



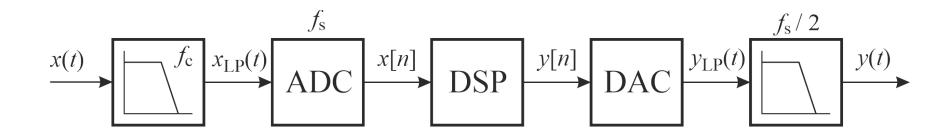
Applications



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- 6.3 Control Systems
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Time-discrete processing of time-continuous signals



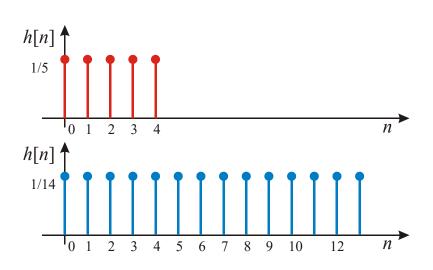


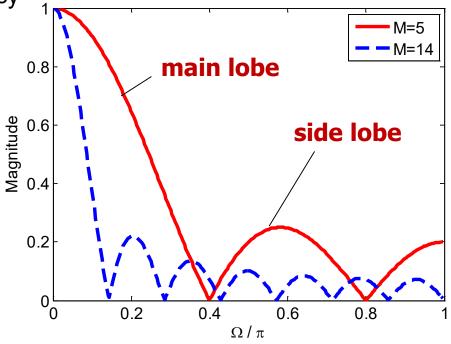
6.1 Time-Selective Filters

- FIR filters: The truncation of the impulse response is defined by a window function.
- Example: Moving average filter with rectangular window

• Filter order $M-1 \rightarrow$ cutoff frequency

Window type → stopband ripple





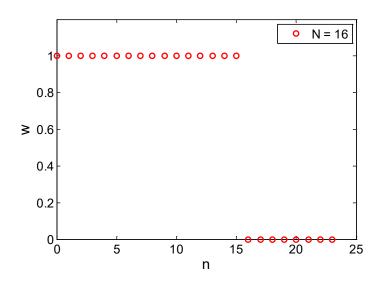


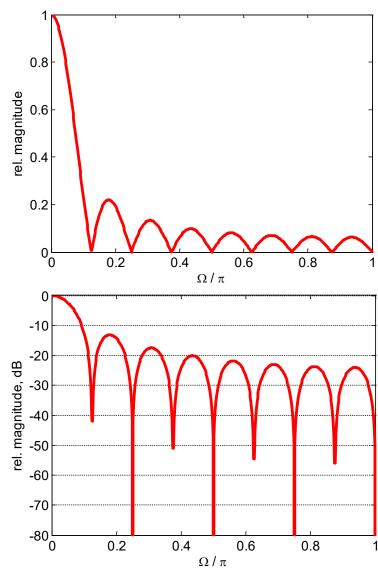
6.1.1 Window Functions

Rectangular Window

$$w[n] = \begin{cases} 1, & n = 0, \dots, N-1 \\ 0, & \text{otherwise} \end{cases}$$

$$W(e^{j\Omega}) = \frac{\sin(\Omega \cdot N/2)}{\sin(\Omega/2)} \cdot e^{-j\Omega \cdot (N-1)/2}$$





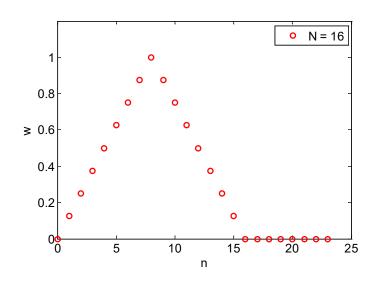
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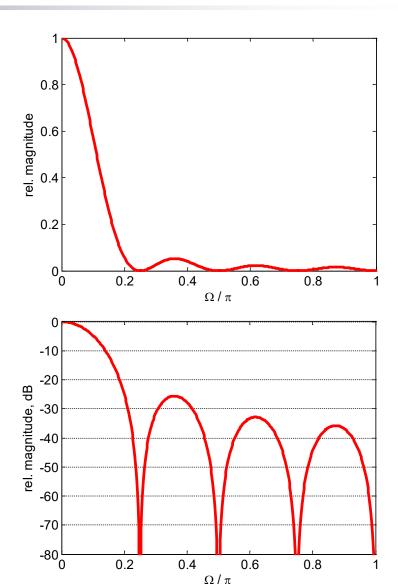


BARLETT Window - triangular window

$$w[n] = \begin{cases} 2n/N & , \ 0 \le n \le N/2 - 1 \\ 2(N-n)/N & , \ N/2 \le n \le N - 1 \\ 0 & , \ \text{otherwise} \end{cases}$$

$$W(e^{j\Omega}) = \frac{\sin^2(\Omega \cdot N/4)}{\sin^2(\Omega/2)} \cdot e^{-j\Omega \cdot (N-1)/2}$$



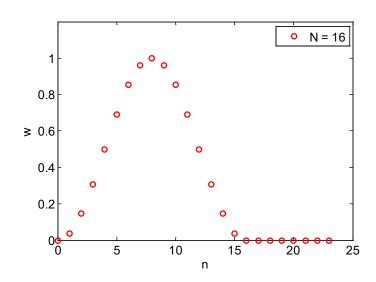


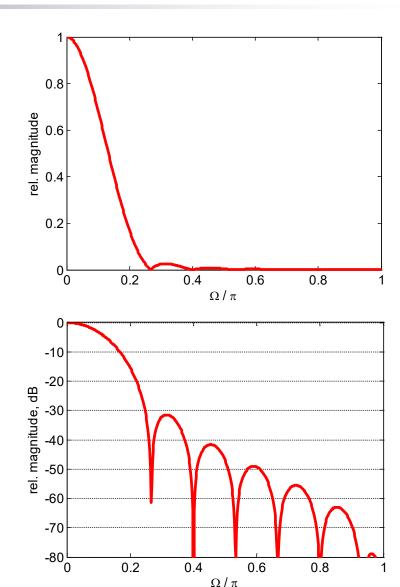
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HANNING Window

$$w[n] = \begin{cases} 0.5 \cdot \left(1 - \cos\left(\frac{2 \cdot \pi \cdot n}{N}\right) \right), & n = 0, ..., N - 1 \\ 0, & \text{otherwise} \end{cases}$$





