phases, i.e. the network learning phase and the evaluation phase. The procedure is described in detail by Masri et al [1993].



Appendix V Gauss-Newton method applied to LS identification in the frequency domain

The difference between the modelled TF and experimentally-determined TF is known as the estimation error and is given by:

$$e = G - \hat{G} \tag{V.1}$$

where e is the error, \hat{G} is the modelled TF and G is the experimentally-determined TF. G is known at discrete frequencies, thus the error can only be evaluated at these frequencies. The error e_r at the r-th frequency is:

$$e_r = G_r - \hat{G}_r \tag{V.2}$$

where G_r and \hat{G}_r are the measured TF and approximate TF at the r-th frequency, given by:

$$G_r = G(j\omega_r) (V.3a)$$

$$\hat{G}_r = \hat{G}(j\omega_r) \tag{V.3b}$$

It can be seen from equations V.2 and V.3 that the error varies with frequency. As a globally accurate model is required, the error squared is summed and minimized over the frequency band. The global error is given by:

$$E = \sum_{r=1}^{R} e_r^2 \tag{V.4}$$

where E is the global error and of the model and R is the number of discrete frequencies.

Both G and \hat{G} are complex (see equations V.3). The global error can therefore be expressed as the sum of the error of the real term squared, plus the error of the imaginary term squared [Schoukens & Pintelon, 1991]:

$$E = \sum_{r=1}^{R} \left[\left(\text{Re} \{ G_r \} - \text{Re} \{ \hat{G}_r \} \right)^2 + \left(\text{Im} \{ G_r \} - \text{Im} \{ \hat{G}_r \} \right)^2 \right]$$
 (V.5)

E is a nonlinear function of the polynomial coefficient vector C:

$$E = E(C) (V.6)$$



C is given as follows in terms of the TF numerator and denominator polynomial coefficients:

$$C = \{p_0, p_1, \dots, p_M, q_0, q_1, \dots, q_N\}^T$$
(V.7)

where M and N are the numerator and denominator polynomial orders respectively.

From equation V.4, E can be expressed in matrix notation as:

$$E = e^T e (V.8)$$

Substitution of equation V.1 into equation V.8 gives:

$$E = (G - \hat{G})^{T} (G - \hat{G})$$
 (V.9)

where G and \hat{G} are $R \times 1$ vectors and E is a 1 x 1 scalar.

The objective of LS identification is to determine the parameter vector C, which minimizes E. C is obtained by setting the partial derivatives of E with respect to each of the parameters C, equal to zero and by solving for C:

$$\frac{\partial E}{\partial C_l} = 0, \quad l = 1, 2, \dots, N + M + 2$$
 (V.10)

where C_l is the *l*-th element of the coefficient vector.

The solution of equation V.10 is determined iteratively. An initial estimate of C is made, which will not necessarily satisfy equation V.10. The initial estimate is known as the starting value. Using the estimated C, the error gradient $\partial E/\partial C$ is calculated, from which the change ΔC in the parameter vector is obtained. C is added to ΔC to obtain an updated parameter vector. The updated C is subsequently used to calculate $\partial E/\partial C$. If the error gradient is sufficiently close to zero, the procedure is stopped. This condition is known as the stopping criterion. If the stopping criterion is not met, ΔC , C and $\partial E/\partial C$ are updated and the procedure is repeated until the stopping criterion is met.

The starting value must be selected within a limited range, known as the system range of convergence. If the starting value is selected outside this range, the method will diverge. If the starting value is selected inside the range of convergence, the method will either converge to a local minimum, or to the global minimum. The selection of starting values of a system with unknown coefficients is generally difficult. The method suggested by Schoukens & Pintelon [1991] is to linearize equation V.4, in order to obtain the linear solution to equation V.10 and to use this solution as a starting value. With a known starting value, ΔC can be calculated, using the procedure set out below.



For a small increment ΔC in equation V.6, the estimation error can be written as follows, using a Taylor series expansion:

$$E(C + \Delta C) = E(C) + (\Delta C)^{T} \frac{\partial E}{\partial C} + \frac{1}{2} (\Delta C)^{T} \frac{\partial^{2} E}{\partial C^{2}} \Delta C + \dots +$$
 (V.11)

where C and $\partial E/\partial C$ are $(N+M+2) \times 1$ vectors. $\partial^2 E/\partial C^2$ is an $(N+M+2) \times (N+M+2)$ matrix.

The partial derivative of equation V.11 with respect to ΔC is:

$$\frac{\partial E(C + \Delta C)}{\partial \Delta C} = \frac{\partial E(C)}{\partial \Delta C} + \frac{\partial E}{\partial C} + \frac{\partial^2 E}{\partial C^2} \Delta C + \dots +$$
 (V.12)

For E to be a minimum, the partial derivative of $E(C + \Delta C)$ with respect to ΔC must be zero:

$$\frac{\partial E(C + \Delta C)}{\partial \Delta C} = 0 \tag{V.13}$$

The first term on the right hand side of equation V.11 is constant, therefore:

$$\frac{\partial E(C)}{\partial \Delta C} = 0 \tag{V.14}$$

Substitution of equations V.13 and V.14 into equation V.12 gives:

$$\frac{\partial E}{\partial C} = -\frac{\partial^2 E}{\partial C^2} \Delta C \tag{V.15}$$

 ΔC is obtained from equation V.15 as:

$$\Delta C = -\left(\frac{\partial^2 E}{\partial C^2}\right)^{-1} \frac{\partial E}{\partial C}$$
 (V.16)

It can be seen from equation V.16 that ΔC depends on both 1st order and 2nd order derivatives of E with respect to C. It is however more convenient to express ΔC in terms of the 1st order derivative only. This is done with the aid of equation V.9.

The 1^{st} order partial derivative of E with respect to C, from equation V.9, is given by:

$$\frac{\partial E}{\partial C} = -2 \left(\frac{\partial \hat{G}}{\partial C} \right)^{T} \left(G - \hat{G} \right)$$
 (V.17)

where $\partial \hat{G}/\partial C$ is an $R \times (N+M+2)$ matrix.



