# Chapter 6

Applications

- **6.1 Time-Selective Filters**
  - Window Functions
  - Properties

- **6.2 Frequency-Selective Filters**
  - Simple Digital Filters,
  - FIR Digital Filter Design, Filter Order
  - Design Concepts

- **6.3 Control Systems**
  - Linear Feedback Systems
  - Stabilization of Unstable Systems
  - Sampled-Data Feedback Systems
  - Root-Locus Analysis, NYQUIST Stability Criterion

- Time-discrete processing of time-continuous signals
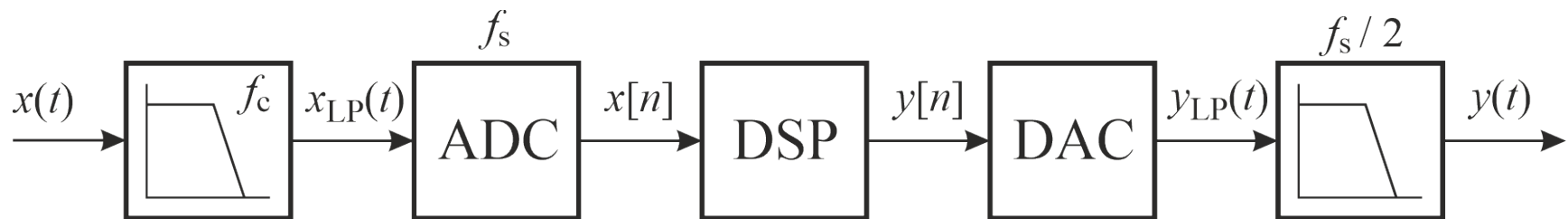
6.1 Time-Selective Filters