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HAMMING Window

$$w[n] = \begin{cases} 0.54 - 0.46 \cdot \cos\left(\frac{2 \cdot \pi \cdot n}{N}\right), & n = 0, \dots, N - 1 \\ 0, & \text{otherwise} \end{cases}$$

BLACKMANN Window

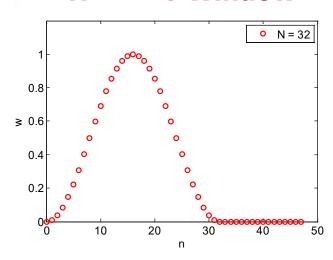
$$w[n] = \begin{cases} 0.42 - 0.5 \cdot \cos\left(\frac{2 \cdot \pi \cdot n}{N}\right) + 0.08 \cdot \cos\left(\frac{4 \cdot \pi \cdot n}{N}\right), & n = 0, \dots, N - 1 \\ 0, & \text{otherwise} \end{cases}$$

GAUSS Window

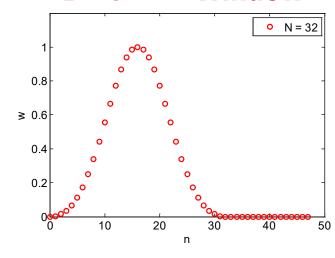
$$w[n] = \begin{cases} e^{-a(n-(N-1)/2)^2}, & n = 0,..., N-1 \\ 0, & \text{otherwise} \end{cases}$$



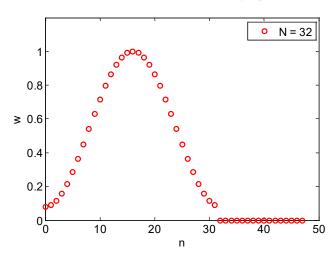
HANNING Window



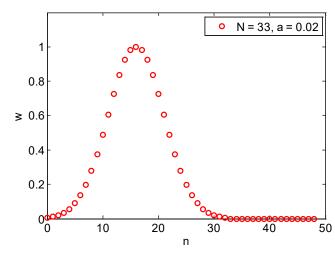
BLACKMAN Window



HAMMING Window



GAUSS Window



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- Adjustable windows
 - Gauss window
 - KAISER window
 - Dolph-Chebyshev window

They allow to adjust the level of the ripple in the stopband.



6.1.2 Properties

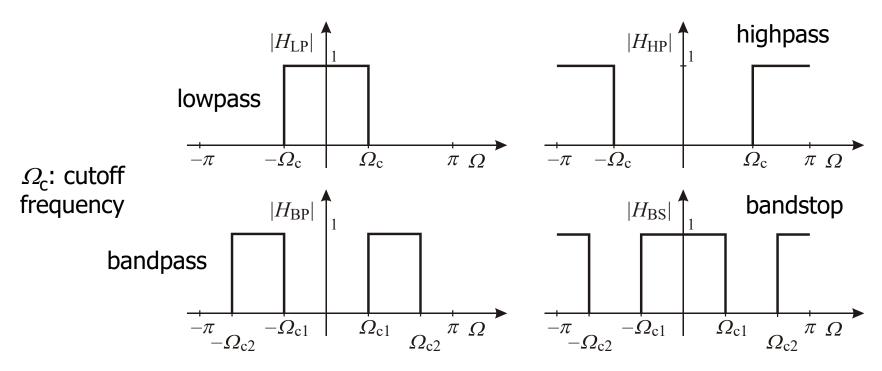
Window	Main Lobe Width	Rel. Sidelobe Level
Rectangular	$4 \cdot \pi/N$	−13.3 dB
BARLETT	$8 \cdot \pi/(N+1)$	−26.5 dB
HANNING	8· π/N	−31.5 dB
HAMMING	8· π/N	−42.7 dB
BLACKMAN	12· π/N	−58.1 dB

- The smooth transition of the window edge is called a taper.
- The Gauss window has the smallest duration-bandwidth product of all window types.
- HANNING, HAMMING, and BLACKMAN are the most popular window types.



6.2 Frequency-Selective Filters

- Ideal frequency-selective filters have a brickwall-type frequency response.
 - Doubly infinite, not causal impulse response.
 - Not realizable.





6.2.1 Simple Digital Filters

 These filters have very simple impulse response coefficients and are employed in a number of practical applications because of their simplicity.