 $\partial^2 E/\partial C^2$ is obtained by partial differentiation of equation V.17 with respect to C. For a small error, $\partial^2 E/\partial C^2$ is given by:

$$\frac{\partial^2 E}{\partial C^2} = 2 \left(\frac{\partial \hat{G}}{\partial C} \right)^T \left(\frac{\partial \hat{G}}{\partial C} \right)$$
 (V.18)

Substitution of equations V.17 and V.18 into equation V.16 gives:

$$\Delta C = \left[\left(\frac{\partial \hat{G}}{\partial C} \right)^T \left(\frac{\partial \hat{G}}{\partial C} \right) \right]^{-1} \left(\frac{\partial \hat{G}}{\partial C} \right)^T \left(G - \hat{G} \right)$$
 (V.19)

Methods of determining the stopping criterion are explained by Schoukens & Pintelon [1991]. For the purpose of this study, the stopping criterion can either be a relatively small change in E, or a negligible increment ΔC in the parameter vector. The former stopping criterion may be mathematically expressed as:

$$\frac{E(C + \Delta C)}{E(C)} \le \alpha \tag{V.20}$$

where α is an arbitrarily selected factor, e.g. 0,1%.

If system sensitivity to parameter variations is low, equation V.20 will not necessarily guarantee a negligible ΔC . For systems of this type, the following stopping criterion is recommended:

$$\frac{\Delta C_l}{C_l} \le \alpha \tag{V.21}$$

The stopping criterion given in equation V.21 is generally stricter than that in equation V.20, since the latter may give a parameter vector whose elements do not satisfy equation V.21.



Appendix W Derivation of system transmissibility and two-input model

System transmissibility (TR) can be expressed in terms of transfer functions such as the input force to output force TF, base angular displacement to output angular displacement TF (θ/θ_b) , base angular velocity to output angular velocity TF $(\dot{\theta}/\dot{\theta}_b)$ and base angular acceleration to output angular acceleration TF $(\ddot{\theta}/\ddot{\theta}_b)$. The most appropriate transfer function is mainly determined by the sensors used for system characterization and control.

Accelerometers were used for dynamic system characterization in chapter 4. The measured vertical accelerations were subsequently used to calculate the base and output angular accelerations (see section 4.5.1). Accelerometers will also be used as sensors during testing of the control system. It is therefore appropriate to express the TR as the transfer function between base angular acceleration and output angular acceleration ($\ddot{\theta}/\ddot{\theta}_b$).

The transmissibility is required in terms of system characteristics, such as state and output parameters, natural frequencies, damping factors and normal mode shapes. A derivation of the TR follows below. In order to simplify the derivation, an assumption is made that the TR is independent of sensor dynamics. The consequence of this assumption is that the transfer functions, θ/θ_b , $\dot{\theta}/\dot{\theta}_b$ and $\ddot{\theta}/\ddot{\theta}_b$, are equal over a wide frequency bandwidth. This makes it possible to derive the transfer function θ/θ_b and set it equal to $\ddot{\theta}/\ddot{\theta}_b$:

$$TR = \frac{\ddot{\theta}}{\ddot{\theta}_b} = \frac{-\omega^2 \theta}{-\omega^2 \theta_b} = \frac{\theta}{\theta_b}$$
 (W.1)

Consider figure 5.5.1. The displacement of the moving base is denoted by y_b . The base mass, stiffness and excitation force, are respectively denoted by k_b , m_b and F_b .

In this derivation, y_h is a known signal, i.e. the disturbance signal. For a known base displacement, the system displacement vector is given by:

$$y = \{ \{y_{a1}\} \quad w_s(x) \quad \{y_{a2}\} \}^T$$
 (W.2)

The equation of motion of actuator II, for a base translational displacement input y_b , is:

$$\begin{bmatrix} m_{a21} & 0 \\ 0 & m_{a22} \end{bmatrix} \begin{Bmatrix} \ddot{y}_{a21} \end{Bmatrix} + \begin{bmatrix} k_{211} + k_{212} & -k_{212} \\ -k_{212} & k_{212} + k_{222} \end{bmatrix} \begin{Bmatrix} y_{a21} \\ y_{a22} \end{Bmatrix} = \begin{Bmatrix} k_{211} \\ k_{222} \end{Bmatrix} y_b$$
 (W.3)



The equation of motion, for a base angular displacement input θ_b , is:

$$\begin{bmatrix} m_{a21} & 0 \\ 0 & m_{a22} \end{bmatrix} \begin{Bmatrix} \ddot{y}_{a21} \\ \ddot{y}_{a22} \end{Bmatrix} + \begin{bmatrix} k_{211} + k_{212} & -k_{212} \\ -k_{212} & k_{212} + k_{222} \end{bmatrix} \begin{Bmatrix} y_{a21} \\ y_{a22} \end{Bmatrix} = l \begin{Bmatrix} k_{211} \\ k_{222} \end{Bmatrix} \theta_b$$
 (W.4)

where l is the horizontal distance between the two actuators.

The equation of motion of actuator II, in modal coordinates, is:

$$M_{a1} \begin{Bmatrix} \ddot{q}_{a21} \\ \ddot{q}_{a22} \end{Bmatrix} + K_{a1} \begin{Bmatrix} q_{a21} \\ q_{a22} \end{Bmatrix} = \{F_{a2}\}$$
 (W.5)

where $\{F_{a2}\}$ is the modal excitation force vector, given by:

$${F_{a2}} = l \psi_{a2}^T {k_{211} \choose k_{222}} \theta_b$$
 (W.6)

The system equation of motion, in component modal coordinates, is:

$$M\ddot{q} + Kq = F \tag{W.7}$$

where F is the modal base excitation force.

M, K, q and F are respectively given by:

$$M = \begin{bmatrix} \begin{bmatrix} M_{a1} \end{bmatrix} & & \\ & \begin{bmatrix} M_s \end{bmatrix} & \\ & \begin{bmatrix} M_{a2} \end{bmatrix} \end{bmatrix}$$
 (W.8a)

$$K = \begin{bmatrix} [K_{a1}] \\ [K_s] \\ [K_{a2}] \end{bmatrix}$$
 (W.8b)

$$q = \begin{cases} \left\{q_{a1}\right\} \\ \left\{q_{x}\right\} \\ \left\{q_{a2}\right\} \end{cases}$$
 (W.8c)

$$F = \begin{cases} \{0\} \\ \{0\} \\ \{F_{a2}\} \end{cases}$$
 (W.8d)

The sizes of M and K are 14 x 14 each, while those of q and F are 14 x 1 each.



