



Two main design approaches: a. discretize the analog controller, b. discretize the process and do the design totally in discrete time a. r(t)u(t) y(t)A/D Controller D/A Process y(t) $r(t_k)$ $u(t_k)$ $y(t_k)$ D/A Process A/D b. Controller $y(t_k)$











The matrix exponential

The solution to the homogenous set of differential equations

 $\dot{x}(t) = Ax(t), \quad x(0) = x_0$

is then $x(t) = e^{At}x(0) = e^{At}x_0$

The term e^{At} is called the *state transition matrix* (in this context).

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The matrix exponential

Solution by using the Laplace transformation

 $\dot{x}(t) = Ax(t), \quad x(0) = x_0 \qquad sX(s) - x_0 = AX(s)$

It follows
$$(sI - A)X(s) = x_0$$

 $(sI - A)^{-1}(sI - A)X(s) = (sI - A)^{-1}x_0$
 $X(s) = (sI - A)^{-1}x_0$

The solution is obtained by the inverse transformation

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Solution for the input-output representation $\dot{x}(t) = Ax(t) + Bu(t), \quad x(0) = x_0$ y(t) = Cx(t)is $x(t) = e^{At}x_0 + \int_0^t e^{A(t-\tau)}Bu(\tau) d\tau$ $y(t) = Ce^{At}x_0 + \int_0^t Ce^{A(t-\tau)}Bu(\tau) d\tau$ Final Action of Electrical Engineering













Example. Discretization by using the series expansion State-space representation of the double integrator $\begin{cases} \dot{\mathbf{x}}(t) = \begin{bmatrix} 0 & 1 \\ 0 & 0 \end{bmatrix} \mathbf{x}(t) + \begin{bmatrix} 0 \\ 1 \end{bmatrix} u(t) \qquad \Phi = \mathbf{e}^{\mathbf{A}h} = \mathbf{I} + \mathbf{A}h + \frac{\mathbf{A}^2h^2}{2} + \cdots \\ y(t) = \begin{bmatrix} 1 & 0 \end{bmatrix} \mathbf{x}(t) \\ \Phi = \begin{bmatrix} 1 & 0 \\ 0 & 1 \end{bmatrix} + \begin{bmatrix} 0 & h \\ 0 & 0 \end{bmatrix} + \begin{bmatrix} 0 & 0 \\ 0 & 0 \end{bmatrix} + \cdots = \begin{bmatrix} 1 & h \\ 0 & 1 \end{bmatrix} \\ \Gamma = \int_{0}^{h} \mathbf{e}^{\mathbf{A}s} ds \mathbf{B} = \int_{0}^{h} \begin{bmatrix} 1 & s \\ 0 & 1 \end{bmatrix} ds \begin{bmatrix} 0 \\ 1 \end{bmatrix} = \int_{0}^{h} \begin{bmatrix} s \\ 1 \end{bmatrix} ds = \int_{0}^{h} \begin{bmatrix} \frac{1}{2} s^2 \\ s \end{bmatrix} = \begin{bmatrix} \frac{1}{2} h^2 \\ h \end{bmatrix}$















So, if we have a discrete pulse sequence only, how can we know what frequency the real continuous time signal has?

Moreover: the continous time model corresponding to a discrete model is not always unique.

(Take the harmonic oscillator $G(s) = \omega^2/(s^2 + \omega^2)$,

do a realization and discretize by using $\omega = \alpha + n2\pi/h$, n = 0, 1, 2, ...

These examples show the **alias effect** for the first time. It is very dangerous. We will return to this later in the course.



Linear time-invariant systems: The pulse transfer function, weighting function, pulse response and convolution sum u(k) y(k) H(z) U(z) Y(z) Y(z) = H(z)U(z)Input-output: Z-domain $y(k) = \sum_{i=0}^{k} h(k-i)u(i) = \sum_{i=0}^{k} h(i)u(k-i)$ Input-output: time domain, convolution sum The z-transform of the weighting function h(k) is the

pulse transfer function H(z).

The (im)pulse response coincides with the weighting function from the time that the pulse enters.

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Input-Output-models, I/O-models, pulse response		
Discrete system:	$\begin{cases} \mathbf{x}(k+1) = \Phi \mathbf{x}(k) + \Gamma \mathbf{u}(k), \text{ initial value } x(k_0) \\ \mathbf{y}(k) = \mathbf{C} \mathbf{x}(k) + \mathbf{D} \mathbf{u}(k) \end{cases}$	
Solution by direct recursive calculus starting from a given initial state $\mathbf{x}(k_0)$ up to any time k .	$\mathbf{x}(k_{0}+1) = \Phi \mathbf{x}(k_{0}) + \Gamma \mathbf{u}(k_{0})$ $\mathbf{x}(k_{0}+2) = \Phi \mathbf{x}(k_{0}+1) + \Gamma \mathbf{u}(k_{0}+1)$ $= \Phi^{2} \mathbf{x}(k_{0}) + \Phi \Gamma \mathbf{u}(k_{0}) + \Gamma \mathbf{u}(k_{0}+1)$ \vdots $\mathbf{x}(k) = \Phi^{k-k_{0}} \mathbf{x}(k_{0}) + \Phi^{k-k_{0}-1} \Gamma \mathbf{u}(k_{0}) + \dots + \Gamma \mathbf{u}(k-1)$ $= \Phi^{k-k_{0}} \mathbf{x}(k_{0}) + \sum_{j=k_{0}}^{k-1} \Phi^{k-j-1} \Gamma \mathbf{u}(j)$	
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Input-Output-models, I/O-models, pulse response
But for a state representation the pulse response is
calculated as