Lowpass FIR digital filters

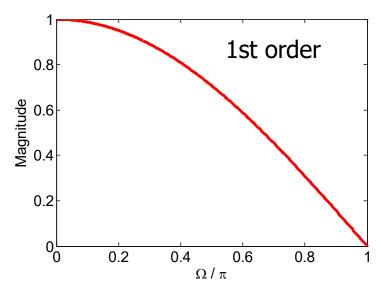
$$h[n] = \frac{1}{2} \cdot (\delta[n] + \delta[n-1]) \circ - \bullet \ H_{LP,1}(z) = \frac{1}{2} \cdot (1 + z^{-1}) = \frac{z+1}{2z}$$

$$H_{LP,1}(e^{j\Omega}) = \cos\left(\frac{\Omega}{2}\right) \cdot e^{-j\Omega/2}$$

$$\xrightarrow{x[n]} H_1(z)$$

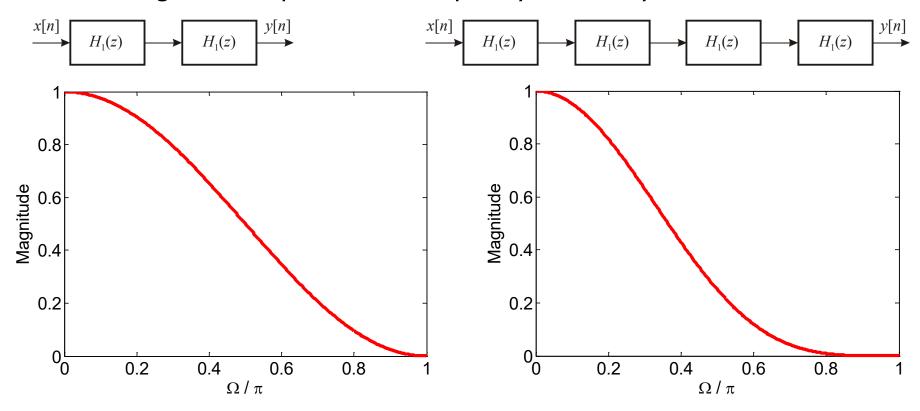
3-dB-cutoff frequency

$$\Omega_{\rm c} = \frac{\pi}{2} = 90 \text{ degree}$$





Cascading filters improves the frequency selectivity.



 Higher-order moving-average filters show a different magnitude response.

$$H_{\text{LP},M-1}(z) = \frac{1}{M} \sum_{i=0}^{M-1} z^{-i}$$



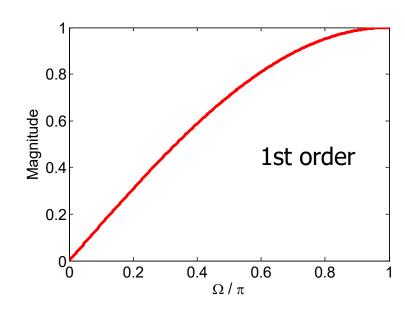
Highpass FIR digital filters

$$h[n] = \frac{1}{2} \cdot (\delta[n] - \delta[n-1]) \quad \circ - \bullet \quad H_{HP,1}(z) = \frac{1}{2} \cdot (1 - z^{-1}) = \frac{z - 1}{2z}$$

$$H_{\text{HP},1}\left(e^{j\Omega}\right) = \sin\left(\frac{\Omega}{2}\right) \cdot e^{j(\pi-\Omega)/2}$$

3-dB cutoff frequency

$$\Omega_{\rm c} = \frac{\pi}{2} = 90$$
 degree



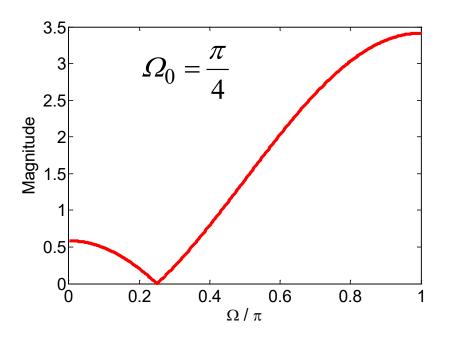
Higher-order filter

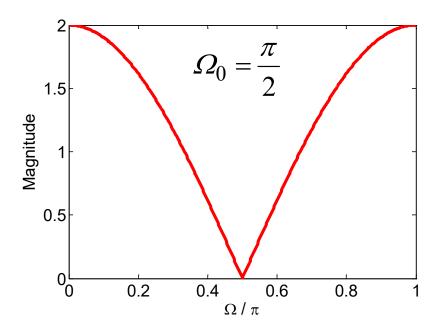
$$H_{\text{HP},M-1}(z) = \frac{1}{M} \sum_{i=0}^{M-1} (-1)^{i} \cdot z^{-i}$$



• FIR notch digital filters are used to suppress a particular sinusoidal component at Ω_0 .

$$H_{\text{Notch}}(z) = 1 - 2 \cdot \cos(\Omega_0) \cdot z^{-1} + z^{-2}$$
$$H_{\text{Notch}}(e^{j\Omega}) = 1 - e^{j(\Omega_0 - \Omega)} - e^{-j(\Omega_0 + \Omega)} + e^{-j2\Omega}$$

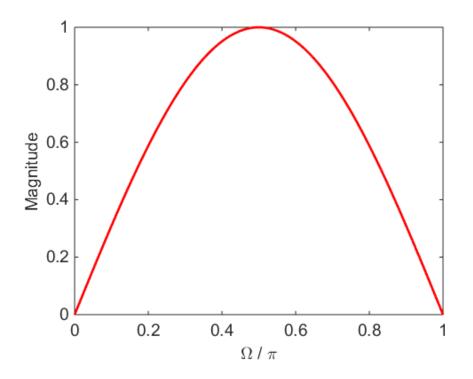


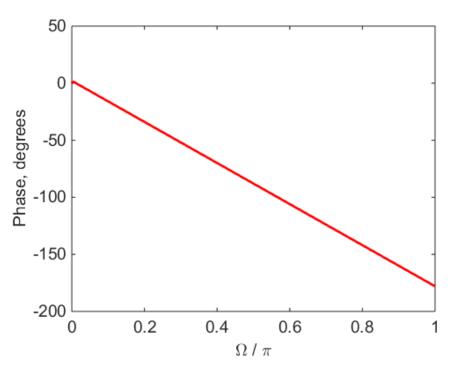




FIR bandpass digital filters ... are not popular

$$H_{\text{BP,2}}(z) = \frac{1}{2} \cdot (1 - z^{-2}) = \frac{1}{2} \cdot \frac{z^2 - 1}{z^2}$$