The equations of motion of the actuators and structure are currently uncoupled. Coupling is done by means of the modal coupling equation:

$$q = Cq_r (W.9)$$

where q_r is the coupled modal displacement vector and C is the coupling matrix, given by:

$$C = \begin{bmatrix} 1 & \{0\} & 0 \\ 1 & -\{\psi_s(0)\} & 0 \\ \{0\} & [I] & \{0\} \\ 0 & \{0\} & 1 \\ 0 & -\{\psi_s(I)\} & 1 \end{bmatrix}$$
 (W.10)

The sizes of C and q_r are 14 x 12 and is 12 x 1 respectively. The coupling matrix given above, is the C-matrix given in equation 5.5.38, whose last row and last column are deleted.

The coupled system modal equation of motion is:

$$M_r \ddot{q}_r + K_r q_r = F_r \tag{W.11}$$

where M_r and K_r are given by equations 5.5.41 and F_r is given by:

$$F_r = C^T F \tag{W.12}$$

The reduced coordinate vector q_r is written as follows in terms of the eigenvector U and the normal mode displacement vector q_n (see equation 5.5.42):

$$q_r = Uq_n \tag{W.13}$$

The sizes of U and q_n are 12 x 12 and 12 x 1 respectively.

Substitution of equation W.13 into equation W.11 and premultiplication of the resulting equation with U^r , gives:

$$M^*\ddot{q}_n + K^*q_n = Q \tag{W.14}$$

where M^* and K^* are the normal mode mass and stiffness matrices respectively, as given by equations 5.5.48. Q is the modal excitation force vector, given by:

$$Q = U^T C^T F (W.15)$$

Premultiplication of equation W.14 with M^{-1} gives the normal mode equation of motion in the following form:

$$\ddot{q}_{n} + \Omega^{2} q_{n} = M^{*-1} Q$$
 (W.16)

where Ω is the diagonal natural frequency matrix.



Adding modal damping terms to equation W.16 gives:

$$\ddot{q}_n + 2Z\Omega\dot{q}_n + \Omega^2 q_n = M^{\bullet - 1}Q \tag{W.17}$$

The modal state and output equations can be written as follows:

$$\dot{x} = A_b x + B_b \theta_b \tag{W.18a}$$

$$\theta = C_b x + D_b \theta_b \tag{W.18b}$$

The state vector *x* is:

$$x = \begin{cases} q_n \\ \dot{q}_n \end{cases}$$
 (W.19)

 A_h , B_h , C_h and D_h are respectively given by:

$$A_b = \begin{bmatrix} 0 & I \\ -\Omega^2 & -2Z\Omega \end{bmatrix}$$
 (W.20a)

$$B_{b} = \begin{cases} \{0\} \\ \{0\} \\ M^{\bullet - 1} U^{T} C^{T} \\ \{0\} \\ l \psi_{a2}^{T} \begin{cases} k_{211} \\ k_{222} \end{cases} \} \end{cases}$$
 (W.20b)

$$C_b = \left\{ \frac{\Phi_x(x_2) - \Phi_x(x_1)}{x_2 - x_1} \quad \{0\} \right\}$$
 (W.20c)

$$D_b = 0 (\mathbf{W.20d})$$

The sizes of A_h , B_h , C_h and D_h are 24 x 24, 24 x 1, 1 x 24 and 1 x 1 respectively.



Combining equations W.18 to W.20 gives:

$$\theta = \left[\frac{\Phi_s(x_2) - \Phi_s(x_1)}{x_2 - x_1} \quad \{0\}\right] \begin{Bmatrix} q_n \\ \dot{q}_n \end{Bmatrix}$$
 (W.21b)

The system transmissibility is the TF between θ_b and θ is given by:

$$\frac{\theta}{\theta_b} = C_b [sI - A_b]^{-1} B_b \tag{W.22}$$

By application of equation W.1, the TR, i.e. the TF between $\ddot{\theta}_b$ and $\ddot{\theta}$, is given by:

$$\frac{\ddot{\theta}}{\ddot{\theta}_b} = C_b [sI - A_b]^{-1} B_b \tag{W.23}$$

The number of state equations, for the selected number of actuator and optical instrument assumed modes, is 24. The system as described by equations W.21 has 24 eigenvalues in complex conjugate pairs. However, many of the eigenvalues occur at high frequencies, well above the bandwidth, and have no significant effect on system performance inside the band. States corresponding with these frequencies, can therefore be eliminated. This was also done for the coil voltage input, where, apart from the three coil states, only the first ten mechanical subsystem states were retained. For the base input, the 5th and 10th states (corresponding with the 5th normal mode) are eliminated, while the 1st to 4th and 6th to 9th states (corresponding with the first four normal modes) are retained.

The transmissibility magnitude and phase spectra, for a frequency bandwidth of 0 to 500 Hz, are shown in figure W.1.



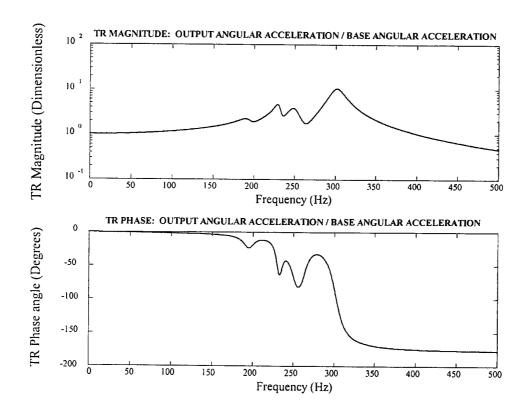


Figure W.1: System base to output transmissibility spectra

Two-input state-space model for control system analysis and design purposes

The state-space model of the system currently consists of two separate state models, one for the coil voltage input, given by equations 5.6.6 to 5.6.9, and one for the base angular acceleration input, given by equations W.18 to W.23. For the purpose of analyzing and designing the control system, it is convenient to combine the two state models into a single state model, with two simultaneous inputs, i.e. coil voltage and base angular acceleration. The coil voltage is the controllable input, while the base motion is the disturbance. The state and output equations, for these two inputs, are:

$$\dot{x} = Ax + B \begin{Bmatrix} V \\ \ddot{\theta}_h \end{Bmatrix}$$
 (W.24a)

$$\dot{x} = Ax + B \begin{Bmatrix} V \\ \ddot{\theta}_h \end{Bmatrix}$$

$$\ddot{\theta} = Cx + D \begin{Bmatrix} V \\ \ddot{\theta}_h \end{Bmatrix}$$
(W.24a)
(W.24b)

