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Z-Transforms (Take 2) Linear Time Invariant Discrete Systems

ELEC 3004: Systems: Signals & Controls
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Lecture 8

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<http://robotics.itee.uq.edu.au/~elec3004/>

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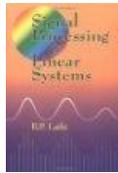


Lecture Schedule:

Week	Date	Lecture Title
1	27-Feb	Introduction
	1-Mar	Systems Overview
2	6-Mar	Systems as Maps & Signals as Vectors
	8-Mar	Systems: Linear Differential Systems
3	13-Mar	Sampling Theory & Data Acquisition
	15-Mar	Aliasing & Antialiasing
4	20-Mar	Discrete Time Analysis & Z-Transform
	22-Mar	Second Order LTID (& Convolution Review)
5	27-Mar	Frequency Response
	29-Mar	Filter Analysis
6	3-Apr	Digital Filters (IIR) & Filter Analysis
	5-Apr	Digital Filter (FIR)
7	10-Apr	Digital Windows
	12-Apr	FFT
8	17-Apr	Active Filters & Estimation & Holiday
	19-Apr	Holiday
24-Apr		
26-Apr		
9	1-May	Introduction to Feedback Control
	3-May	Servoregulation/PID
10	8-May	PID & State-Space
	10-May	State-Space Control
11	15-May	Digital Control Design
	17-May	Stability
12	22-May	State Space Control System Design
	24-May	Shaping the Dynamic Response
13	29-May	System Identification & Information Theory
	31-May	Summary and Course Review



Follow Along Reading:



B. P. Lathi
*Signal processing
and linear systems*
1998
[TK5102.9.L38 1998](#)

Today

- **Chapter 11 (Discrete-Time System Analysis Using the z -Transform)**
 - § 11.1 The \mathcal{Z} -Transform
 - § 11.2 Some Properties of the \mathcal{Z} -Transform

- **Chapter 9 (Time-Domain Analysis of Discrete-Time Systems)**
 - § 9.4 System Response to External Input
 - § 9.6 System Stability

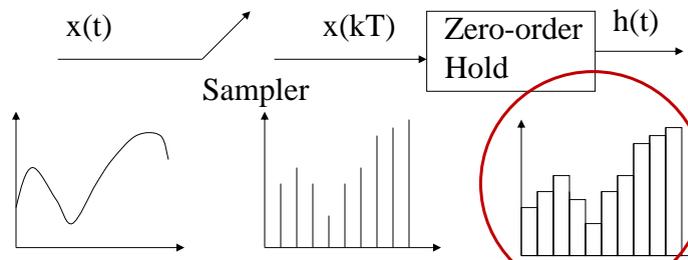
Next Time



z Transforms (Digital Systems Made eZ)

Extended Explanation 😊

Back to the Zero-order Hold (ZOH)



- Assume that the signal $x(t)$ is zero for $t < 0$, then the output $h(t)$ is related to $x(t)$ as follows:

$$\begin{aligned}
 h(t) &= x(0)[1(t) - 1(t - T)] + x(T)[1(t - T) - 1(t - 2T)] + \dots \\
 &= \sum_{k=0}^{\infty} x(kT)[1(t - kT) - 1(t - (k + 1)T)]
 \end{aligned}$$

→ The “hold” adds a delay. The delay leads to difference equations



Coping with Complexity

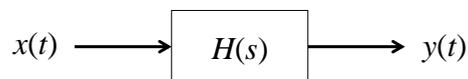
Transfer functions help control complexity

- Recall the Laplace transform:

$$\mathcal{L}\{f(t)\} = \int_0^{\infty} f(t)e^{-st} dt = F(s)$$

where

$$\mathcal{L}\{\dot{f}(t)\} = sF(s)$$



- Is there a something similar for sampled systems?



The z-Transform

- It is defined by:

$$z = re^{j\omega}$$

- Or in the Laplace domain:

$$z = e^{sT}$$

- Thus: $Y(z) = \sum_{k=-\infty}^{\infty} h[k]z^{-k}$ $y[n] \xleftrightarrow{\mathcal{Z}} Y(z)$

- That is \rightarrow it is a discrete version of the Laplace:

$$f(kT) = e^{-akT} \Rightarrow \mathcal{Z}\{f(k)\} = \frac{z}{z - e^{-aT}}$$



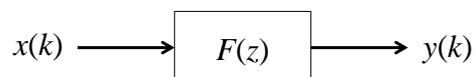
The z-transform

- The discrete equivalent is the z-Transform[†]:

$$\mathcal{Z}\{f(k)\} = \sum_{k=0}^{\infty} f(k)z^{-k} = F(z)$$

and

$$\mathcal{Z}\{f(k-1)\} = z^{-1}F(z)$$



Convenient!

[†]This is not an approximation, but approximations are easier to derive



The z-Transform [2]

- Thus:

$$Y(z) = \sum_{k=-\infty}^{\infty} h[k]z^{-k} \quad y[n] \xleftrightarrow{\mathcal{Z}} Y(z)$$

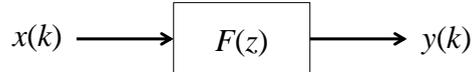
- z-Transform is analogous to other transforms:

$$\mathcal{Z}\{f(k)\} = \sum_{k=0}^{\infty} f(k)z^{-k} = F(z)$$

and

$$\mathcal{Z}\{f(k-1)\} = z^{-1}F(z)$$

- ∴ Giving:

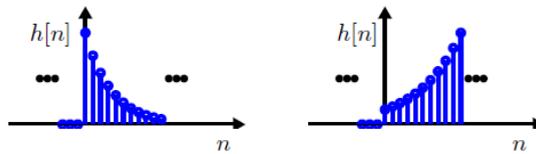


z-Transforms for Difference Equations

- First-order linear constant coefficient difference equation:

First-order linear constant coefficient difference equation:

$$y[n] = ay[n-1] + bu[n]$$



$$h[n] = \begin{cases} ba^n & n \geq 0, \\ 0 & \text{otherwise.} \end{cases}$$

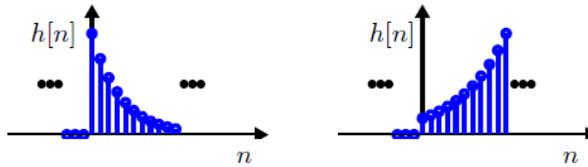
$$H(z) = \sum_{k=0}^{\infty} ba^k z^{-k} = b \sum_{k=0}^{\infty} \left(\frac{a}{z}\right)^k = \frac{b}{1 - az^{-1}}, \quad \text{when } |z| > |a|.$$



z-Transforms for Difference Equations

First-order linear constant coefficient difference equation:

$$y[n] = ay[n - 1] + bu[n]$$



$$\begin{aligned} y[n] - ay[n - 1] &= bu[n] \\ \Downarrow \\ Y(z) - az^{-1}Y(z) &= bU(z) \end{aligned}$$

$$H(z) = \frac{Y(z)}{U(z)} = \frac{b}{1 - az^{-1}}, \text{ when does it converge?}$$



The z-transform

- In practice, you'll use look-up tables or computer tools (ie. Matlab) to find the z-transform of your functions

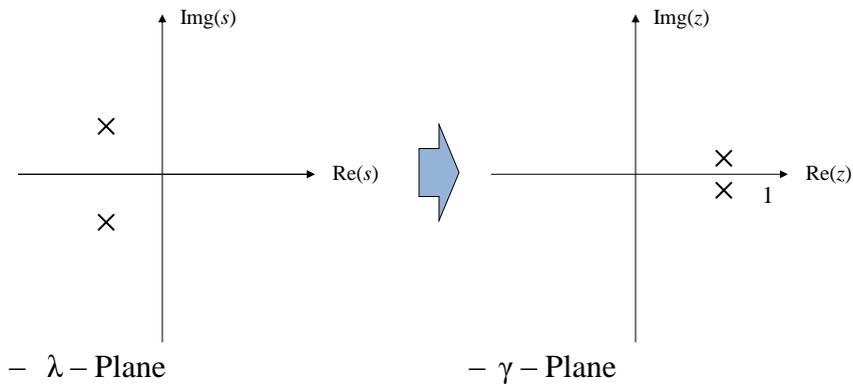
$F(s)$	$F(kt)$	$F(z)$
$\frac{1}{s}$	1	$\frac{z}{z - 1}$
$\frac{1}{s^2}$	kT	$\frac{Tz}{(z - 1)^2}$
$\frac{1}{s + a}$	e^{-akt}	$\frac{z}{z - e^{-aT}}$
$\frac{1}{(s + a)^2}$	kTe^{-akt}	$\frac{zTe^{-aT}}{(z - e^{-aT})^2}$
$\frac{1}{s^2 + a^2}$	$\sin(akt)$	$\frac{z \sin aT}{z^2 - (2 \cos aT)z + 1}$



The z-Plane

z-domain poles and zeros can be plotted just like s-domain poles and zeros (of the \mathcal{L}):