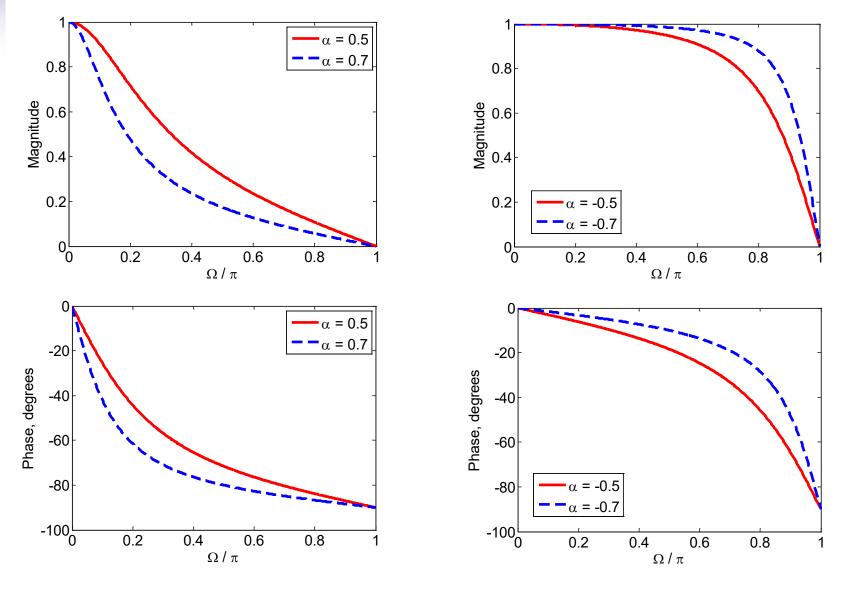
- All FIR digital filters have their poles at the origin. → Only the zeros determine the filter characteristic.
- The filter characteristic of IIR digital filters are determined by zeros and poles.

Lowpass IIR digital filters

$$H_{\text{LP},1}(z) = \frac{1-\alpha}{2} \cdot \frac{1+z^{-1}}{1-\alpha \cdot z^{-1}} = \frac{1-\alpha}{2} \cdot \frac{z+1}{z-\alpha}$$

$$H_{\text{LP},1}(e^{j\Omega}) = \frac{1-\alpha}{2} \cdot \frac{e^{j\Omega}+1}{e^{j\Omega}-\alpha}$$





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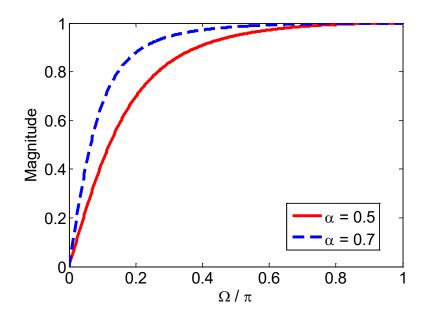


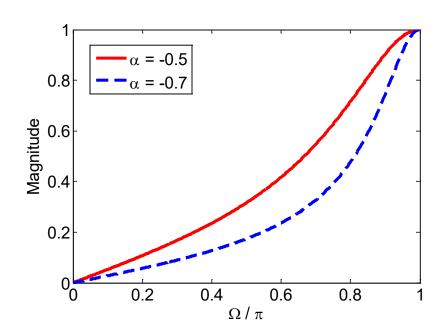
Highpass IIR digital filters

$$H_{\text{HP,1}}(z) = \frac{1+\alpha}{2} \cdot \frac{1-z^{-1}}{1-\alpha \cdot z^{-1}} = \frac{1+\alpha}{2} \cdot \frac{z-1}{z-\alpha}$$

$$H_{\text{HP,1}}(e^{j\Omega}) = \frac{1+\alpha}{2} \cdot \frac{e^{j\Omega}-1}{e^{j\Omega}-\alpha}$$

$$H_{\text{HP},1}\left(e^{j\Omega}\right) = \frac{1+\alpha}{2} \cdot \frac{e^{j\Omega}-1}{e^{j\Omega}-\alpha}$$