The state vector, from equation 5.6.6, is:

$$x = \begin{cases} q_n \\ \dot{q}_n \\ x_c \end{cases}$$
 (W.25)



A is the system coefficient matrix, given in equation 5.6.9a:

$$A = \begin{bmatrix} [0] & [I] & [0] \\ -\Omega^2 & -2Z\Omega & M^{*-1}\Phi_x^T F_I C_c \\ [0] & [0] & A_c \end{bmatrix}$$
 (W.26)

The first ten rows and columns of A correspond with A_b in equation equation W.20a.

B is a matrix with two vector columns:

$$B = [B_{\nu} \quad B_{\ddot{a}k}] \tag{W.27}$$

 B_V is given by equation 5.6.9b:

$$B_{\nu} = \begin{cases} \{0\} \\ \{0\} \\ B_{c} \end{cases}$$
 (W.28a)

For the first four normal modes, $B_{\partial h}$ is given by:

$$B_{\theta b} = \begin{cases} \left\{ B_{b} \right\}_{rows1-4} \\ 0 \\ \left\{ B_{b} \right\}_{rows13-16} \\ 0 \\ 0 \\ 0 \\ 0 \end{cases}$$
 (W.28b)

C is the output matrix given in equation 5.6.9c:

$$C = \frac{\Phi_s(x_2) - \Phi_s(x_1)}{x_2 - x_1} \left[-\Omega^2 - 2Z\Omega M^{*-1} \Phi_s^T F_i C_c \right]$$
 (W.29)

D is a row vector with two zero elements:

$$D = \{0 \quad 0\} \tag{W.30}$$

Note that the C-matrix in equation W.29 corresponds with the C-matrix in equation 5.6.9c. However, this C-matrix does not correspond with C_h in equation W.20c. In order to obtain a C-matrix, which is valid for both inputs, the elements of $B_{\partial h}$ are modified as explained next.



The procedure is as follows: Equations W.21 are written in canonical form, whereby uncoupled state and output equations for each normal mode are obtained. Each modal state equation is then written as follows in *TF* form, using equation W.1:

$$\left(\frac{\ddot{\theta}}{\ddot{\theta}_{b}}\right)_{i} = \left(\frac{\theta}{\theta_{b}}\right)_{i} = C_{i}[sI - A_{i}]^{-1}B_{i}$$
 (W.31)

where the subscript i denotes the i-th normal mode. The sizes of A_i , B_i , C_i and D_i are respectively 2×2 , 2×1 , 1×2 and 1×1 .

Equation W.31 is written as follows for the *i*-th mode:

$$\left(\frac{\ddot{\theta}}{\ddot{\theta}_{b}}\right)_{i} = \left\{C_{1} \quad C_{2}\right\}_{i} \begin{bmatrix} \frac{s - A_{22}}{s^{2} - (A_{11} + A_{22})s + A_{11}A_{22} - A_{21}A_{12}} & \frac{A_{12}}{s^{2} - (A_{11} + A_{22})s + A_{11}A_{22} - A_{21}A_{12}} \\ \frac{A_{21}}{s^{2} - (A_{11} + A_{22})s + A_{11}A_{22} - A_{21}A_{12}} & \frac{s - A_{11}}{s^{2} - (A_{11} + A_{22})s + A_{11}A_{22} - A_{21}A_{12}} \end{bmatrix}_{i} \begin{Bmatrix} B_{1} \\ B_{2} \end{Bmatrix}_{i}$$
(W.32)

Expanding equation W.32 gives:

$$\left(\frac{\ddot{\theta}}{\ddot{\theta}_b}\right)_i = \left(\frac{p_1 s + p_0}{s^2 + q_1 s + q_0}\right)_i \tag{W.33}$$

where:

$$p_{0i} = \left(-A_{11}B_2C_2 + A_{12}B_2C_1 + A_{21}B_1C_2 - A_{22}B_1C_1\right)_i$$
 (W.34a)

$$p_{1i} = (B_1C_1 + B_2C_2)_i$$
 (W.34b)

$$q_{0i} = (A_{11}A_{22} - A_{12}A_{21})_i$$
 (W.34c)

$$q_{1i} = (-A_{11} - A_{22})_i \tag{W.34d}$$

The input voltage to angular acceleration TF is subsequently written in canonical form to obtain an uncoupled state and output equation for each of the first four normal modes. The modal state equation for the i-th mode is then written as follows, in the TF form of equation W.33:

$$\left(\frac{\ddot{\theta}}{V}\right)_{i} = C_{i} \left[sI - A_{i}\right]^{-1} B_{i} \tag{W.35}$$



Equations W.34a and W.34b are written in matrix form as:

$$\begin{bmatrix} C_1 & C_2 \\ A_{21}C_2 - A_{22}C_1 & A_{12}C_1 - A_{11}C_2 \end{bmatrix}_i \begin{cases} B_1 \\ B_2 \\ i \end{cases} = \begin{cases} p_1 \\ p_0 \\ i \end{cases}$$
 (W.36)

where C_{1i} and C_{2i} are the elements of C_i in equation W.35. From equation W.36, B_{1i} and B_{2i} are determined as follows:

$$\begin{cases}
B_1 \\
B_2
\end{cases}_{i} = \begin{bmatrix}
C_1 & C_2 \\
A_{21}C_2 - A_{22}C_1 & A_{12}C_1 - A_{11}C_2
\end{bmatrix}_{i}^{-1} \begin{Bmatrix} p_1 \\ p_0 \end{Bmatrix}_{i}$$
(W.37)

The resulting $B_{\ddot{\theta}b}$ -vector, is:

$$B_{\ddot{\partial}b} = \{ (B_1)_{i=1} \quad (B_1)_{i=2} \quad (B_1)_{i=3} \quad (B_1)_{i=4} \quad 0 \quad (B_2)_{i=1} \quad (B_2)_{i=2} \quad (B_2)_{i=3} \quad (B_2)_{i=3} \quad 0 \quad 0 \quad 0 \quad 0 \}^T$$
(W.38)