- 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
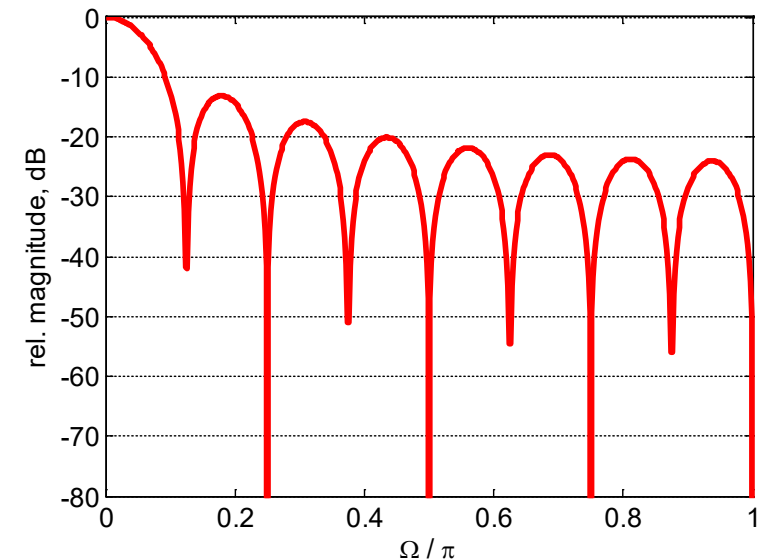   - Window type $\rightarrow$ stopband ripple
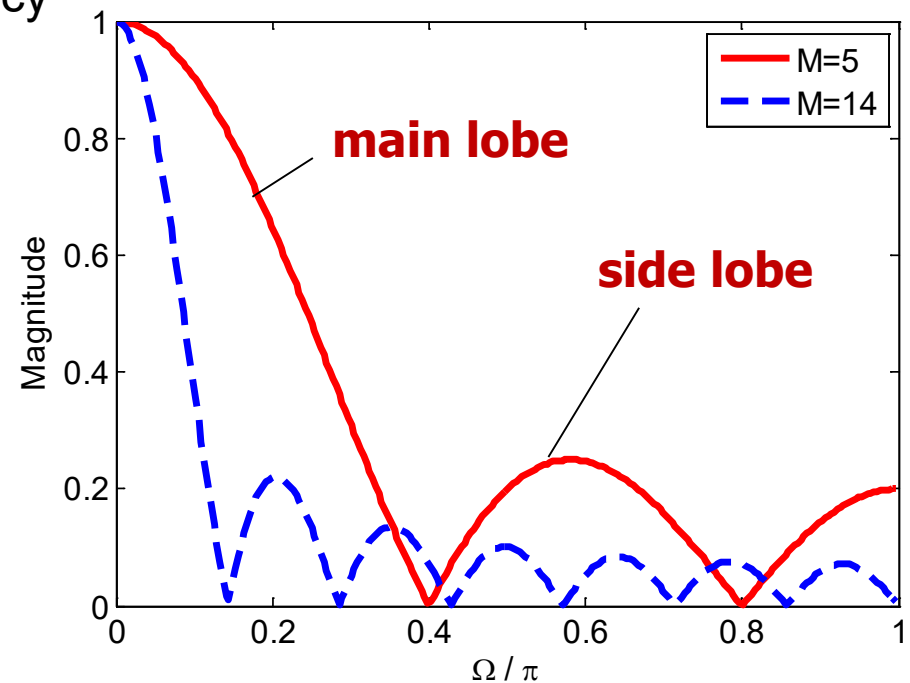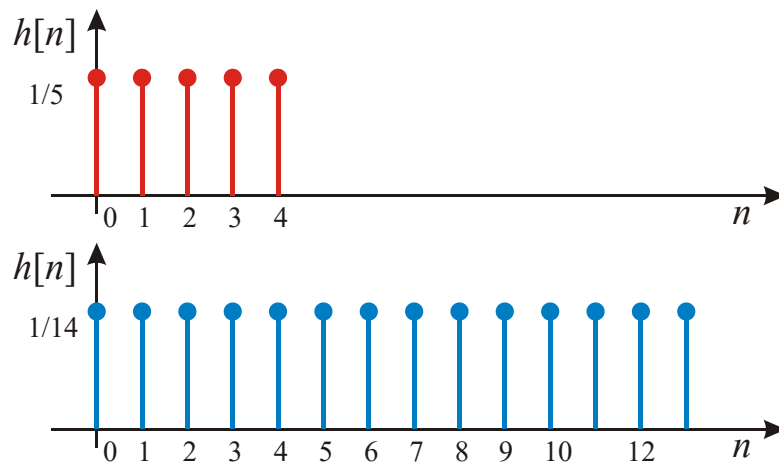
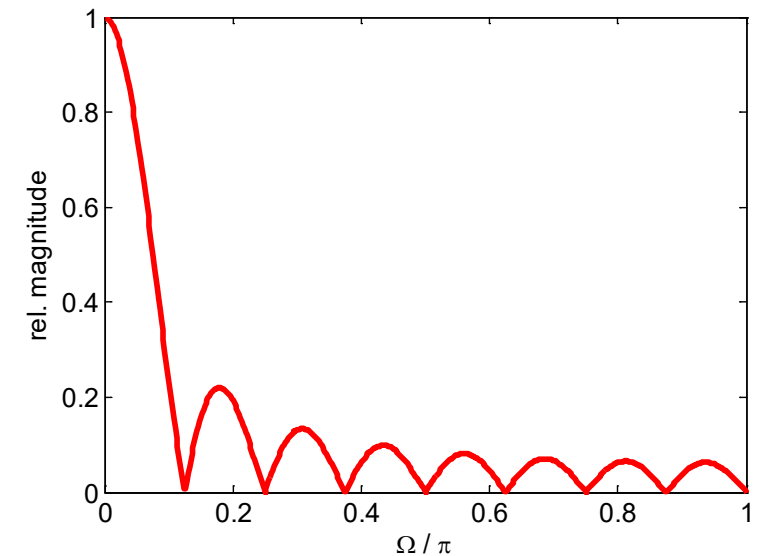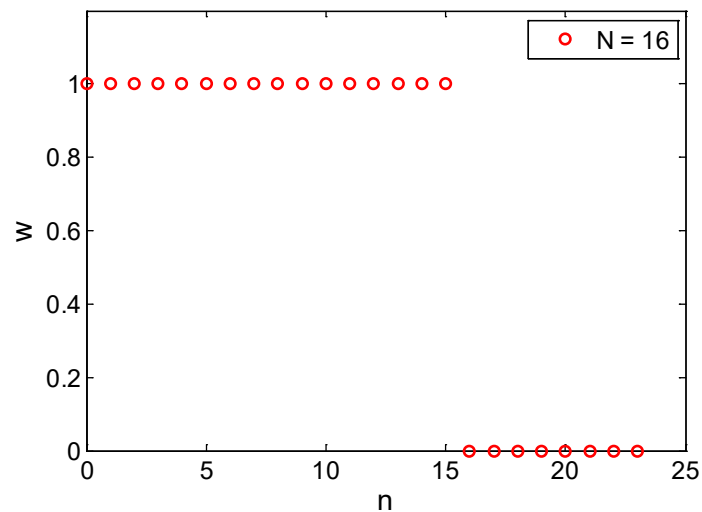# 6.1.1 Window Functions