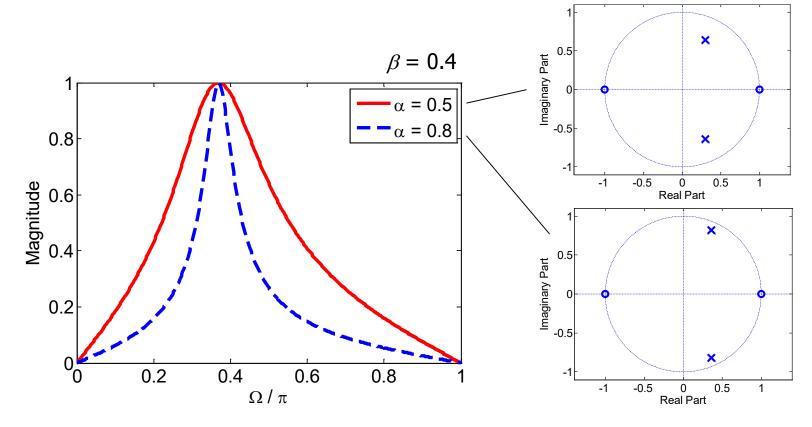






Bandpass IIR digital filters

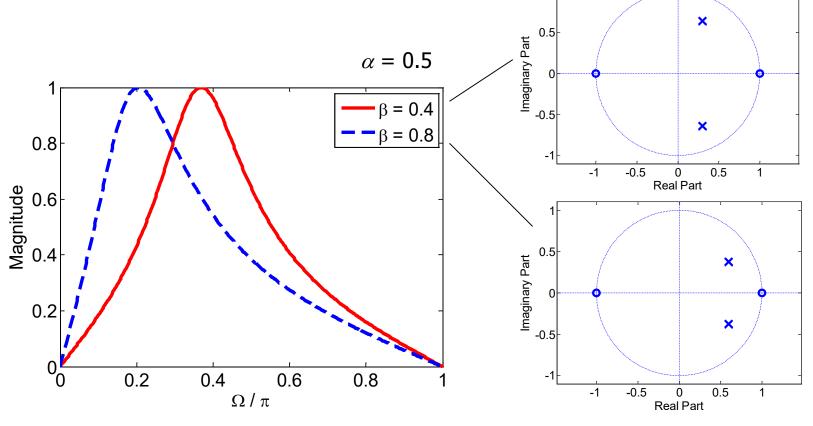
$$H_{\mathrm{BP},2}(z) = \frac{1 - \alpha}{2} \cdot \frac{1 - z^{-2}}{1 - \beta \cdot (1 + \alpha) \cdot z^{-1} + \alpha \cdot z^{-2}} = \frac{1 - \alpha}{2} \cdot \frac{z^{2} - 1}{z^{2} - \beta \cdot (1 + \alpha) \cdot z + \alpha}$$





Poles:
$$p_{1/2} = r \cdot e^{\pm j\phi}$$
; $r = \sqrt{\alpha}$; $\cos \phi = \frac{\beta \cdot (1+\alpha)}{2 \cdot \sqrt{\alpha}}$

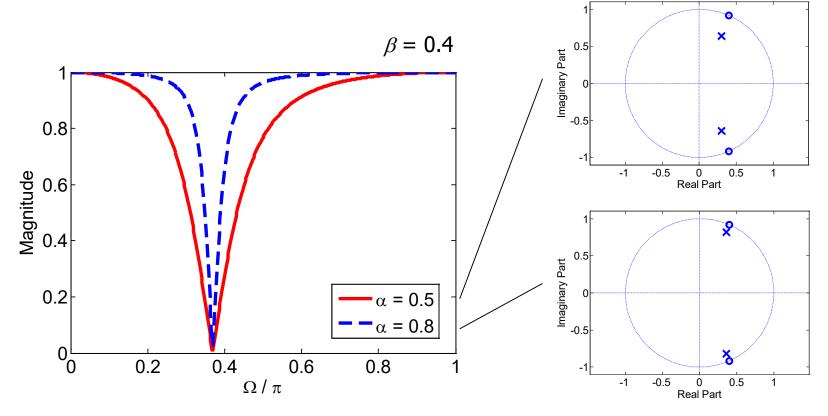
• Maximum: $\Omega = \Omega_0$ with $\beta = \cos \Omega_0$





IIR notch digital filters

$$H_{\text{BP},2}(z) = \frac{1+\alpha}{2} \cdot \frac{1-2 \cdot \beta \cdot z^{-1} + z^{-2}}{1-\beta \cdot (1+\alpha) \cdot z^{-1} + \alpha \cdot z^{-2}} = \frac{1-\alpha}{2} \cdot \frac{z^2 - 2 \cdot \beta \cdot z + 1}{z^2 - \beta \cdot (1+\alpha) \cdot z + \alpha}$$

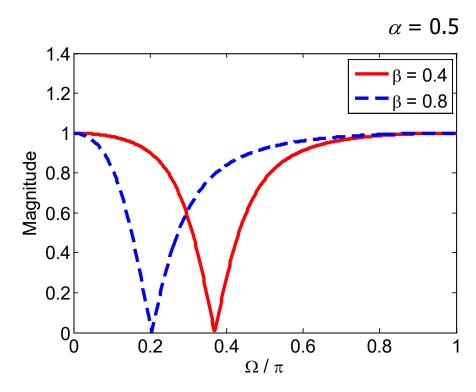


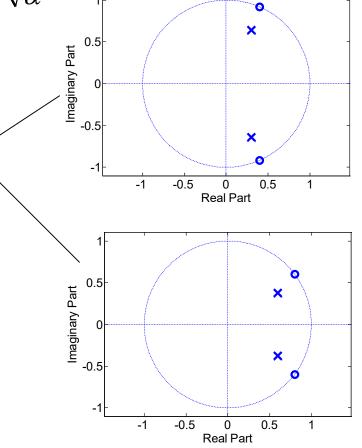
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- Zeros: $z_{1/2} = r \cdot e^{\pm j\phi}$; r = 1; $\cos \phi = \beta$
- Poles: $p_{1/2} = r \cdot e^{\pm j\phi}$; $r = \sqrt{\alpha}$; $\cos \phi = \frac{\beta \cdot (1+\alpha)}{2 \cdot \sqrt{\alpha}}$
- Notch: $\Omega = \Omega_0$ with $\beta = \cos \Omega_0$





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6.2.2 FIR Digital Filter Design6.2.2.1 Linear-Phase FIR Transfer Functions

Extended Definition

$$H(e^{j\Omega}) = \widetilde{H}(\Omega) \cdot e^{j\varphi}$$
 with $\varphi = a \cdot \Omega + b$
$$\widetilde{H}(\Omega)$$
: Zero-phase response
$$\text{phase delay } \tau_{\mathrm{p}} = -\frac{\varphi}{O} = -a - \frac{b}{O} \qquad \text{group delay } \tau_{\mathrm{g}} = -\frac{\mathrm{d}\,\varphi}{\mathrm{d}\,O} = -a$$

• $H(z) = \sum_{n=0}^{N} h[n] \cdot z^{-n}$ with h[n] real.