Equations W.24 therefore become:

$$\begin{cases}
\dot{q}_n \\
\ddot{q}_n \\
\dot{x}_c
\end{cases} = \begin{bmatrix}
0 \\
-\Omega^2 \\
0 \end{bmatrix} \begin{bmatrix}
I \\
I \\
0 \\
I \\
0 \end{bmatrix} \begin{bmatrix}
0 \\
A_c
\end{bmatrix} \begin{cases}
q_n \\
\dot{q}_n \\
x_c
\end{bmatrix} + \begin{bmatrix}B_{\nu} \\ B_{\theta h}\end{bmatrix} \begin{cases}
V \\
\ddot{\theta}_h
\end{cases}$$
(W.39a)

$$\ddot{\theta} = \frac{\Phi_{s}(x_{2}) - \Phi_{s}(x_{1})}{x_{2} - x_{1}} \left[-\Omega^{2} - 2Z\Omega M^{*-1}\Phi_{s}^{T}F_{l}C_{c} \right] \begin{pmatrix} q_{n} \\ \dot{q}_{n} \\ x_{c} \end{pmatrix}$$
 (W.39b)



Appendix X General control system requirements

General controller requirements are normally expresessed in terms of performance factors such as accuracy, stability, sensitivity, reaction speed, control effort, robustness, observability and controllability. These factors are discussed in short below, together with methods to meet the requirements.

X.1 Accuracy

Control accuracy is influenced by parameters such as transient overshoot, settling time, DC error, tracking error and disturbance attenuation over a given bandwidth. Excessive transient overshoot may damage the plant and must therefore be reduced to within a safe range. Two simple methods of transient overshoot control, are rate feedback and PID-control [Schwarzenbach & Gill, 1986]. Differentiation of the error signal, by means of a differentiator (D) element, in parallel with the P and I elements, will lower the overshoot peak. The higher the D-element gain, the lower the overshoot. However, the higher the D-element gain, the longer the transient rise time.

Rate feedback is accomplished by placing a differentiator in parallel with the output feedback elements. A disadvantage of rate feedback is that it may drive the plant unstable if control is non-collocated.

Settling time is the time it takes the system output to reach and stay within 5% of the steady-state value, during step excitation [Schwarzenbach & Gill, 1986]. The settling time T of a system with distinct, complex poles, can be obtained from the following approximate equation:

$$T \approx \frac{5}{\zeta_1 \omega_1} \tag{X.1.1}$$

where ω_1 is the lowest angular natural frequency and ζ_1 is the corresponding damping factor.

DC error is the deviation of the system output from the setpoint, after the settling time has elapsed. A simple method of eliminating DC error is to integrate the error signal, by means of an integrator element, placed in parallel with the proportional element. The disadvantage of this measure is that it will lower the damping required to limit transient overshoot [Schwarzenbach & Gill, 1986].

Tracking error is the relative error between the reference signal and output. It is, as the name indicates, only applicable to tracking controllers. Tracking error may depend on a variety of factors, such as the transient phenomena discussed above, system frequency bandwidth, nonlinearities, measurement noise, control hardware throughput delay, as well as sensor range, resolution and bandwidth. All of these factors must necesssarily be addressed during the design of a tracking controller. This is, however, beyond the scope of this study.

Disturbance attenuation, on the other hand, is important in this study. The plant must provide the necessary filtering characteristics to reject the base disturbance, in order to obtain a sufficiently low output angular acceleration. Due to noise and unmodelled plant characteristics, such as nonlinearities, it will be extremely difficult to totally reject the disturbance. Instead, the remaining output must be minimized. The ratio of the remaining output to the base disturbance, is known as the attenuation factor. This factor is a measure of determining the performance of the system. The attenuation factor was discussed in more detail in section 6.3.1.

Control accuracy may also be affected by the control system type. It may for instance be attractive, from a cost point of view, to use an analogue controller, instead of a digital controller. Simple analogue controllers, although outdated, do not require digital signal processing equipment, analogue-to-digital (A/D) and digital-to-analogue (D/A) converters.

The limitation of an analogue controller, however, is that the required feedforward and feedback characteristics may be difficult to obtain. If the controller does not perform as originally anticipated during the design, certain parameters may have to be changed. Whereas this is easily accomplished with digital controllers, it may be extremely difficult with analogue controllers. The reason is that analogue equipment such as resistors, capacitors and inductors, may not have the required range of adjustment. While rheostats have been around for a long time, adjustable capacitances are currently only available in capacitors with relatively small capacitances.

X.2 Stability

Stability, together with accuracy, is probably the most important requirement of a control system. Stability of a linear time invariant (LTI) system is determined by the real parts of its eigenvalues: An LTI system is unconditionally stable if the real parts of all its eigenvalues are negative.

This statement must, however, be qualified for a *linearized* system. Linear systems analysis applied to a nonlinear system, may only be accurate over a limited working range. This phenomenon was discussed in detail in section 2.7. The equivalent linear range will depend on the nature of the nonlinearity. A weakly nonlinear system may have a fairly large linear range of operation and may therefore be accurately represented by a linear model. Linear stability (gain and phase) margins of the linearized system will closely approximate those of the nonlinear system. This system type can be relatively easily stabilized by a linear controller.

On the other hand, a strongly nonlinear system may have relatively small linear ranges of operation, also known as regions of local stability. If the system is operated inside any of these regions, it will be stable. If operated outside these regions, it will be unstable. Such a system is globally unstable. Local and global stability are discussed in more detail by Slotine & Li [1991].

Nonlinear systems may be accurately controlled and stabilized by nonlinear controllers. The design and implementation of nonlinear controllers is however beyond the scope of this study.



Alternatively, a nonlinear plant may be controlled by a linear controller, with sufficient robustness to compensate for changes in characteristics (e.g. DC gains, natural frequencies and damping factors), without sacrificing other important performance parameters (e.g. accuracy). Techniques of improving system robustness will be discussed in section X.6.