Rectangular Window

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

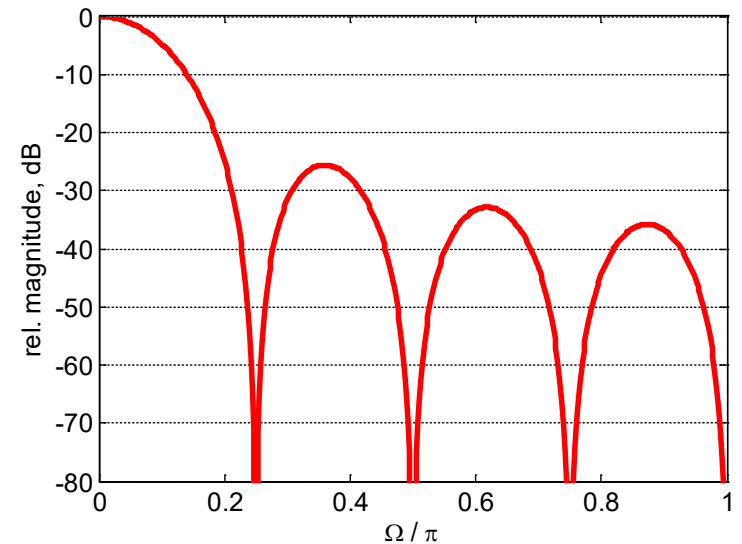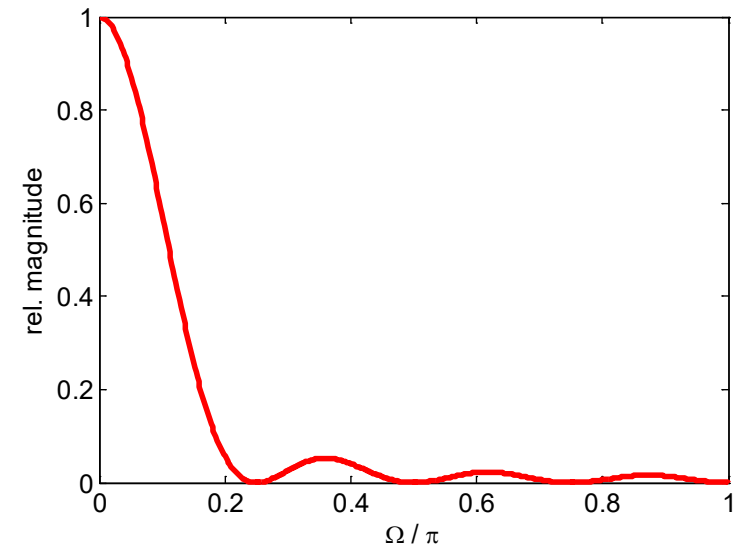$$W\left(e^{j\Omega}\right) = \frac{\sin(\Omega \cdot N/2)}{\sin(\Omega/2)} \cdot e^{-j\Omega \cdot (N-1)/2}$$

BARLETT Window - triangular window

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

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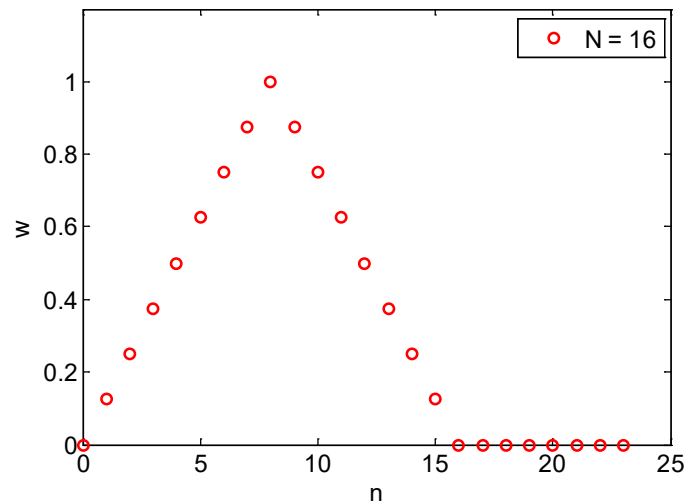
HANNING Window