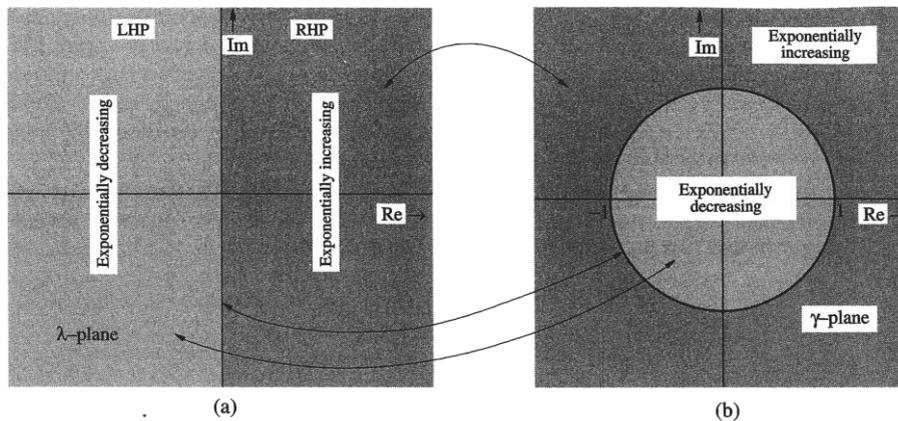
- S-plane:
- $z = e^{sT}$ Plane



Discrete-Time Exponential γ^k

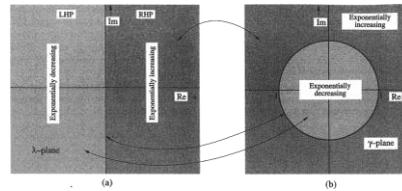
Recall we defined $z = e^{sT}$
Thus, this is the γ plane \rightarrow

$$e^{\lambda k} = \gamma^k$$



Discrete-Time Exponential γ^k

- $e^{\lambda k} = \gamma^k$
- $\gamma = e^\lambda$ or $\lambda = \ln \gamma$



- In discrete-time systems, unlike the continuous-time case, the form γ^k proves more convenient than the form $e^{\lambda k}$

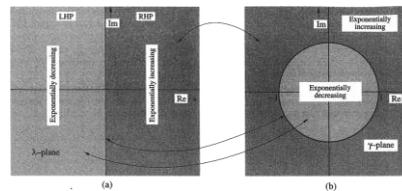
Why?

- Consider $e^{j\Omega k}$ ($\lambda = j\Omega \therefore$ constant amplitude oscillatory)
- $e^{j\Omega k} \rightarrow \gamma^k$, for $\gamma \equiv e^{j\Omega}$
- $|e^{j\Omega}| = 1$, hence $|\gamma| = 1$



Discrete-Time Exponential γ^k

- Consider $e^{\lambda k}$
When λ : LHP
- Then
- $\gamma = e^\lambda$
- $\gamma = e^\lambda = e^{a+jb} = e^a e^{jb}$
- $|\gamma| = |e^a e^{jb}| = |e^a| \because |e^{jb}| = 1$



Properties of the the z-transform

- Some useful properties
 - **Delay by n samples:** $Z\{f(k - n)\} = z^{-n}F(z)$
 - **Linear:** $Z\{af(k) + bg(k)\} = aF(z) + bG(z)$
 - **Convolution:** $Z\{f(k) * g(k)\} = F(z)G(z)$

So, all those block diagram manipulation tools you know and love will work just the same!



More Z-Transform Properties

- Time Reversal

$$x[n] \leftrightarrow X(z) \quad \text{ROC} = R$$

$$x[-n] \leftrightarrow X\left(\frac{1}{z}\right) \quad R' = \frac{1}{R}$$

- Multiplication by z^n

$$x[n] \leftrightarrow X(z) \quad \text{ROC} = R$$

$$z_0^n x[n] \leftrightarrow X\left(\frac{z}{z_0}\right) \quad R' = |z_0| R$$

- Multiplication by n (or Differentiation in z):

$$x[n] \leftrightarrow X(z) \quad \text{ROC} = R$$

$$nx[n] \leftrightarrow -z \frac{dX(z)}{dz} \quad R' = R$$

- Convolution

$$x_1[n] \leftrightarrow X_1(z) \quad \text{ROC} = R_1$$

$$x_2[n] \leftrightarrow X_2(z) \quad \text{ROC} = R_2$$

$$x_1[n] * x_2[n] \leftrightarrow X_1(z)X_2(z) \quad R' \supset R_1 \cap R_2$$



Z-Transform Properties: Time Shifting

$$y[n - n_0] \xleftrightarrow{\mathcal{Z}} z^{-n_0} Y(z)$$

$$\begin{aligned} y_2[n] &= y[n - n_0] \\ Y_2(e^{j\omega}) &= \sum_{k=-\infty}^{\infty} y[k - n_0] z^{-k} \\ &= \sum_{l=-\infty}^{\infty} y[l] z^{-(l+n_0)} \\ &= z^{-n_0} Y(z) \end{aligned}$$

- Two Special Cases:
- z^{-1} : the *unit-delay operator*:

$$x[n - 1] \leftrightarrow z^{-1} X(z) \quad R' = R \cap \{0 < |z|\}$$

- z : *unit-advance operator*:

$$x[n + 1] \leftrightarrow z X(z) \quad R' = R \cap \{|z| < \infty\}$$



An example!

- Back to our difference equation:

$$y(k) = x(k) + Ax(k - 1) - By(k - 1)$$

becomes

$$\begin{aligned} Y(z) &= X(z) + Az^{-1}X(z) - Bz^{-1}Y(z) \\ (z + B)Y(z) &= (z + A)X(z) \end{aligned}$$

which yields the transfer function:

$$\frac{Y(z)}{X(z)} = \frac{z + A}{z + B}$$

Note: It is also not uncommon to see systems expressed as polynomials in z^{-n}



BREAK

LTI(D) Systems Properties

System Stability

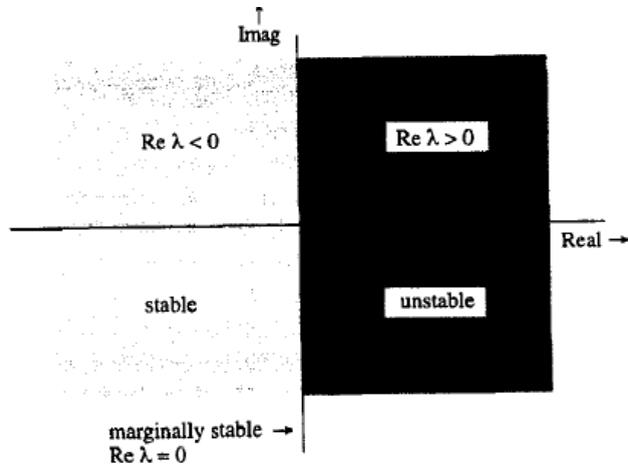
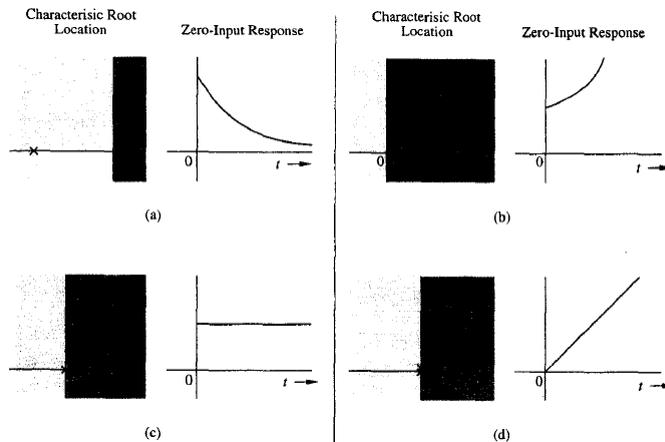


Fig. 2.15 Characteristic roots location and system stability.

Lathi, p. 149

System Stability [II]



Lathi, p. 150

System Stability [III]

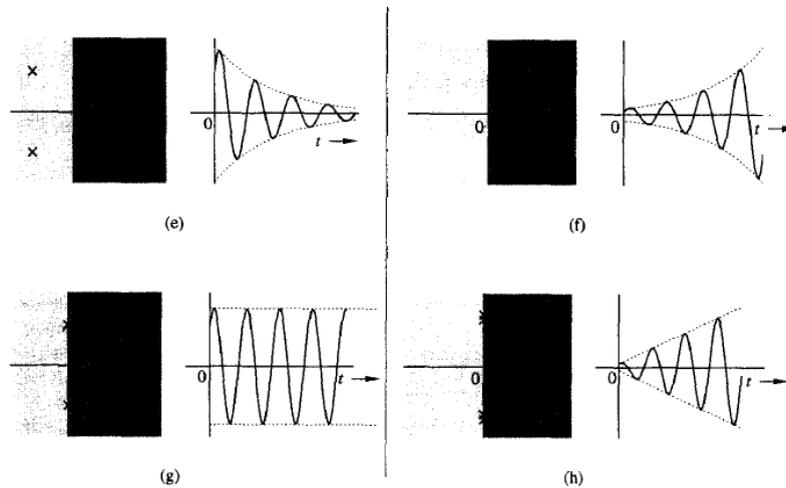
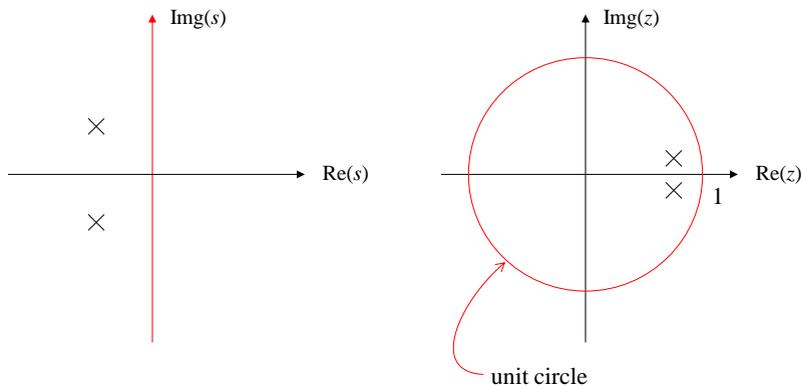


Fig. 2.16 Location of characteristic roots and the corresponding characteristic modes.