X.3 Sensitivity

Sensitivity of a control system depends on two factors, i.e. parameter sensitivity and sensor sensitivity. Parameter sensitivity can be described as the "inverse phenomenon" of robustness, i.e. a more robust system is less sensitive to a change in the value of a specific parameter. Robustness will be discussed in short in section X.6. More detailed discussions on the subject are provided by D'Azzo & Houpis [1986], Maciejowski [1989], Skogestad & Postlethwaite [1997], Slotine & Li [1991] and Tsui [1996].

Sensor sensitivity mainly depends on sensor resolution and sensor calibration factor. Sensor resolution is the smallest increment in a measured signal to which a sensor will respond [Cooper & Helfrick, 1985]. The smaller the increment, the better the resolution. Sensor resolution is an important parameter in vibration isolation applications. The objective of vibration isolation is to obtain output signals with low magnitudes, in comparison with the disturbances. In order to measure and control accurately, sensors must be able to respond to small changes in outputs.

Calibration factor is the ratio of the sensor output signal to the measured variable [Cooper & Helfrick, 1985]. The calibration factor of an accelerometer, for instance, is the ratio of the output voltage to the measured acceleration, and is normally expressed in mV/g. The most suitable calibration factor is determined by the full-scale range of the data acquisition equipment. It is preferable that this range correspond to the output range of the sensor, as it will eliminate the need for unduly high control signal gains. (In section 4.3.3, for example, it was mentioned that the coil voltage exceeded the signal analyzer range, and had to be scaled by means of a voltage divider, before it could be measured).

X.4 Reaction speed

Reaction speed is determined by two main factors, i.e. controller frequency bandwidth and throughput lag. The required frequency bandwidth can be obtained during the design phase, by calculation of the controller feedforward and feedback parameters. Optimal and poleplacement design techniques are often used for this purpose. The cutoff frequency of the controlled system can be verified by comparing it with the required cutoff-frequency.

Throughput lag is the time it takes to execute the A/D conversion of sensor signals, apply digital control (e.g. by feedforward and feedback compensation, digital filtering, frequency shaping, addition, subtraction, gaining and clipping of signals) and convert the signals back to analogue form. The lag is often also expressed in terms of data buffer length or number of samples.



The lag depends, inter alia, on the control hardware configuration. If the controller runs on its own processor (DSP), the DSP does the A/D and D/A conversion, as well as the control calculations. The processor must be specifically programmed for this purpose. Programming is normally done externally, in languages like Assembler, C, Pascal, Quick Basic, or Visual Basic, whereafter the program is compiled and loaded onto the processor. Some processors may be programmed and compiled from graphic environments like Simulink (see section 4.4). Processing speed mainly depends on the DSP clock speed. Compiling of the program necessarily improves throughput speed.

X.5 Control effort

Control effort is the input required to drive the plant to its desirable output. In the case of the magnetostrictive LOS stabilization system, coil voltage is the input, and the optical instrument angular acceleration is the output. The objective is to minimize the output. While this is an important goal, it must be achieved with as little control effort as possible. Both the output and control effort must therefore be minimized. This process is known as optimization. A type of controller which simultaneously minimizes output and control input, is an optimal regulator.

The optimal feedforward and feedback control parameters are determined mathematically, by application of methods such as variational optimization and linear quadratic optimal regulator (LQR) design. In order to obtain a criterion which can be used to evaluate system optimality, and which can be expressed in terms of a single numerical value, an objective function is defined. This function, also known as the cost function, must be minimized to obtain the optimal control parameters.

The linear quadratic (LQ) optimal cost function for the regulator problem under discussion, is:

$$J = \int_{-0}^{-T} \left[y^{2}(t) + u^{2}(t) \right] dt$$
 (X.5.1)

where *J* is the cost function.

In equation X.5.1, the output and control are equally weighed. The output and control can be separately weighed, as follows:

$$J = \int_{0}^{2\pi} \left[Q_y y^2(t) + Ru^2(t) \right] dt$$
 (X.5.2)

where Q_{ν} and R are the output and control weights, respectively.

The cost function is more conveniently expressed in matrix form, where y and u are vectors:

$$J = y^T Q_{\nu} y + u^T R u \tag{X.5.3}$$



The cost function can be extended to include a term which contains the product of the input and output:

$$J = \int_{-0}^{-T} [Q_y y^2(t) + Ru^2(t) + Ny(t)u(t)] dt$$
 (X.5.4)

where N is the weight of the product yu. Inclusion of this term may result in a cost function which cannot strictly be termed "linear quadratic", as yu will only be positive for all t if u and y are in phase. This is however not the case with the magnetostrictive active vibration isolation system (see figure 4.5.2.3). For this reason, the yu-term will henceforth not be included in the cost function.

Optimization can either be unconstrained or constrained. Unconstrained optimization means that neither y nor u is bounded. An example of an unconstrained optimization equation is given by equation X.5.2. The equation does not supply sufficient information to minimize J, since there is no relationship between u and y. The required relationship is supplied by the state and output equations, which are the constraint equations for the LQR state regulation problem under discussion:

$$\dot{x} = Ax + B_u u + B_d d \tag{X.5.5a}$$

$$y = Cx (X.5.5b)$$

Equations X.5.5 may not necessarily be the only constraint equations. Constraints can be added to suit the problem. Although the input is accounted for in the cost function, it may exceed the available supply. To avoid this, an additional constraint, i.e. available source voltage (+28 V for the magnetostrictive active isolation system) can be added. Another constraint, which can be added in the interest of robustness, is minimum damping.

A relationship between the states and control, which satisfies equations X.5.3 and X.5.5 above, is required. In a state-controlled linear system, the control is a linear combination of the states:

$$u = -Kx (X.5.6)$$

where K is the optimal state feedback gain, or "state feedback control law". A necessary condition for equation X.5.6 is that all the states be known from measured outputs. This property, which is known as observability, will be discussed later.

The elements of K can be determined in a number of ways, e.g. by variational optimization methods and by solving the algebraic Riccati equation (ARE). The Riccati equation, in its steady state form, is:

$$Q_{x} + A^{T}S_{xx} + S_{xx}A - S_{xx}B_{y}R^{-1}B_{y}^{T}S_{xx} = 0$$
 (X.5.7)

where S_{xx} is the steady-state Riccati matrix and Q_x is the state weight, related to Q_y by:

$$Q_x = C^T Q_y C (X.5.8)$$



Equation X.5.7 must be solved to obtain the steady-state Riccati matrix. A solution method often used, is Hamiltonian eigenvector decomposition, which is described in detail by Meirovitch [1990].