Order N of transfer function can be even or odd.



• $\widetilde{H}(\Omega)$ even requires a symmetric impulse response.

$$h[n] = h[N-n]$$
 with $0 \le n \le N$; $b = 0$ or π

 $\widetilde{H}(\Omega)$ odd requires an antisymmetric impulse response.

$$h[n] = -h[N-n]$$
 with $0 \le n \le N$; $b = \pm \frac{\pi}{2}$

General form of frequency response

$$H(e^{j\Omega}) = \widetilde{H}(\Omega) \cdot e^{jb} \cdot e^{-jN\Omega/2}$$

$$\tau_{g} = \frac{N}{2}$$



■ **Type I:** Symmetric impulse response, *N* even

$$\widetilde{H}(\Omega) = h \left[\frac{N}{2}\right] + 2 \cdot \sum_{n=1}^{N/2} h \left[\frac{N}{2} - n\right] \cdot \cos(n \cdot \Omega)$$

■ **Type II:** Symmetric impulse response, *N* odd

$$\widetilde{H}(\Omega) = 2 \cdot \sum_{n=1}^{(N+1)/2} h \left[\frac{N+1}{2} - n \right] \cdot \cos \left(\left(n - \frac{1}{2} \right) \cdot \Omega \right)$$



Type III: Antisymmetric impulse response, N even

$$\widetilde{H}(\Omega) = 2 \cdot \sum_{n=1}^{N/2} h \left[\frac{N}{2} - n \right] \cdot \sin(n \cdot \Omega)$$

Type IV: Antisymmetric impulse response, N odd

$$\widetilde{H}(\Omega) = 2 \cdot \sum_{n=1}^{(N+1)/2} h \left[\frac{N+1}{2} - n \right] \cdot \sin \left(\left(n - \frac{1}{2} \right) \cdot \Omega \right)$$



6.2.2.2 Zero Locations of Linear-Phase FIR Filter

Type I and II

$$H(z) = z^{-N} \cdot H(z^{-1})$$
 \rightarrow mirror-image polynomial (MIP)

Type III and IV

$$H(z) = -z^{-N} \cdot H(z^{-1}) \rightarrow \text{antimirror-image polynomial (AIP)}$$



Requirements for zeros

- Real zero ($z \neq \pm 1$): z = r and z = 1/r (2 zeros)
- Zero on unit circle ($z \neq \pm 1$): $z = e^{j\phi}$ and $z = e^{-j\phi}$ (2 zeros)
- Complex zero ($|z| \neq 1$, not real): $z = r \cdot e^{\pm j\phi}$ and $z = \frac{1}{r} \cdot e^{\pm j\phi}$ (4 zeros)

Additional zero requirements

- Filter type I: None or even number at $z = \pm 1$.
- Filter type II: Odd number at z = -1. None or even number at z = 1. → no highpass
- Filter type III: Odd number at z = 1 and at z = -1. \rightarrow only bandpass
- Filter type IV: Odd number at z = 1. None or even number at z = -1. → no lowpass



6.2.3 FIR Filter Order

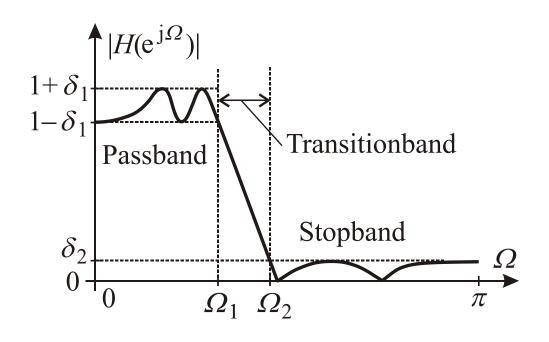
Kaiser's formula

$$N \approx \frac{-20 \cdot \log_{10} \left(\sqrt{\delta_1 \cdot \delta_2} \right) - 13}{14.6 \cdot \left(\Omega_2 - \Omega_1 \right) / (2\pi)} \qquad \begin{array}{c} 1 + \delta_1 \\ 1 - \delta_1 \end{array}$$
Passb

Bellanger's formula

$$N \approx \frac{-2 \cdot \log_{10} (10 \cdot \delta_1 \cdot \delta_2)}{3 \cdot (\Omega_2 - \Omega_1)/(2\pi)} - 1$$

 The formulas can also be used for highpass, bandpass, and bandstop filters. Dominant is the smallest transition band.



Typical magnitude specifications [MIT]