$$\mathbf{x}(k) = \Phi^{k-k_0}\mathbf{x}(k_0) + \sum_{j=k_0}^{k-1} \Phi^{k-j-1}\Gamma u(j) \qquad y(k) = \mathbf{C}\mathbf{x}(k) + Du(k)$$

$$y(k) = \mathbf{C}\Phi^{k-k_0}\mathbf{x}(k_0) + \sum_{j=k_0}^{k-1} \mathbf{C}\Phi^{k-j-1}\Gamma u(j) + Du(k)$$

$$h(k) = \begin{cases} 0 \qquad k < 0 \\ D \qquad k = 0 \\ \mathbf{C}\Phi^{k-1}\Gamma \qquad k \ge 1 \end{cases} \text{ (assume } k_0 = 0)$$

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Shift-operators

Corresponding to the differential operator p used in continuous systems a (forward) shift operator q is defined for discrete systems

$$q \cdot f(k) = f(k+1)$$

$$q^{-1} \cdot f(k) = f(k-1)$$

By using the shift operator input-output relationships (difference equations) can easily be described

$$y(k+n_{a}) + a_{1}y(k+n_{a}-1) + \dots + a_{n_{a}}y(k) = b_{0}u(k+n_{b}) + \dots + b_{n_{b}}u(k)$$
$$(q^{n_{a}} + a_{1}q^{n_{a}-1} + \dots + a_{n_{a}})y(k) = (b_{0}q^{n_{b}} + b_{1}q^{n_{b}-1} + \dots + b_{n_{b}})u(k)$$

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Shift-operators

Difference equations can be described by polynomials

$$A(z) = z^{n_a} + a_1 z^{n_a - 1} + \dots + a_{n_a}$$

$$B(z) = b_0 z^{n_b} + b_1 z^{n_b - 1} + \dots + b_{n_b}$$

$$A(q)y(k) = B(q)u(k)$$

A backward shift operator q^{-1} can also be used

$$y(k) + a_1 y(k-1) + \dots + a_{n_a} y(k-n_a) = b_0 u(k-d) + \dots + b_{n_b} u(k-d-n_b)$$

 $d = n_a - n_b$

$$(1+a_1q^{-1}+\dots+a_{n_a}q^{-n_a})y(k) = (b_0+b_1q^{-1}+\dots+b_{n_b}q^{-n_b})u(k-d)$$

Shift-operators

 $A^{*}(z) = 1 + a_{1}z + \dots + a_{n_{a}}z^{n_{a}} \qquad A^{*}(q^{-1})y(k) = B^{*}(q^{-1})u(k-d)$ $B^{*}(z) = b_{0} + b_{1}z + \dots + b_{n_{b}}z^{n_{b}} \qquad A^{*}(q^{-1})y(k) = q^{-d} \cdot B^{*}(q^{-1})u(k)$

Resiprocal polynomials A^*, B^*

Polynomial representations are used, because in some control design methods they are more natural to use than state-space representations.

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Pulse-transfer operator, H(q)From polynomial representations A(q)y(k) = B(q)u(k) $H(q) = \frac{B(q)}{A(q)}$ or $A^*(q^{-1})y(k) = q^{-d} \cdot B^*(q^{-1})u(k)$ $H^*(q^{-1}) = \frac{q^{-d}B^*(q^{-1})}{A^*(q^{-1})}$ Representation by *reciprocal polynomials* $H^*(q^{-1}) = H(q)$

Pulse-transfer operator, H(q)

Consider a simple example of a pulse transfer function.

$$\begin{cases} x(k+1) = 0.5x(k) + 0.5u(k) \\ y(k) = 2x(k) \end{cases}$$

We obtain

$$H(q) = C(q - \Phi)^{-1}\Gamma + D = \frac{2 \cdot 0.5}{q - 0.5} = \frac{1}{q - 0.5}$$