The optimal feedback gain matrix K is given in terms of the Riccati matrix as:

$$K = R^{-1} B_u^T S_{ss} (X.5.9)$$

The closed-loop state and output equations are obtained by substitution of equation X.5.6 into equation X.5.5:

$$\dot{x} = (A - B_u K)x + B_d d$$

$$y = Cx$$
(X.5.10a)
(X.5.10b)

It can be seen from equation X.5.10a that the only input to the closed-loop regulator, is the disturbance d.

The closed-loop eigenvalues are the values of s for which:

$$|sI - (A - B_u K)| = 0$$
 (X.5.11)

The block diagram of the plant and state feedback regulator, with state and control equations given by equations X.5.5 and X.5.6 respectively, is shown in figure X.5.1.

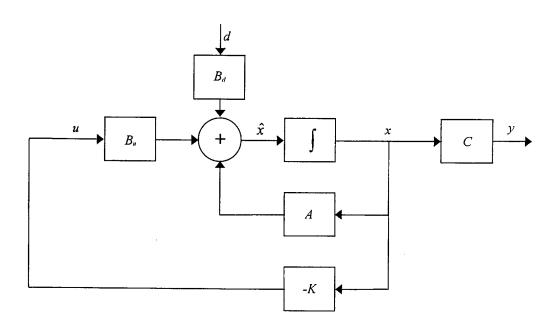


Figure X.5.1: State feedback regulator block diagram

More thorough discussions on optimal control can be found in Anderson & Moore [1989], Brogan [1985], D'Azzo & Houpis [1986], Kirk [1970], Kuo [1982] and Meirovitch [1990].



X.6 Robustness

The ability of a controller to meet its performance criteria in the presence of plant, input, disturbance and measurement noise uncertainties, is known as robustness. An example of a plant uncertainty is a nonlinearity, which may cause the plant to display different characteristics at different excitation levels. Depending on the nature of the nonlinearity, a robust controller will partly or fully compensate for the changes in characteristics.

Examples of input and disturbance uncertainties are DC offsets, which may be attributed to unmodelled sensor characteristics. A typical example is the DC characteristic of an accelerometer. Theoretically, an accelerometer has a zero DC output, but practically, the output voltage may contain a DC component. Resistive accelerometers, in particular, display this characteristic. Noise uncertainties may include input and output sensor noise, noise generated by measurement equipment, such as amplifiers, and electromagnetic noise in unscreened cables.

Linear quadratic (LQ) optimal state feedback controllers are seldom robust [Kuo, 1982]. The reason is that these controllers often operate on the principle of pole-zero cancellation, i.e. the frequency and damping factor of a TF pole coincide with those of a TF zero. When the plant is excited at the design drive level, the zero will exactly cancel the pole. If, however, the drive level is slightly altered, phenomena such as nonlinear plant behaviour may cause the frequency and damping factor of the pole to shift, to such an extent that it may no longer coincide with those of the zero. The pole may not be cancelled at all, with the result that transmissibility of the closed-loop system may be worse than that of a system without vibration isolation.

A number of solutions exist to improve robustness, three of which are as follows: Firstly, pole damping may be increased, using pole-placement techniques. If the pole shifts within a narrow frequency band, it will still be partly cancelled by the zero. Although performance, in terms of attenuation, will not match that of the LQ optimal controller, robustness will improve.

Secondly, the optimality constraints (see equations X.5.5) may be expanded to include minimum closed-loop damping factors. Although this measure may complicate the solution of the Riccati equation, the optimal feedback gain may be obtained by application of alternative optimization methods, such as variational methods.

Thirdly, a solution suggested by Kuo [1982], is to place an output integrator, with gain H_I , in parallel with the state feedback matrix, as shown in figure X.6.1. The effect of this method on improving the robustness of an absolute motion controller was discussed in section 6.3.8.5.



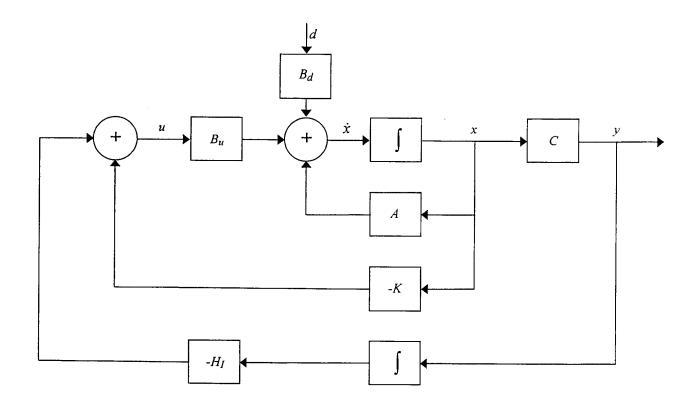


Figure X.6.1: State plus output integral feedback regulator block diagram

The control law is of the following form:

$$u = -Kx - H_1 \int y dt \tag{X.6.1}$$

Differentiation of equation X.6.1 with respect to time, gives:

$$\dot{u} = -K\dot{x} - H_t y \tag{X.6.2}$$

Substitution of equations X.5.5 into equation X.6.2 and simplification of the resulting equation gives:

$$\dot{u} = -(KA + H_1C)x - KB_u u - KB_d d \qquad (X.6.3)$$

Equation X.6.3 is the controller state equation. The term $KA + H_1C$ couples the plant and controller state equations.



Combination of the plant and controller state and output equations X.5.5 and X.6.3 gives the closed-loop system state-space equations as follows:

$$\dot{x}_{cl} = A_{cl} x_{cl} + B_{cl} d ag{X.6.4a}$$

$$y_{cl} = C_{cl} x_{cl} \tag{X.6.4b}$$

where x_{cl} is the closed-loop state vector, given by:

$$x_{cl} = \begin{cases} x \\ u \end{cases}$$
 (X.6.5)

 A_{cl} , B_{cl} and C_{cl} are the closed-loop coefficient, driving and output matrices, respectively given by:

$$A_{cl} = \begin{bmatrix} A & B_{ll} \\ -(KA + H_{l}C) & -KB_{ll} \end{bmatrix}$$
 (X.6.6a)

$$B_{cl} = \begin{bmatrix} B_d \\ -KB_d \end{bmatrix}$$
 (X.6.6b)

$$C_{cl} = \begin{bmatrix} C & 0 \end{bmatrix}$$
 (X.6.6c)

The eigenvalues of the closed-loop system are the values of s for which:

$$\left| sI - A_{cI} \right| = 0 \tag{X.6.7}$$

The integrator raises the DC and low-frequency content of the control signal, thereby making the system more sensitive to feedback at low frequencies and less sensitive at high frequencies. A typical application is DC disturbance rejection in electric motor control, as explained by Kuo [1982].