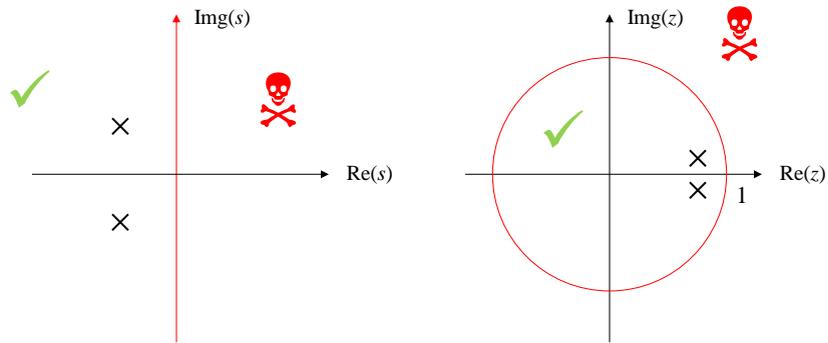
γ -plane Stability

- For a γ -Plane (e.g. the one the z -domain is embedded in) the unit circle is the system stability bound



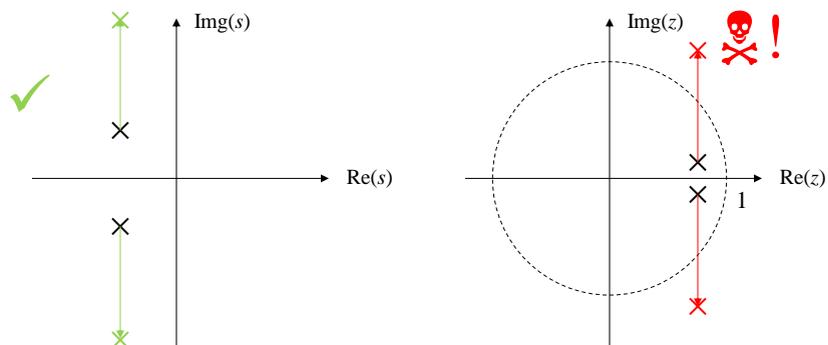
γ -plane Stability

- That is, in the z -domain, the unit circle is the system stability bound



z-plane stability

- The z -plane root-locus in closed loop feedback behaves just like the s -plane:



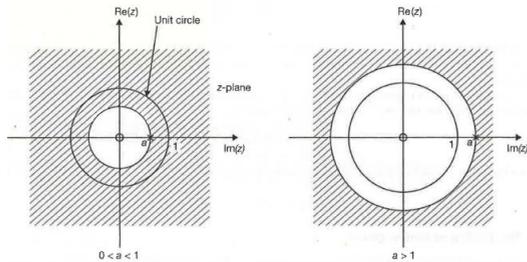
Region of Convergence

- For the convergence of $X(z)$ we require that

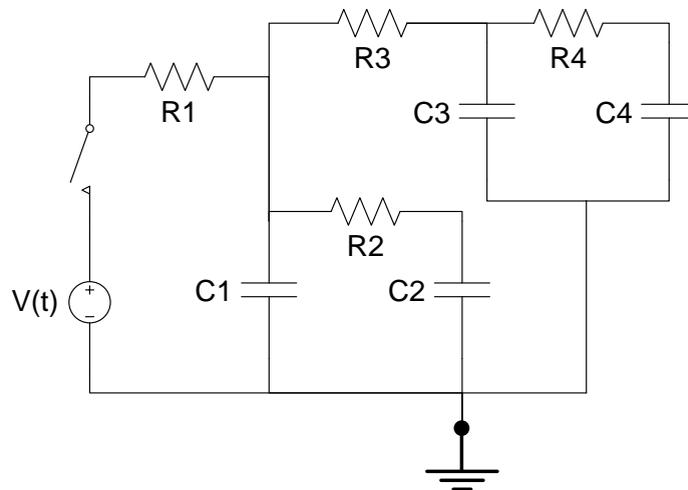
$$\sum_{n=0}^{\infty} |az^{-1}|^n < \infty$$

- Thus, the ROC is the range of values of z for which $|az^{-1}| < 1$ or, equivalently, $|z| > |a|$. Then

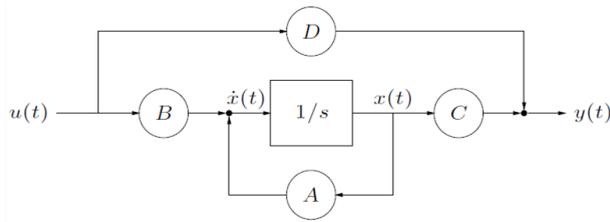
$$X(z) = \frac{z}{z-a} \quad |z| > |a|$$



An Example Circuit...



Frame as a LDS → LTI LTS → LTID LTS



$$\dot{x} = Ax + Bu$$
$$y = Cx + Du$$

- Ax is the drift term (of \dot{x})
- Bu is the input term (of \dot{x})

Ref: Boyd, EE263, 13-4



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Transfer Function

Take the Laplace transform of $\dot{x} = Ax + Bu$

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

$$X(s) = (sI - A)^{-1}x(0) + (sI - A)^{-1}BU(s)$$

$$\Rightarrow x(t) = e^{tA}x(0) + \int_0^t e^{(t-\tau)A} \cdot Bu(\tau) d\tau$$

- $e^{tA}x(0)$: unforced or autonomous response
- $e^{tA}B$: input-to-state → impulse response matrix
- $(sI - A)^{-1}B$: transfer function or transfer matrix

Ref: Boyd, EE263, 13-6



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Transfer Function [2]

with $y = \mathbf{C}x + \mathbf{D}u$ we have:

$$Y(s) = \mathbf{C}(s\mathbf{I} - \mathbf{A})^{-1}x(0) + (\mathbf{C}(s\mathbf{I} - \mathbf{A})^{-1}\mathbf{B} + \mathbf{D})U(s)$$

$$\Rightarrow y(t) = \mathbf{C}e^{t\mathbf{A}}x(0) + \int_0^t \mathbf{C}e^{(t-\tau)\mathbf{A}} \cdot \mathbf{B}u(\tau) d\tau + \mathbf{D}u(t)$$

- $\mathbf{C}e^{t\mathbf{A}}x(0)$: initial condition
- $H(s) = \mathbf{C}(s\mathbf{I} - \mathbf{A})^{-1}\mathbf{B} + \mathbf{D}$: transfer function/matrix
- $h(t) = \mathbf{C}e^{t\mathbf{A}} \cdot \mathbf{B} + \mathbf{D}\delta(t)$: impulse response

With zero initial conditions we have:

- $Y(s) = H(s)U(s), y = h * u$

Ref: Boyd, EE263, 13-7



Impulse Response

impulse response $h(t) = \mathbf{C}e^{t\mathbf{A}}\mathbf{B} + \mathbf{D}\delta(t)$

with $x(0) = 0, y = h * u, i.e.,$

$$y_i(t) = \sum_{j=1}^m \int_0^t h_{ij}(t-\tau)u_j(\tau) d\tau$$

interpretations:

- $h_{ij}(t)$ is impulse response from j th input to i th output
- $h_{ij}(t)$ gives y_i when $u(t) = e_j\delta$
- $h_{ij}(\tau)$ shows how dependent output i is, on what input j was, τ seconds ago
- i indexes output; j indexes input; τ indexes time lag

Ref: Boyd, EE263, 13-9



Step Response

the *step response* or *step matrix* is given by

$$s(t) = \int_0^t h(\tau) d\tau$$

interpretations:

- $s_{ij}(t)$ is step response from j th input to i th output
- $s_{ij}(t)$ gives y_i when $u = e_j$ for $t \geq 0$

for invertible A , we have

$$s(t) = CA^{-1}(e^{tA} - I)B + D$$

Ref: Boyd, EE263, 13-10



Static (DC) Gain Matrix

- transfer function at $s = 0$ is $H(0) = -CA^{-1}B + D \in \mathbf{R}^{m \times p}$
- DC transfer function describes system under *static* conditions, *i.e.*, x , u , y constant:

$$0 = \dot{x} = Ax + Bu, \quad y = Cx + Du$$

eliminate x to get $y = H(0)u$

- if system is stable,

$$H(0) = \int_0^\infty h(t) dt = \lim_{t \rightarrow \infty} s(t)$$

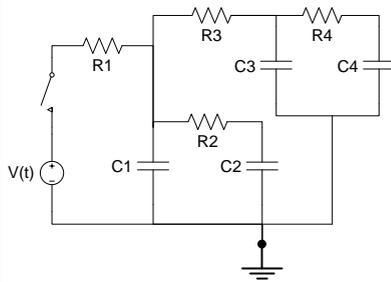
$$\text{(recall: } H(s) = \int_0^\infty e^{-st}h(t) dt, s(t) = \int_0^t h(\tau) d\tau)$$

if $u(t) \rightarrow u_\infty \in \mathbf{R}^m$, then $y(t) \rightarrow y_\infty \in \mathbf{R}^p$ where $y_\infty = H(0)u_\infty$

Ref: Boyd, EE263, 13-18



Back to the Example



$$\rightarrow \dot{x} = \begin{bmatrix} -3 & 1 & 1 & 0 \\ 1 & -1 & 0 & 0 \\ 1 & 0 & -2 & 1 \\ 0 & 0 & 1 & -1 \end{bmatrix} x + \begin{bmatrix} 1 \\ 0 \\ 0 \\ 0 \end{bmatrix} u, \quad y = x$$

> eig(A)

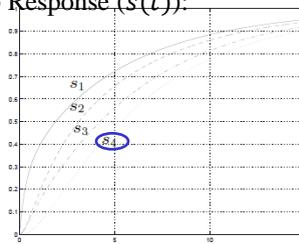
> -3.9563

-2.2091

-0.6617

-0.1729

Step Response, $s(t)$:



Ref: Boyd, EE263, 13-15

BREAK

Discretization with piecewise constant inputs

linear system $\dot{x} = Ax + Bu, y = Cx + Du$

suppose $u_d : \mathbf{Z}_+ \rightarrow \mathbf{R}^m$ is a sequence, and

$$u(t) = u_d(k) \quad \text{for } kh \leq t < (k+1)h, \quad k = 0, 1, \dots$$

define sequences

$$x_d(k) = x(kh), \quad y_d(k) = y(kh), \quad k = 0, 1, \dots$$

- $h > 0$ is called the *sample interval* (for x and y) or *update interval* (for u)
- u is piecewise constant (called *zero-order-hold*)
- x_d, y_d are sampled versions of x, y

Ref: Boyd, EE263, 13-20



Discretization with piecewise constant inputs [2]

$$\begin{aligned} x_d(k+1) &= x((k+1)h) \\ &= e^{hA}x(kh) + \int_0^h e^{\tau A}Bu((k+1)h-\tau) d\tau \\ &= e^{hA}x_d(k) + \left(\int_0^h e^{\tau A} d\tau \right) B u_d(k) \end{aligned}$$

$x_d, u_d,$ and y_d satisfy discrete-time LDS equations

$$x_d(k+1) = A_d x_d(k) + B_d u_d(k), \quad y_d(k) = C_d x_d(k) + D_d u_d(k)$$

where

$$A_d = e^{hA}, \quad B_d = \left(\int_0^h e^{\tau A} d\tau \right) B, \quad C_d = C, \quad D_d = D$$

called *discretized system*

if A is invertible, we can express integral as

$$\int_0^h e^{\tau A} d\tau = A^{-1} (e^{hA} - I)$$

stability: if eigenvalues of A are $\lambda_1, \dots, \lambda_n$, then eigenvalues of A_d are $e^{h\lambda_1}, \dots, e^{h\lambda_n}$

discretization preserves stability properties since

$$\Re \lambda_i < 0 \Leftrightarrow |e^{h\lambda_i}| < 1$$

for $h > 0$

Ref: Boyd, EE263, 13-22



Convolution

Convolution Definition

The **convolution** of two functions $f_1(t)$ and $f_2(t)$ is defined as:

$$\begin{aligned} f(t) &= \int_{-\infty}^{\infty} f_1(\tau) f_2(t - \tau) d\tau \\ &= f_1(t) * f_2(t) \end{aligned}$$

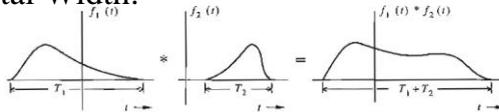


Convolution Properties

$$f_1(t) * f_2(t) \equiv \int_{-\infty}^{\infty} f_1(\tau) f_2(t - \tau) d\tau$$

Properties:

- **Commutative:** $f_1(t) * f_2(t) = f_2(t) * f_1(t)$
- **Distributive:** $f_1(t) * [f_2(t) + f_3(t)] = f_1(t) * f_2(t) + f_1(t) * f_3(t)$
- **Associative:** $f_1(t) * [f_2(t) * f_3(t)] = [f_1(t) * f_2(t)] * f_3(t)$
- **Shift:**
if $f_1(t) * f_2(t) = c(t)$, then $f_1(t - \mathbf{T}) * f_2(t) = f_1(t) * f_2(t - \mathbf{T}) = c(t - \mathbf{T})$
- **Identity (Convolution with an Impulse):**
 $f(t) * \delta(t) = f(t)$
- **Total Width:**



Based on Lathi, SPLS, Sec 2.4-1



Convolution Properties [II]

- Convolution systems are **linear**:
$$h * (\alpha u_1 + \beta u_2) = \alpha(h * u_1) + \beta(h * u_2)$$
- Convolution systems are **causal**: the output $y(t)$ at time t depends only on past inputs
- Convolution systems are **time-invariant**
(if we shift the signal, the output similarly shifts)

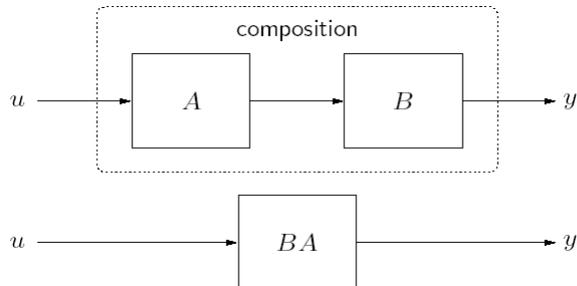
$$\rightarrow \tilde{u}(t) = \begin{cases} 0 & t < T \\ u(t - T) & t \geq 0 \end{cases}$$

$$\tilde{y}(t) = \begin{cases} 0 & t < T \\ y(t - T) & t \geq 0 \end{cases}$$



Convolution Properties [III]

- Composition of convolution systems corresponds to:
 - multiplication of transfer functions
 - convolution of impulse responses

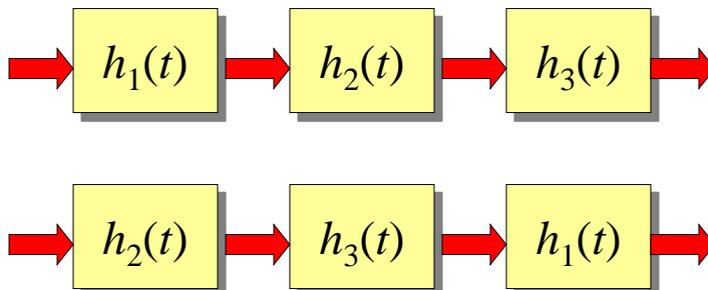


- Thus:
 - We can manipulate block diagrams with transfer functions as if they were simple gains
 - convolution systems commute with each other



Properties of Convolution: Distributive Property

$$[f_1(t) * f_2(t)] * f_3(t) = f_1(t) * [f_2(t) * f_3(t)]$$



- The two systems are identical!