 Ω_1 , Ω_2 : passband, stopband edge frequencies δ_1 , δ_2 : peak ripple values

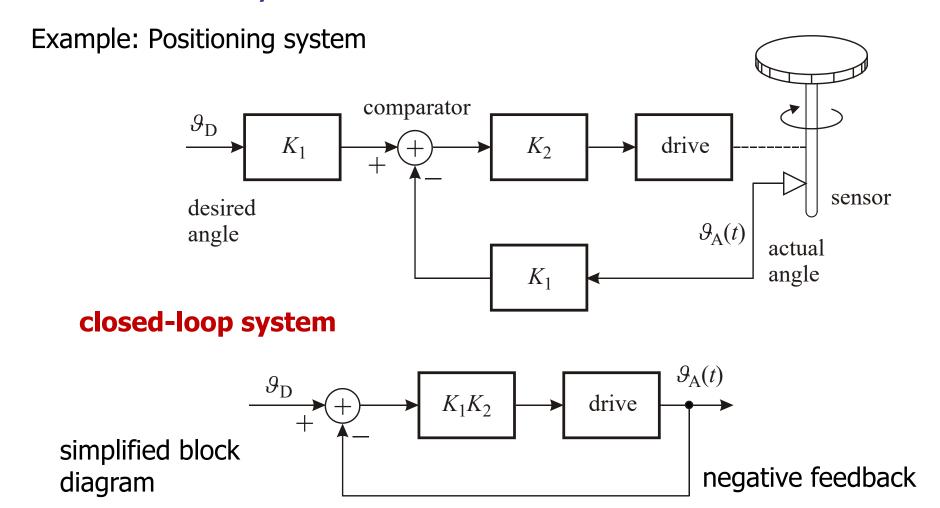


6.2.4 Design Concepts

- Contrary to IIR filters, the design of FIR filters doesn't has any relations to the design of analog filters.
- The order of the FIR transfer functions is usually much higher compared to IIR filters.
- Design goal
 - Linear-phase design for constant group delay.
 - Minimum-phase design for minimal group delay.
 - Implementation efficient approaches with minimum number of multipliers.
- Design approach
 - Frequency sampling approach.
 - Windowed Fourier series approach.



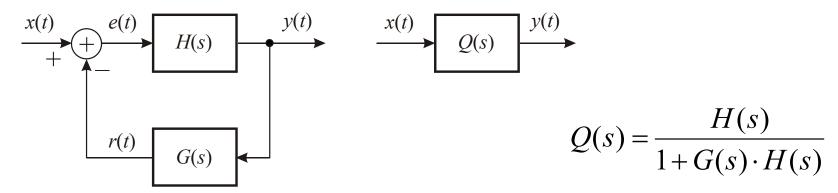
6.3 Control Systems



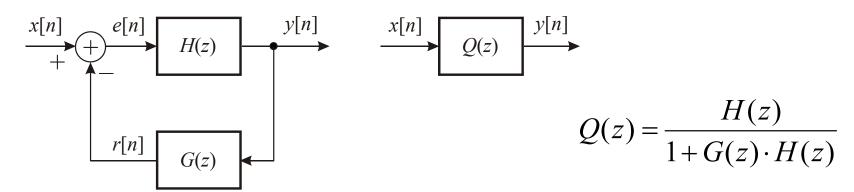


6.3.1 Linear Feedback Systems

Continuous-time LTI feedback system



Discrete-time LTI feedback system





- H(s), H(z): System function of the forward path
- G(s), G(z): System function of the feedback path
- Q(s), Q(z): Closed-loop system function

Open-loop system

- $G(s) \cdot H(s)$
- $G(z) \cdot H(z)$



6.3.2 Stabilization of Unstable Systems

- One use of feedback is to stabilize systems, which are unstable without feedback.
- Continuous-time system

$$H(s) = \frac{b}{s-a}$$
; $G(s) = K$; a,b,K real

$$r(t) = K \cdot y(t)$$
 \rightarrow proportional feedback system

Feedback system is stable for $K > \frac{a}{b}$ with b > 0 or $K < \frac{a}{b}$ with b < 0



Discrete-time system: population growth

$$y[n] = 2 \cdot y[n-1] + e[n] \leftarrow \text{unstable}$$

y[n]: population at generation n

e[n]: external influences (migration, disease, ...)

$$e[n] = x[n] - r[n]; r[n] = 2 \cdot \beta \cdot y[n-1] \leftarrow \text{feedback approach}$$

r[n]: regulative influences; β real

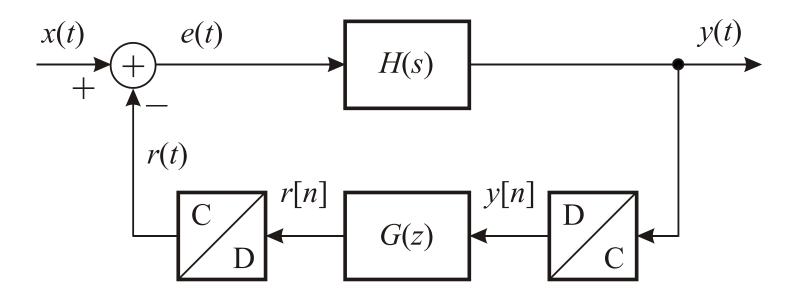
x[n]: external effects (migration, disease, ...)

Feedback system is stable for $\frac{1}{2} < \beta < \frac{3}{2}$



6.3.3 Sampled-Data Feedback Systems

 Discrete-time feedback systems apply discrete processing in the feedback path and consider a real-world continuous-time system in the forward path

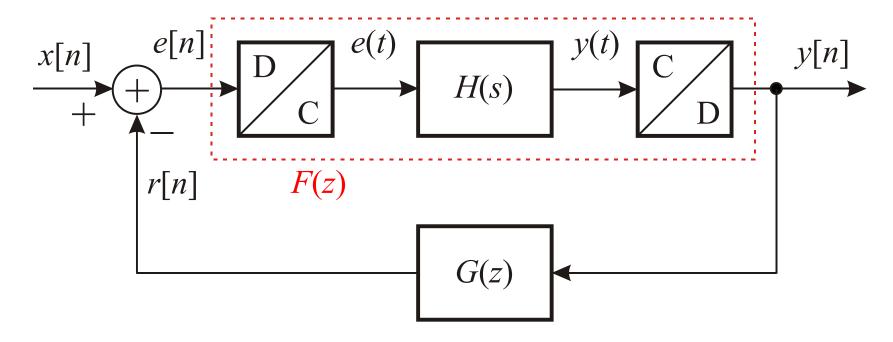




Assumption

$$x(t) = x[n]$$
 for $n \cdot T_s \le t < (n+1) \cdot T_s$

Resultant is a completely discrete-time feedback system.