Starting from IO-difference equation the calculation is equally easy

Pulse-transfer operator, $H(q)$		
$\begin{cases} x(k) = \frac{1}{2}y(k) \\ x(k+1) = 0.5x(k) + 0.5u(k) \end{cases} \Rightarrow$	0.5y(k+1) = 0.25y(k) + 0.5u(k)	
Form A- and B-polynomials		
(0.5q - 0.25)y(k) = 0.5u(k)	(q-0.5)y(k) = u(k)	
A(q)y(k) = B(q)u(k)	A(q) = q - 0.5 $B(q) = 1$	
The pulse transfer operator is	$H(q) = \frac{B(q)}{A(q)} = \frac{1}{q - 0.5}$	
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Pulse-transfer operator, H(q)All methods lead to the same pulse transfer operator $\left(\frac{q^{-1}}{1-0.5q^{-1}} = \frac{1}{q-0.5}\right)$ $H^*(q^{-1}) = H(q)$



Poles and zeros

Some basics from matrix calculus. For the pulse transfer operator we can write

$$H(q) = C(qI - \Phi)^{-1}\Gamma + D = \frac{C \cdot \operatorname{adj}(qI - \Phi)\Gamma + D \cdot \det(qI - \Phi)}{\det(qI - \Phi)}$$

where 'adj' means the adjungate matrix. For the matrix Φ the eigenvalues λ_i and eigenvectors e_i are defined as follows

 $\Phi e_i = \lambda_i e_i \Longrightarrow (\lambda_i I - \Phi) e_i = 0$

There exist non-zero eigenvectors when $det(\lambda_I I - \Phi) = 0$

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Poles and zeros The poles are the roots of the characteristic polynomial. They Belong to the set of eigenvalues of the system matrix Φ . Consider the polynomial $f(\lambda) = \alpha_0 \lambda^m + \alpha_1 \lambda^{m-1} + \dots + \alpha_m$ The corresponding function for the square matrix *A* is defined as $f(A) = \alpha_0 A^m + \alpha_1 A^{m-1} + \dots + \alpha_m I$ Result: Let the eigenvalues of *A* be λ_i and the corresponding eigenvectors e_i . It holds **Alto University** School of Electrical Engineering

Poles and zeros

$$f(A)e_i = f(\lambda_i)e_i$$

meaning that $f(\lambda_i)$ is the eigenvalue of f(A) and the Corresponding eigenvector is e_i .

The proof is straightforward by writing the above formula and using repeatedly the definitions of eigenvalue and eigenvector \Rightarrow Exercise.



Poles and zeros

The poles of a discrete system are the zeros of the denominator of H(z). The zeros are the zeros of the numerator. The poles are also eigenvalues of the the system matrix Φ .

From the location of poles in the complex plane stability, oscillations and speed of the system can be deduced.

The poles of a continuous *n*:th order system are mapped to the discrete system poles according to:

 $\begin{cases} \dot{\mathbf{x}}(t) = \mathbf{A}\mathbf{x}(t) + \mathbf{B}\mathbf{u}(t) & \text{poles: } \lambda_i(\mathbf{A}), \quad i = 1, \dots, n \\ \mathbf{y}(t) = \mathbf{C}\mathbf{x}(t) \end{cases}$

Poles and zeros

Discrete system
$$\begin{cases} \mathbf{x}(kh+h) = \mathbf{\Phi} \, \mathbf{x}(kh) + \Gamma \, \mathbf{u}(kh) \\ \mathbf{y}(kh) = \mathbf{C} \mathbf{x}(kh) \end{cases}$$
poles: $\lambda_i(\mathbf{\Phi}), \quad i = 1, ..., n$

$$\mathbf{\Phi} = \mathbf{e}^{\mathbf{A}h} \quad \Longrightarrow \quad \lambda_i(\mathbf{\Phi}) = \mathbf{e}^{\lambda_i(\mathbf{A})h}$$

A simple relationship for the mapping of zeros does **not** hold. Even the number of zeros does not necessarily remain invariant. The mapping of zeros is a complicated issue.

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Unstable inverse, non-minimum phase systems

•A continuous time system is *non-minimum phase*, if has zeros on the right half plane (RHP) or if it contains a delay.

•A discrete system has an unstable inverse, if it has zeros outside the unit circle.

•Zeros are not mapped in a similar way as poles, so a minimum phase continuous system may have a discrete counterpart with an unstable inverse and a non-minimum phase continuous system may have a discrete counterpart with a stable inverse.





Selection of the sampling rate

- The proper choice of the sampling interval is very important. Too low sampling frequency may lose so much information that the control performance deteriorates and the system dynamics is lost.
- Too high sampling rate increases the burden of the processor; also, it may lead to discrete representation with bad numerical properties.
- For oscillating systems the sampling interval is often tied to the frequency of the dominating oscillation. For damped systems the sampling interval is usually chosen to be in relation to the time constant.



Selection of the sampling rate

N_r means the amount of samples during the *rise time*.

$$N_r = \frac{T_r}{h}$$

For a self-oscillating system (2nd order, damping ratio ς and natural frequency ω_0) the rise time is:

$$T_r = \omega_0^{-1} e^{\frac{\varphi}{\tan \varphi}}, \quad \zeta = \cos \varphi$$

A usual sampling rate is $N_r = \frac{T_r}{h} \approx 4 \cdots 10$ which leads to a

"Rule of Thumb" $\omega_0 h = 0.2...0.6$