Properties of Convolution: Commutative Property

$$f_1(t) * f_2(t) = f_2(t) * f_1(t)$$

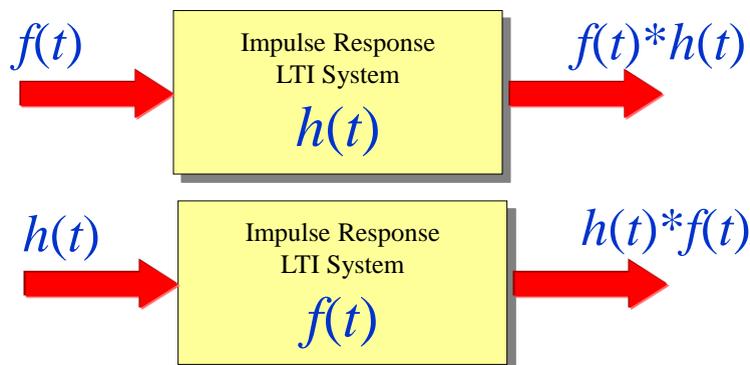
$$\begin{aligned} f_1(t) * f_2(t) &= \int_{-\infty}^{\infty} f_1(\tau) f_2(t - \tau) d\tau = \int_{\tau=-\infty}^{\tau=\infty} f_1(\tau) f_2(t - \tau) d\tau \\ &= \int_{t-\tau=-\infty}^{t-\tau=\infty} f_1(t - \tau) f_2[t - (t - \tau)] d(t - \tau) \\ &= - \int_{\tau=\infty}^{\tau=-\infty} f_1(t - \tau) f_2(\tau) d\tau \\ &= \int_{-\infty}^{\infty} f_1(t - \tau) f_2(\tau) d\tau = f_2(t) * f_1(t) \end{aligned}$$

Source: URI ELE436



Properties of Convolution: LTI System Response

$$f_1(t) * f_2(t) = f_2(t) * f_1(t)$$



Source: URI ELE436



Properties of Convolution

$$f(t) * \delta(t) = f(t) \quad f(t) \rightarrow \delta(t) \rightarrow f(t)$$

$$\begin{aligned} f(t) * \delta(t) &= \int_{-\infty}^{\infty} f(\tau) \delta(t - \tau) d\tau \\ &= \int_{-\infty}^{\infty} f(t - \tau) \delta(\tau) d\tau \\ &= f(t) \end{aligned}$$

Source: URI ELE436



Properties of Convolution

$$f(t) * \delta(t) = f(t) \quad f(t) \rightarrow \delta(t) \rightarrow f(t)$$

$$f(t) * \delta(t - T) = f(t - T)$$

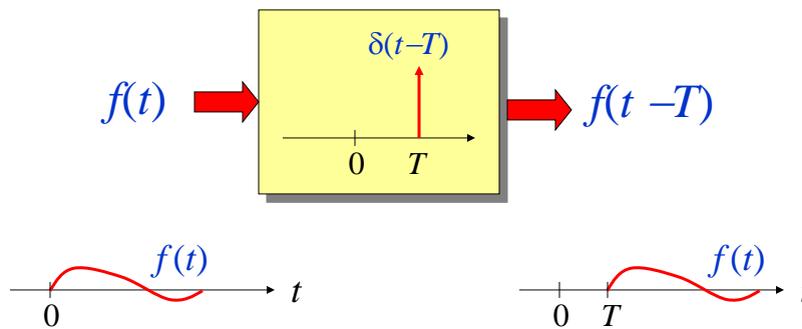
$$\begin{aligned} f(t) * \delta(t - T) &= \int_{-\infty}^{\infty} f(\tau) \delta(t - T - \tau) d\tau \\ &= \int_{-\infty}^{\infty} f(t - T - \tau) \delta(\tau) d\tau \\ &= f(t - T) \end{aligned}$$

Source: URI ELE436



Properties of Convolution

$$f(t) * \delta(t-T) = f(t-T)$$



Source: URI ELE436



Properties of Convolution

$$f_1(t) * f_2(t) \xleftrightarrow{\mathcal{F}} F_1(j\omega)F_2(j\omega)$$

$$\begin{aligned} F[f_1(t) * f_2(t)] &= \int_{-\infty}^{\infty} \left[\int_{-\infty}^{\infty} f_1(\tau) f_2(t-\tau) d\tau \right] e^{-j\omega t} dt \\ &= \int_{-\infty}^{\infty} f_1(\tau) \left[\int_{-\infty}^{\infty} f_2(t-\tau) e^{-j\omega t} dt \right] d\tau \\ &= \int_{-\infty}^{\infty} f_1(\tau) F_2(j\omega) e^{-j\omega \tau} d\tau \\ &= F_2(j\omega) \int_{-\infty}^{\infty} f_1(\tau) e^{-j\omega \tau} d\tau = F_1(j\omega) F_2(j\omega) \end{aligned}$$

Time Domain
convolution

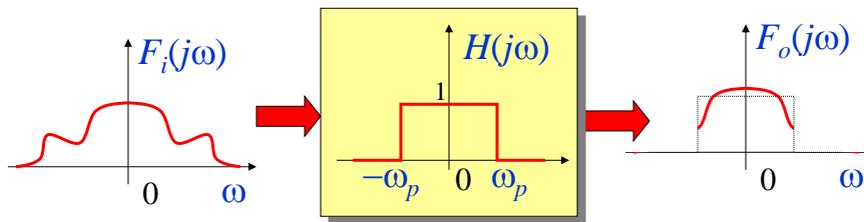
Frequency Domain
multiplication

Source: URI ELE436



Properties of Convolution

$$f_1(t) * f_2(t) \xrightarrow{\mathcal{F}} F_1(j\omega)F_2(j\omega)$$



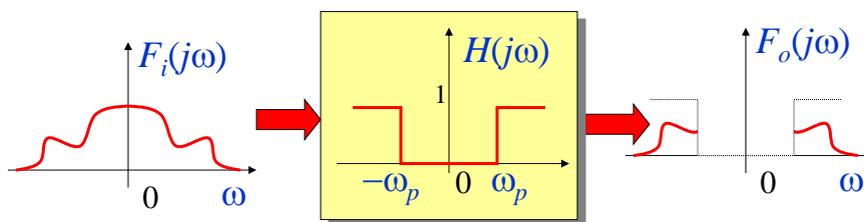
An Ideal Low-Pass Filter

Source: URI ELE436



Properties of Convolution

$$f_1(t) * f_2(t) \xrightarrow{\mathcal{F}} F_1(j\omega)F_2(j\omega)$$



An Ideal High-Pass Filter

Source: URI ELE436



Discrete Convolution

$$y[n] = x[n] * h[n] = \sum_{m=-\infty}^{\infty} x[m] \cdot h[n-m] = \sum_{m=-\infty}^{\infty} h[m] \cdot x[n-m]$$

Consider this for the discrete case:

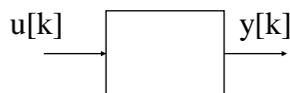
1. Rename the independent variable as **m**. You now have **x[m]** and **h[m]**. Flip **h[m]** over the origin. This is **h[-m]**
2. Shift **h[-m]** as far left as possible to a point “**n**”, where the two signals barely touch. This is **h[n-m]**
3. Multiply the two signals and sum over all values of **m**. This is the convolution sum for the specific “**n**” picked above.
4. Shift / move **h[-m]** to the right by one sample, and obtain a new **h[n-m]**. Multiply and sum over all **m**.
5. Repeat 2~4 until **h[n-m]** no longer overlaps with **x[m]**, i.e., shifted out of the **x[m]** zone.

→ The “**n**” dependency of $y[n]$ deserves some care:
For each value of “**n**” the convolution sum must be computed *separately* over all values of a dummy variable “**m**”.



Discrete-Time Systems & Discrete Convolution [I]

Will consider linear time-invariant (LTI) systems



Linear :

input $u_1[k]$ -> output $y_1[k]$

input $u_2[k]$ -> output $y_2[k]$

hence $a \cdot u_1[k] + b \cdot u_2[k] \rightarrow a \cdot y_1[k] + b \cdot y_2[k]$

Time-invariant (shift-invariant)

input $u[k]$ -> output $y[k]$

hence input $u[k-T]$ -> output $y[k-T]$



Discrete-Time Systems & Discrete Convolution [2]

Will consider causal systems

iff for all input signals with $u[k]=0, k < 0 \rightarrow$ output $y[k]=0, k < 0$

Impulse response

input $\dots, 0, 0, \overset{k=0}{1}, 0, 0, 0, \dots \rightarrow$ output $\dots, 0, 0, \overset{k=0}{h[0]}, h[1], h[2], h[3], \dots$

General input $u[0], u[1], u[2], u[3]$ (cfr. linearity & shift-invariance!)

$$\begin{bmatrix} y[0] \\ y[1] \\ y[2] \\ y[3] \\ y[4] \\ y[5] \end{bmatrix} = \begin{bmatrix} h[0] & 0 & 0 & 0 \\ h[1] & h[0] & 0 & 0 \\ h[2] & h[1] & h[0] & 0 \\ 0 & h[2] & h[1] & h[0] \\ 0 & 0 & h[2] & h[1] \\ 0 & 0 & 0 & h[2] \end{bmatrix} \begin{bmatrix} u[0] \\ u[1] \\ u[2] \\ u[3] \end{bmatrix}$$

this is called a 'Toeplitz' matrix

Discrete-Time Systems & Discrete Convolution [3]

$u[0], u[1], u[2], u[3]$  $y[0], y[1], \dots$

$$\begin{bmatrix} y[0] \\ y[1] \\ y[2] \\ y[3] \\ y[4] \\ y[5] \end{bmatrix} = \begin{bmatrix} h[0] & 0 & 0 & 0 \\ h[1] & h[0] & 0 & 0 \\ h[2] & h[1] & h[0] & 0 \\ 0 & h[2] & h[1] & h[0] \\ 0 & 0 & h[2] & h[1] \\ 0 & 0 & 0 & h[2] \end{bmatrix} \begin{bmatrix} u[0] \\ u[1] \\ u[2] \\ u[3] \end{bmatrix}$$

$h[0], h[1], h[2], 0, 0, \dots$

$$y[k] = \sum_{\bar{k}} h[k - \bar{k}] \cdot u[\bar{k}] \stackrel{D}{=} h[k] * u[k] = \text{'convolution sum'}$$