The method can also be applied to improve low-frequency behaviour of the magnetostrictive LOS stabilization system, which is the topic of this study. All but one of the plant poles are well in excess of the disturbance band (see table 5.3.2), therefore an integrator will make the system less sensitive to feedback at most of the plant natural frequencies.

However, a disadvantage of this method, is that an integrator in the feedback loop may drive the control to infinity, although the output may be bounded. A cure to the problem is to highpass filter the control signal. If the filter cutoff frequency is sufficiently low, the DC component will be eliminated, but the desirable low-frequency characteristics will be retained. More sophisticated robust control UNIVERSITY OF PRETORIA: loop transfer recovery (LTR), as discussed by Maciejowski [1989], Skogestad & Postlethwaite [1997] and Tsui [1996].

X.7 Observability

State feedback is a powerful control method, especially for high-order systems. A condition for state control is that the states be observable from output measurements. If all the states can be reconstructed from the output, a system is said to be completely observable, or simply, observable.

Mathematically, the n-th order LTI system, whose state and output equations are given by

$$\dot{x} = Ax + Bu \tag{X.7.1a}$$

$$y = Cx (X.7.1b)$$

is observable if and only if the matrix

$$\left[C^{T} : A^{T}C^{T} : \left(A^{T}\right)^{2}C^{T} : \cdots : \left(A^{T}\right)^{n-1}C^{T}\right] \tag{X.7.2}$$

has rank n [Kirk, 1970].

Observability makes it possible to observe the states by means of a state observer, or estimator. State observer dynamics can be described by means of observer state and output equations. The inputs to the observer depend on the controller configuration. For an absolute motion controller, i.e. for pure output feedback (see section 6.3.8.2), the observer input is the plant output. For a relative motion controller and an absolute plus base motion controller (see sections 6.3.8.1 and 6.3.8.4), the observer inputs are the plant output and disturbance.

The observer state-space equations, for an absolute plus base motion controller, are:

$$\dot{\hat{x}} = A\hat{x} + B_u u + B_d d + L(y - \hat{y})$$
 (X.7.3a)

$$\hat{y} = C\hat{x} \tag{X.7.3b}$$

where \hat{x} is the observed state vector, which differs from the true plant state vector, A is the plant coefficient matrix, B_u is the plant driving matrix for the controllable input u, B_d is the plant driving matrix for the disturbance input, y is the plant output, \hat{y} is the observed output and L is the observer driving matrix. The observer state equation for a pure output feedback controller is given by equation X.7.3a, with a zero disturbance driving matrix B_d .

For full-state feedback control, the control is a linear combination of the observed states:

$$u = -K\hat{x} \tag{X.7.4}$$

where K is the state feedback gain matrix.



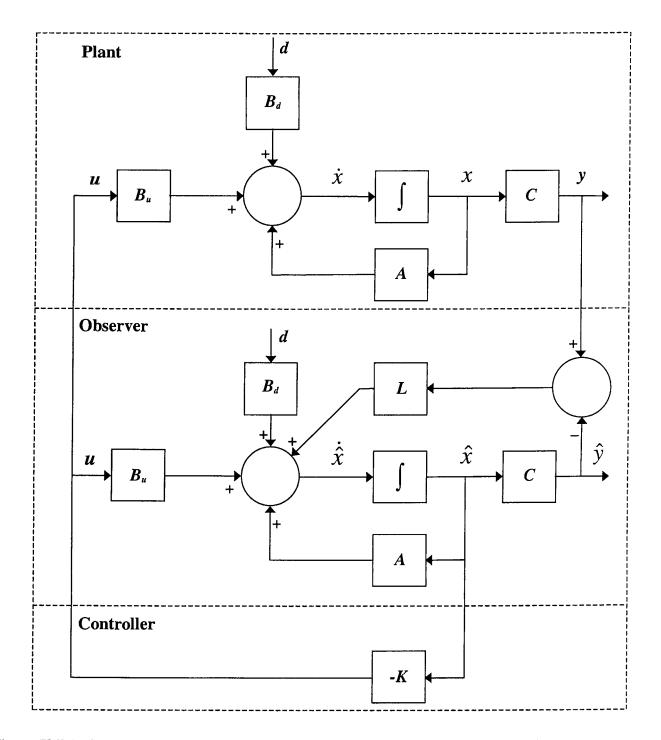


Figure X.7.1: Block diagram of plant, state observer and state feedback controller

To conclude, if the system is non-observable, the observable states may still be estimated, by means of a reduced-order observer. Reduced-order observers are described in more detail by Anderson & Moore [1989], Franklin et al [1990] and Kwakernaak & Sivan [1972].



X.8 Controllability

A system is said to be completely controllable if every state variable can be controlled to reach a certain objective in finite time by some unconstrained control u [Kuo, 1982]. The physical interpretation of controllability is that, if any one of the state variables is independent of the control u, driving this particular state to a desired state in finite time by means of u will be impossible. This state is said to be uncontrollable, which means that the system is not completely controllable, or simply uncontrollable [Kuo, 1982].

Controllability can be expressed as follows in terms of the plant state matrices A and B. Mathematically, the n-th order LTI system, whose state and output equations are given by

$$\dot{x} = Ax + Bu$$
 (X.8.1a)
$$y = Cx$$
 (X.8.1b)

is controllable if and only if the matrix

$$\begin{bmatrix} B & \vdots & AB & \vdots & A^2B & \vdots & \cdots & \vdots & A^{n-1}B \end{bmatrix}$$
 (X.8.2)

has rank n [Kirk, 1970].

Alternative practical methods of testing controllability of a system are discussed by Kuo [1982].