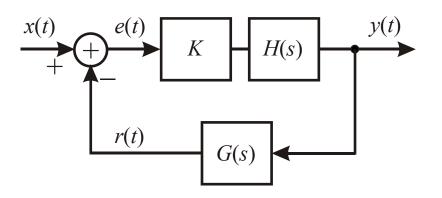




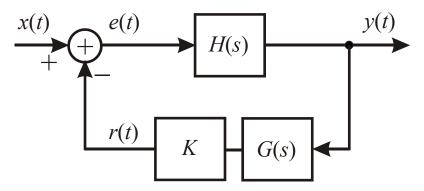
6.3.4 Root-Locus Analysis

The feedback systems are supplemented by an adjustable gain element.

Continuous-time systems



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



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

Discrete-time systems

$$Q(z) = \frac{K \cdot H(z)}{1 + K \cdot G(z) \cdot H(z)}$$

$$Q(z) = \frac{H(z)}{1 + K \cdot G(z) \cdot H(z)}$$



Closed-loop poles

$$G(...) \cdot H(...) = -\frac{1}{K}$$
 with K real, '...'= s or z

- $K \to 0 \Longrightarrow G(...) \cdot H(...) \to -\infty \leftarrow \text{poles of } G(...) \cdot H(...)$
- $K \to \pm \infty \Longrightarrow G(...) \cdot H(...) \to 0 \leftarrow \text{zeros of } G(...) \cdot H(...)$
- $arg\{G(...) \cdot H(...)\} = n \cdot \pi$; n: integer
 - $arg\{G(...) \cdot H(...)\} = k \cdot \pi ; k : odd$

$$K = \frac{1}{|G(\dots) \cdot H(\dots)|} > 0$$

• $arg\{G(...) \cdot H(...)\} = k \cdot \pi ; k$: even

$$K = \frac{-1}{|G(\ldots) \cdot H(\ldots)|} < 0$$



The **root-locus curve** is a plot of the closed-loop poles with varying gain element.

The root-locus curve

- starts at zeros of $G(...) \cdot H(...)$ for $K \to -\infty$,
- equals poles of $G(...) \cdot H(...)$ for K = 0,
- ends at zeros of $G(...) \cdot H(...)$ for $K \to \infty$.
- Branches between two real poles must break off into the complex plane for |K| > 0 .



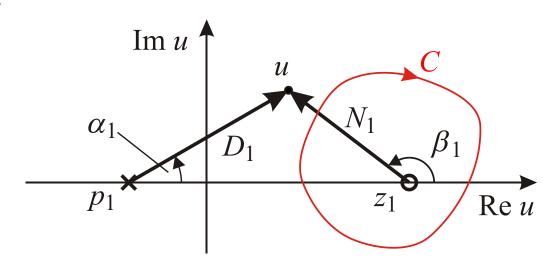
6.3.5 NYQUIST Stability Criterion

- The root—locus analysis requires the analytic description of the forward and feedback system functions in a rational form.
- The Nyquist criterion
 - determines whether or not the closed-loop system is stable for a given value of the adjustable gain parameter K,
 - can be applied to nonrational system functions and in situations in which to analytic descriptions of the system functions are available.



Encirclement Property

$$W(u) = \frac{u - z_1}{u - p_1} = \frac{N_1 \cdot e^{j\beta_1}}{D_1 \cdot e^{j\alpha_1}}$$



If u traverses along a closed contour \mathcal{C} in the u-plane <u>once</u> in the clockwise direction,

the corresponding plot of W(u) will encircle the origin M-N times in the clockwise direction, where

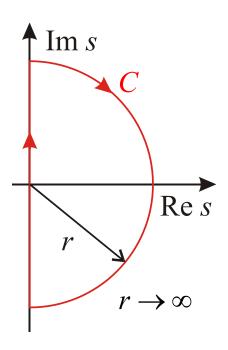
M is the number of zeros within C and

N is the number of poles within C.



Continuous-Time NYQUIST Stability Criterion

- Requirement for stability: No zero of $W(s) = \frac{1}{K} + H(s) \cdot G(s)$ lies in the right half of the *s*-plane.
- The plot of $H(j\omega) \cdot G(j\omega)$ as ω varies from $-\infty$ to $+\infty$ is called the NYQUIST plot.
- The number of clockwise encirclements of the point -1/K by the NYQUIST plot equals the number of right-half plane closed-loop poles minus the number of right-half plane poles of $H(s) \cdot G(s)$.

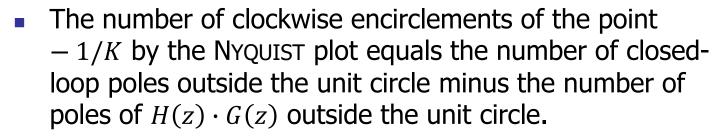


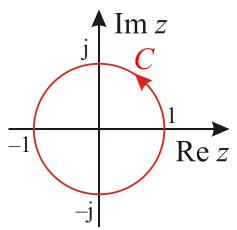
■ **Requirement for stability:** The number of <u>counterclockwise</u> encirclements of the point -1/K by the NYQUIST plot of $H(j\omega) \cdot G(j\omega)$ must equal the number of right-half-plane poles of $H(s) \cdot G(s)$.



Discrete-Time Nyquist Stability Criterion

- Requirement for stability: No zero of $W(z) = \frac{1}{K} + H(z) \cdot G(z)$ lies outside the unit circle of the z-plane.
- The plot of $H(e^{j\Omega}) \cdot G(e^{j\Omega})$ as Ω varies from 0 to 2π is called the NYQUIST plot.





■ Requirement for stability: The number of counterclockwise encirclements of the point -1/K by the NYQUIST plot of $H(e^{j\Omega}) \cdot G(e^{j\Omega})$ as Ω varies from 0 to 2π must equal the number of poles of $H(z) \cdot G(z)$ outside the unit circle.