(=more convenient than Toeplitz matrix notation when considering (infinitely) long input and impulse response sequences)

Discrete-Time Systems & Discrete Convolution [4]

Z-Transform of system $h[k]$ and signals $u[k], y[k]$

Definition:

Input/output relation: $H(z) = \sum_k h[k].z^{-k}$ $U(z) = \sum_k u[k].z^{-k}$ $Y(z) = \sum_k y[k].z^{-k}$

$$\underbrace{\begin{bmatrix} y[0] \\ y[1] \\ y[2] \\ y[3] \\ y[4] \\ y[5] \end{bmatrix}}_{Y(z)} = \underbrace{\begin{bmatrix} 1 & z^{-1} & z^{-2} & z^{-3} & z^{-4} & z^{-5} \end{bmatrix}}_{H(z)} \cdot \underbrace{\begin{bmatrix} h[0] & 0 & 0 & 0 \\ h[1] & h[0] & 0 & 0 \\ h[2] & h[1] & h[0] & 0 \\ 0 & h[2] & h[1] & h[0] \\ 0 & 0 & h[2] & h[1] \\ 0 & 0 & 0 & h[2] \end{bmatrix}}_{\begin{bmatrix} 1 & z^{-1} & z^{-2} & z^{-3} \end{bmatrix}} \cdot \underbrace{\begin{bmatrix} u[0] \\ u[1] \\ u[2] \\ u[3] \end{bmatrix}}_{U(z)}$$

$\Rightarrow Y(z) = H(z).U(z)$ $H(z)$ is 'transfer function'



Matrix Formulation of Convolution

$y = Hx$

$$\begin{bmatrix} 3 \\ 8 \\ 14 \\ 20 \\ 26 \\ 14 \\ 5 \end{bmatrix} = \begin{bmatrix} 1 & 2 & 3 & 0 & 0 & 0 & 0 & 0 & 0 \\ 0 & 1 & 2 & 3 & 0 & 0 & 0 & 0 & 0 \\ 0 & 0 & 1 & 2 & 3 & 0 & 0 & 0 & 0 \\ 0 & 0 & 0 & 1 & 2 & 3 & 0 & 0 & 0 \\ 0 & 0 & 0 & 0 & 1 & 2 & 3 & 0 & 0 \\ 0 & 0 & 0 & 0 & 0 & 1 & 2 & 3 & 0 \\ 0 & 0 & 0 & 0 & 0 & 0 & 1 & 2 & 3 \end{bmatrix} \cdot \begin{bmatrix} 0 \\ 0 \\ 1 \\ 2 \\ 3 \\ 4 \\ 5 \\ 0 \\ 0 \end{bmatrix}$$

Toeplitz Matrix



Graphical Understanding of Convolution

→ For $c(\tau) = (f * g)(t) = \int_{-\infty}^{\infty} f(\tau)g(t - \tau) d\tau$:

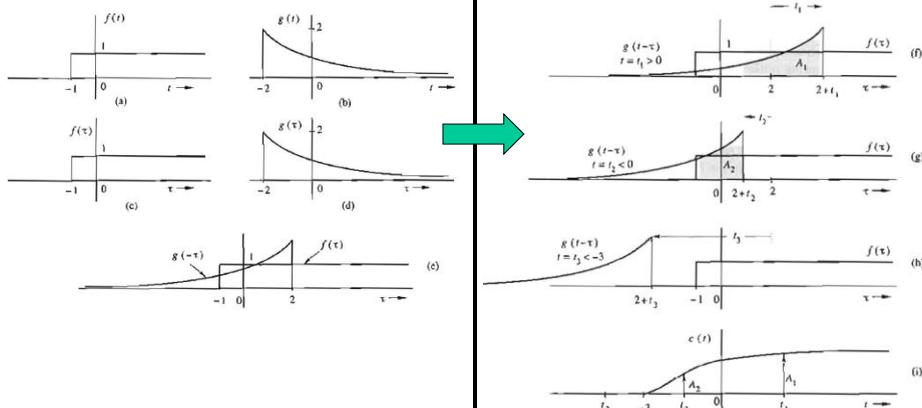
1. Keep the function $f(\tau)$ fixed
2. **Flip** (invert) the function $g(\tau)$ about the vertical axis ($\tau=0$)
= this is $g(-\tau)$
3. **Shift** this frame ($g(-\tau)$) along τ (horizontal axis) by t_0 .
= this is $g(t_0 - \tau)$

→ For $c(t_0)$:

4. $c(t_0)$ = the area under the product of $f(\tau)$ and $g(t_0 - \tau)$
5. Repeat this procedure, shifting the frame by different values (positive and negative) to obtain $c(t)$ for all values of t .



Graphical Understanding of Convolution (Ex)



Another View

e.g. convolution

$$x(n) = 1 \ 2 \ 3 \ 4 \ 5$$

$$h(n) = 3 \ 2 \ 1$$

x(k)	0 0 1	2 3 4 5	0 0 1 2	3 4 5	0 0 1 2 3	4 5	
h(n,k)	1 2 3	0 0 0 0	0 1 2 3	0 0 0	0 0 1 2 3	0 0	h(n-k)
y(n,k)	3		2 6		1 4 9		
y(n)	3		8		14		

Sum over all k

Notice the gain



Convolution & Systems

- Convolution system with input u ($u(t) = 0, t < 0$) and output y :

$$y(t) = \int_0^t h(\tau)u(t - \tau) d\tau = \int_0^t h(t - \tau)u(\tau) d\tau$$

- abbreviated:

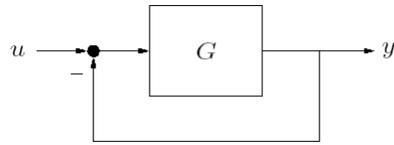
$$y = h * u$$

- in the frequency domain:

$$Y(s) = H(s)U(s)$$



Convolution & Feedback



- In the time domain:

$$y(t) = \int_0^t g(t - \tau)(u(\tau) - y(\tau)) d\tau$$

- In the frequency domain:

– $Y = G(U - Y)$

→ $Y(s) = H(s)U(s)$

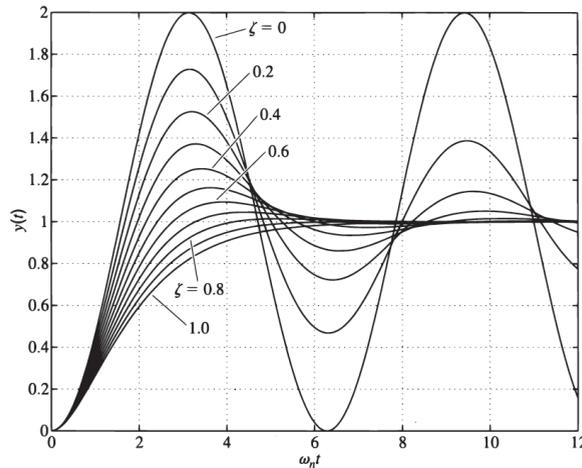
$$H(s) = \frac{G(s)}{1 + G(s)}$$



2nd Order LTID

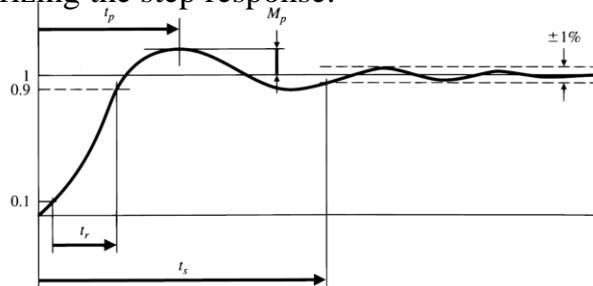
2nd Order System Response

- Response of a 2nd order system to increasing levels of damping:



2nd Order System Specifications

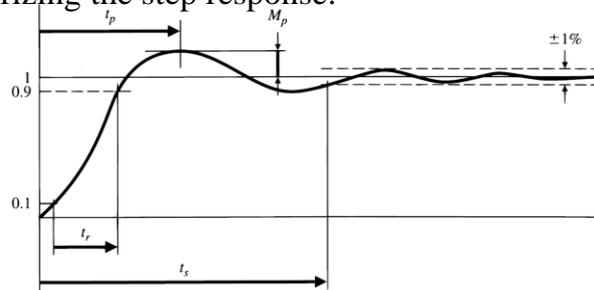
Characterizing the step response:



- Rise time (10% \rightarrow 90%): $t_r \approx \frac{1.8}{\omega_0}$
- Overshoot: $M_p \approx \frac{e^{-\pi\zeta}}{\sqrt{1-\zeta^2}}$
- Settling time (to 1%): $t_s = \frac{4.6}{\zeta\omega_0}$
- Steady state error to unit step: e_{ss}
- Phase margin: $\phi_{PM} \approx 100\zeta$

2nd Order System Specifications

Characterizing the step response:



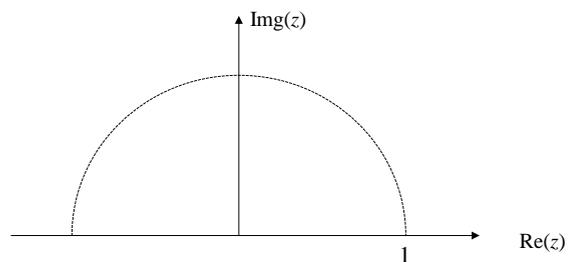
- Rise time (10% \rightarrow 90%) & Overshoot:
 $t_r, M_p \rightarrow \zeta, \omega_0$: Locations of dominant poles
- Settling time (to 1%):
 $t_s \rightarrow$ radius of poles: $|z| < 0.01^{1/T}$
- Steady state error to unit step:
 $e_{ss} \rightarrow$ final value theorem $e_{ss} = \lim_{z \rightarrow 1} \{(z-1)F(z)\}$



The z-plane [for all pole systems]

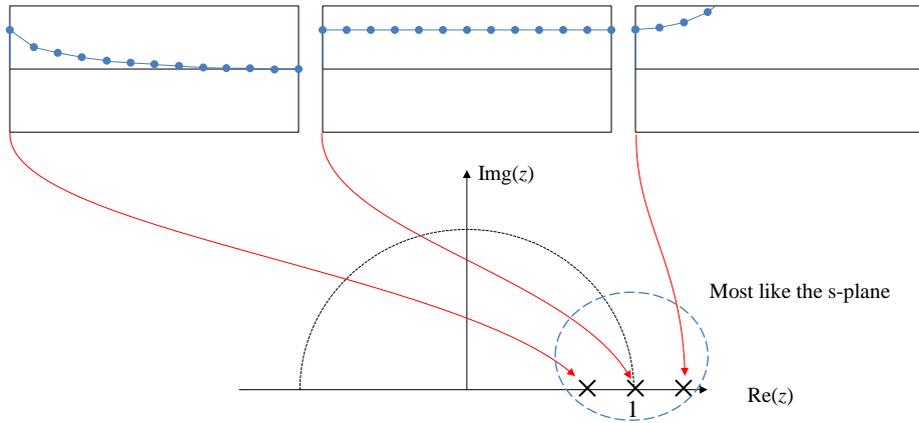
- We can understand system response by pole location in the z-plane

[Adapted from Franklin, Powell and Emami-Naeini]



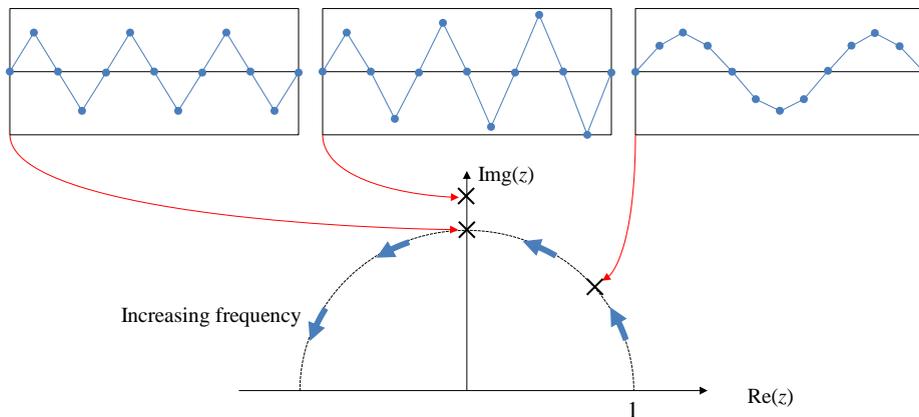
Effect of pole positions

- We can understand system response by pole location in the z -plane



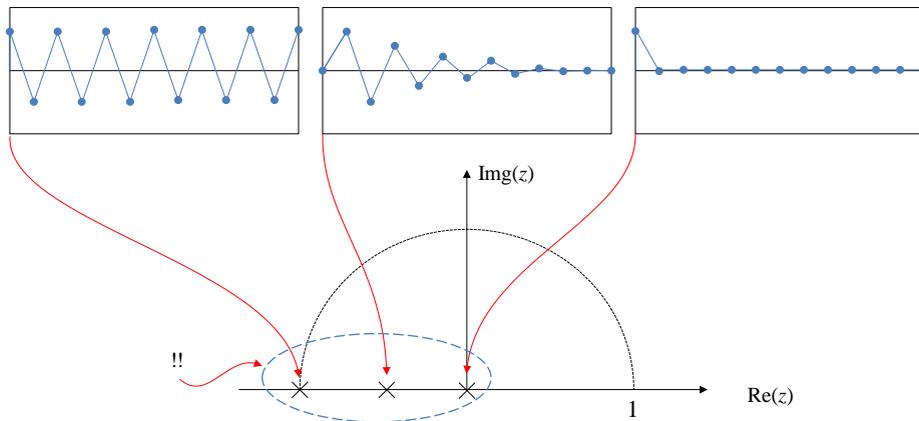
Effect of pole positions

- We can understand system response by pole location in the z -plane



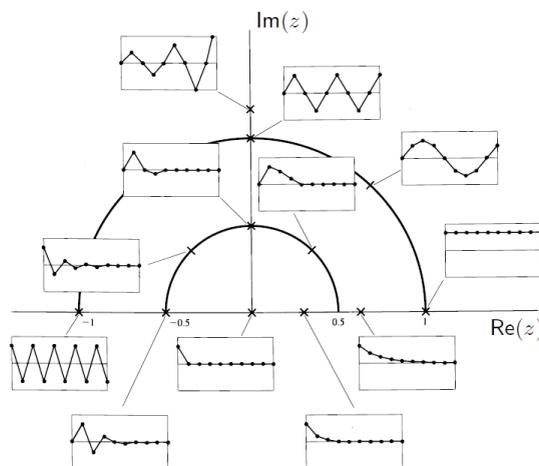
Effect of pole positions

- We can understand system response by pole location in the z -plane



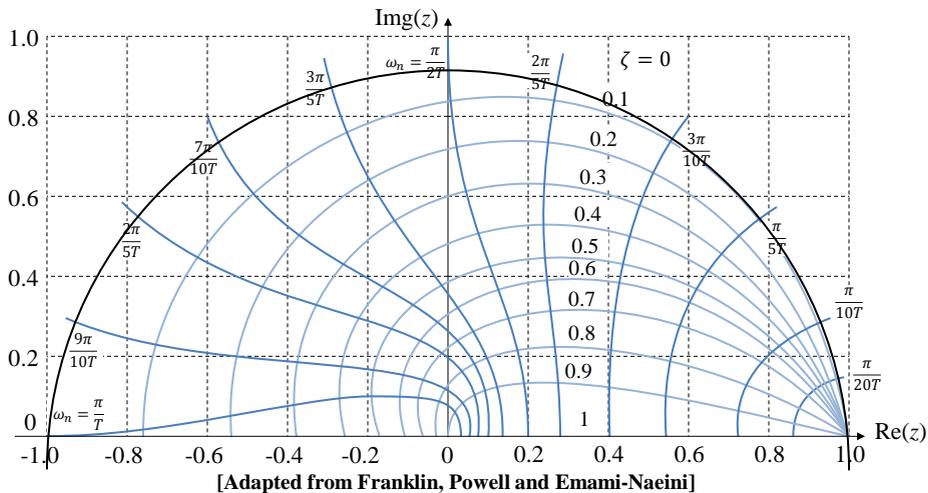
Pole positions in the z -plane

- Poles inside the unit circle are **stable**
- Poles outside the unit circle are **unstable**
- Poles on the unit circle are oscillatory
- Real poles at $0 < z < 1$ give exponential response
- Higher frequency of oscillation for larger r
- Lower apparent damping for larger r



Damping and natural frequency

$$z = e^{sT} \text{ where } s = -\zeta\omega_n \pm j\omega_n\sqrt{1-\zeta^2}$$



Second Order Digital Systems

Consider the z-transform of a decaying exponential signal:

$$y(t) = e^{-at} \cos(bt) \mathcal{U}(t)$$

$$(\mathcal{U}(t) = \text{unit step})$$

★ sample: $y(kT) = r^k \cos(k\theta) \mathcal{U}(kT)$

with $r = e^{-aT}$ & $\theta = bT$

★ transform:
$$Y(z) = \frac{1}{2} \frac{z}{(z - re^{j\theta})} + \frac{1}{2} \frac{z}{(z - re^{-j\theta})}$$

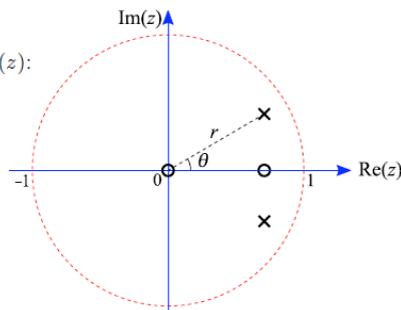
$$= \frac{z(z - r \cos \theta)}{(z - re^{j\theta})(z - re^{-j\theta})}$$