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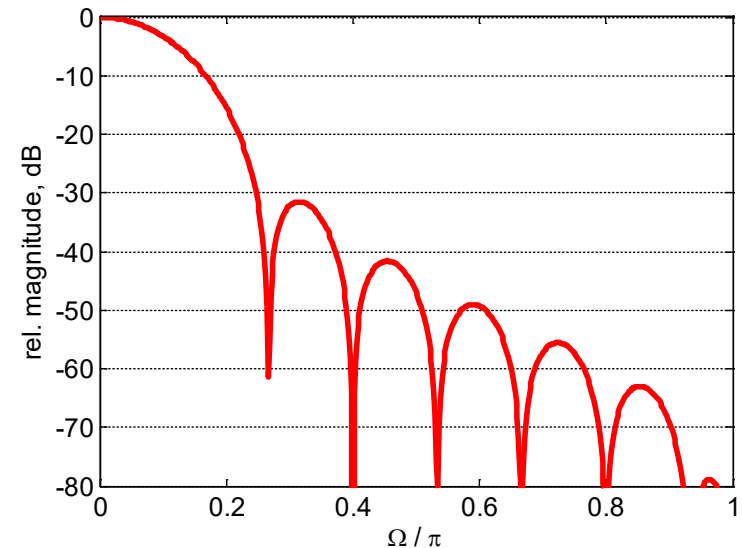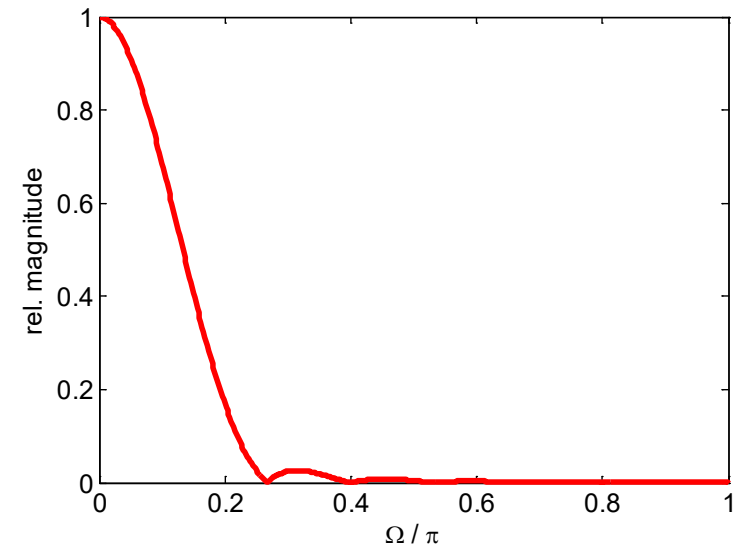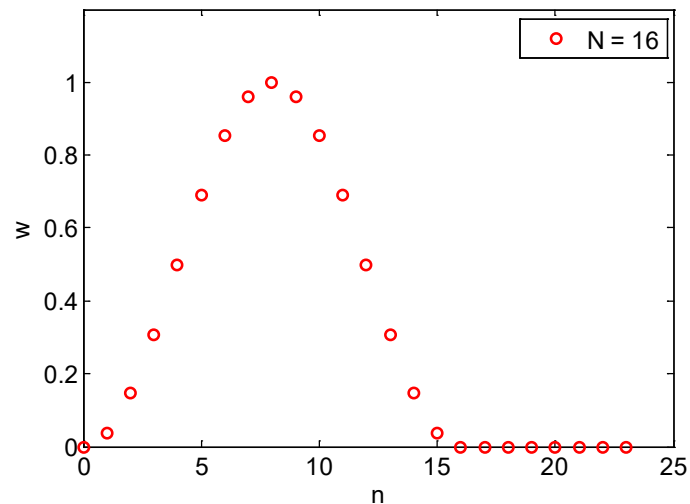
HAMMING Window