★ e.g. y_k is the pulse response of $G(z)$:

$$G(z) = \frac{z(z - r \cos \theta)}{(z - re^{j\theta})(z - re^{-j\theta})}$$

poles: $\begin{cases} z = re^{j\theta} \\ z = re^{-j\theta} \end{cases}$

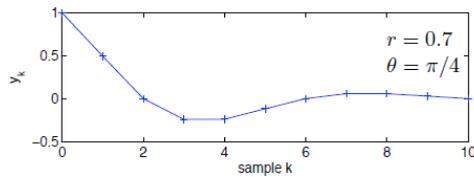
zeros: $\begin{cases} z = 0 \\ z = r \cos \theta \end{cases}$



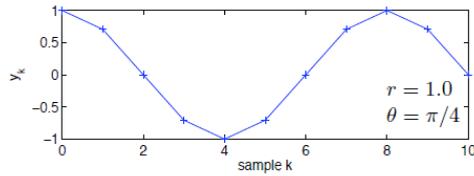
Response of 2nd order system [1/3]

Responses for varying r :

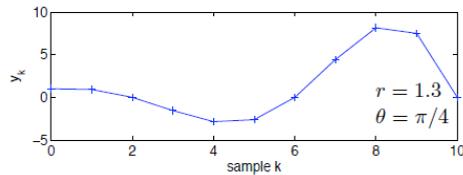
▷ $r < 1$
 ↓
 exponentially decaying envelope



▷ $r = 1$
 ↓
 sinusoidal response with $2\pi/\theta$ samples per period



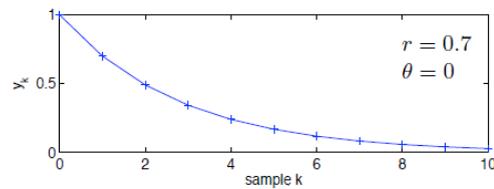
▷ $r > 1$
 ↓
 exponentially increasing envelope



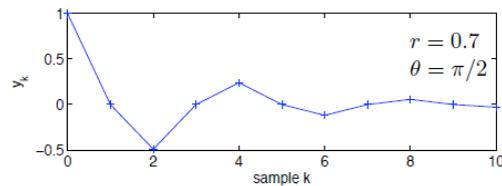
Response of 2nd order system [2/3]

Responses for varying θ :

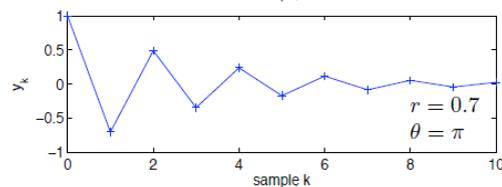
▷ $\theta = 0$
 ↓
 decaying exponential



▷ $\theta = \pi/2$
 ↓
 $2\pi/\theta = 4$ samples per period



▷ $\theta = \pi$
 ↓
 2 samples per period



Response of 2nd order system [3/3]

Some special cases:

- ▷ for $\theta = 0$, $Y(z)$ simplifies to:

$$Y(z) = \frac{z}{z - r}$$

⇒ exponentially decaying response

- ▷ when $\theta = 0$ and $r = 1$:

$$Y(z) = \frac{z}{z - 1}$$

⇒ unit step

- ▷ when $r = 0$:

$$Y(z) = 1$$

⇒ unit pulse

- ▷ when $\theta = 0$ and $-1 < r < 0$:

samples of alternating signs



Discrete-time transfer function

take \mathcal{Z} -transform of system equations

$$x(t+1) = Ax(t) + Bu(t), \quad y(t) = Cx(t) + Du(t)$$

yields

$$zX(z) - zx(0) = AX(z) + BU(z), \quad Y(z) = CX(z) + DU(z)$$

solve for $X(z)$ to get

$$X(z) = (zI - A)^{-1}zx(0) + (zI - A)^{-1}BU(z)$$

(note extra z in first term!)

hence

$$Y(z) = H(z)U(z) + C(zI - A)^{-1}zx(0)$$

where $H(z) = C(zI - A)^{-1}B + D$ is the *discrete-time transfer function*

note power series expansion of resolvent:

$$(zI - A)^{-1} = z^{-1}I + z^{-2}A + z^{-3}A^2 + \dots$$

Source: Boyd, Lecture Notes for EE263, 13-39



Ex: System Specifications → Control Design [1/4]

Design a controller for a system with:

- A continuous transfer function: $G(s) = \frac{0.1}{s(s + 0.1)}$
- A discrete ZOH sampler
- Sampling time (T_s): $T_s = 1$ s
- Controller:

$$u_k = -0.5u_{k-1} + 13(e_k - 0.88e_{k-1})$$

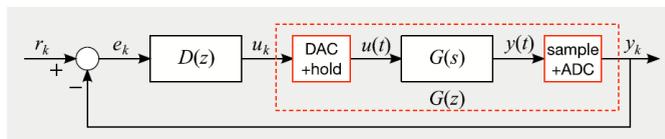
The closed loop system is required to have:

- $M_p < 16\%$
- $t_s < 10$ s
- $e_{ss} < 1$



Ex: System Specifications → Control Design [2/4]

- (a) Find the pulse transfer function of $G(s)$ plus the ZOH



$$G(z) = (1 - z^{-1})\mathcal{Z}\left\{\frac{G(s)}{s}\right\} = \frac{(z - 1)}{z}\mathcal{Z}\left\{\frac{0.1}{s^2(s + 0.1)}\right\}$$

e.g. look up $\mathcal{Z}\{a/s^2(s + a)\}$ in tables:

$$\begin{aligned} G(z) &= \frac{(z - 1)}{z} \frac{z \left((0.1 - 1 + e^{-0.1})z + (1 - e^{-0.1} - 0.1e^{-0.1}) \right)}{0.1(z - 1)^2(z - e^{-0.1})} \\ &= \frac{0.0484(z + 0.9672)}{(z - 1)(z - 0.9048)} \end{aligned}$$

- (b) Find the controller transfer function (using $z =$ shift operator):

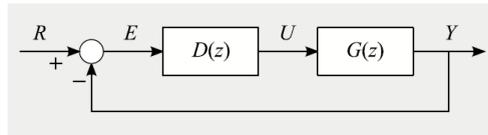
$$\frac{U(z)}{E(z)} = D(z) = 13 \frac{(1 - 0.88z^{-1})}{(1 + 0.5z^{-1})} = 13 \frac{(z - 0.88)}{(z + 0.5)}$$



Ex: System Specifications → Control Design [3/4]

2. Check the steady state error e_{ss} when $r_k =$ unit ramp

$$e_{ss} = \lim_{k \rightarrow \infty} e_k = \lim_{z \rightarrow 1} (z-1)E(z)$$



$$\frac{E(z)}{R(z)} = \frac{1}{1 + D(z)G(z)}$$

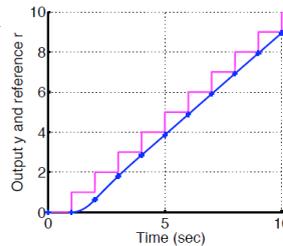
$$R(z) = \frac{Tz}{(z-1)^2}$$

$$\text{so } e_{ss} = \lim_{z \rightarrow 1} \left\{ (z-1) \frac{Tz}{(z-1)^2} \frac{1}{1 + D(z)G(z)} \right\} = \lim_{z \rightarrow 1} \frac{T}{(z-1)D(z)G(z)}$$

$$= \lim_{z \rightarrow 1} \frac{T}{(z-1) \frac{0.0484(z+0.9672)}{(z-1)(z-0.9048)} D(1)}$$

$$= \frac{1 - 0.9048}{0.0484(1 + 0.9672)D(1)} = 0.96$$

$$\Rightarrow e_{ss} < 1 \quad (\text{as required})$$



Ex: System Specifications → Control Design [4/4]

3. Step response: overshoot $M_p < 16\% \Rightarrow \zeta > 0.5$
 settling time $t_s < 10 \Rightarrow |z| < 0.01^{1/10} = 0.63$

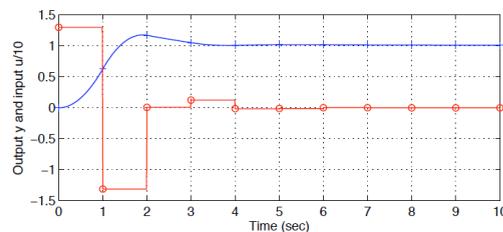
The closed loop poles are the roots of $1 + D(z)G(z) = 0$, i.e.

$$1 + 13 \frac{(z-0.88)}{(z+0.5)} \frac{0.0484(z+0.9672)}{(z-1)(z-0.9048)} = 0$$

$$\Rightarrow z = 0.88, -0.050 \pm j0.304$$

But the pole at $z = 0.88$ is cancelled by controller zero at $z = 0.88$, and

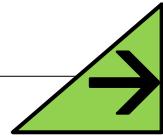
$$z = -0.050 \pm j0.304 = re^{\pm j\theta} \Rightarrow \begin{cases} r = 0.31, \theta = 1.73 \\ \zeta = 0.56 \end{cases}$$



all specs satisfied!



Next Time...



- **Frequency Response**
- **Review:**
 - Chapter 10 of Lathi
- Nothing like a filter to smooth a rough signals 😊