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

BLACKMANN Window

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

GAUSS Window

$$w[n] = \begin{cases} e^{-a\left(n-(N-1)/2\right)^2} & , \ n = 0, \ldots, N-1 \\ 0 & , \ \text{otherwise} \end{cases}$$
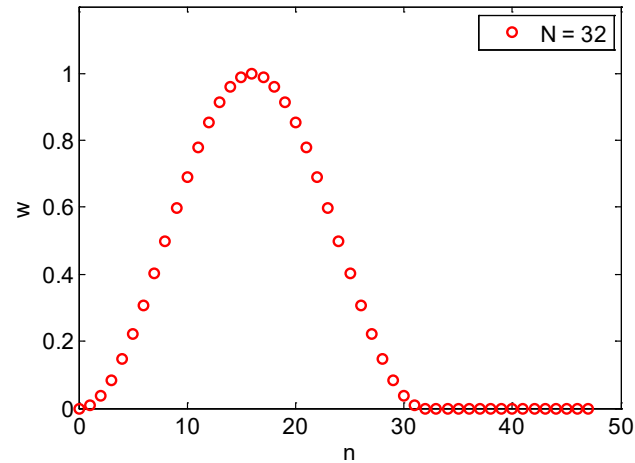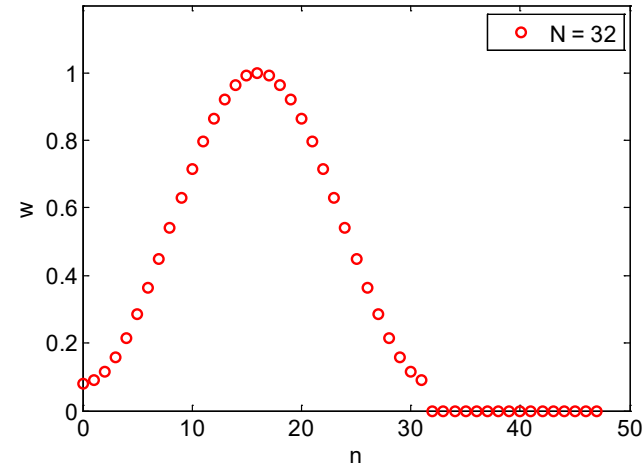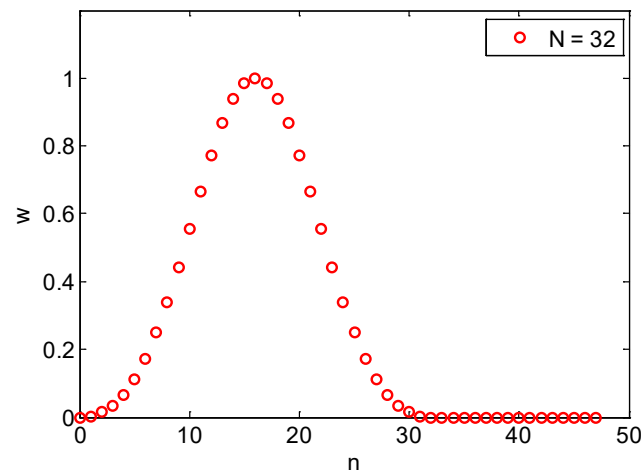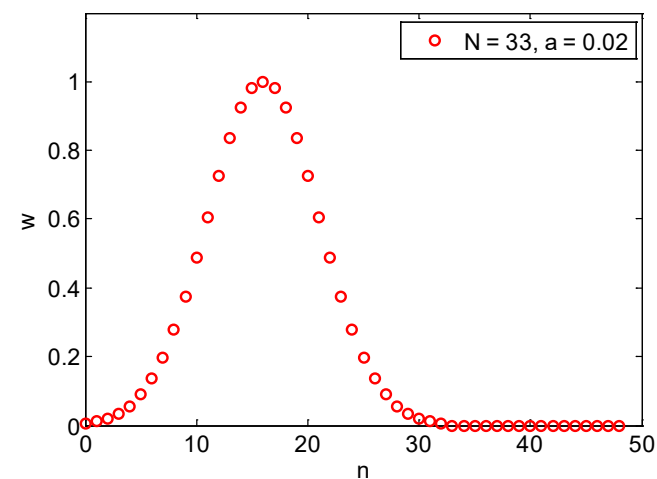
- **Adjustable windows**
  - GAUSS window
  - KAISER window
  - DOLPH-CHEBYSHEV window

  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	$8 \cdot \pi/N$	−31.5 dB
HAMMING	$8 \cdot \pi/N$	−42.7 dB
BLACKMAN	$12 \cdot \pi/N$	−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.

Ω_c: cutoff frequency
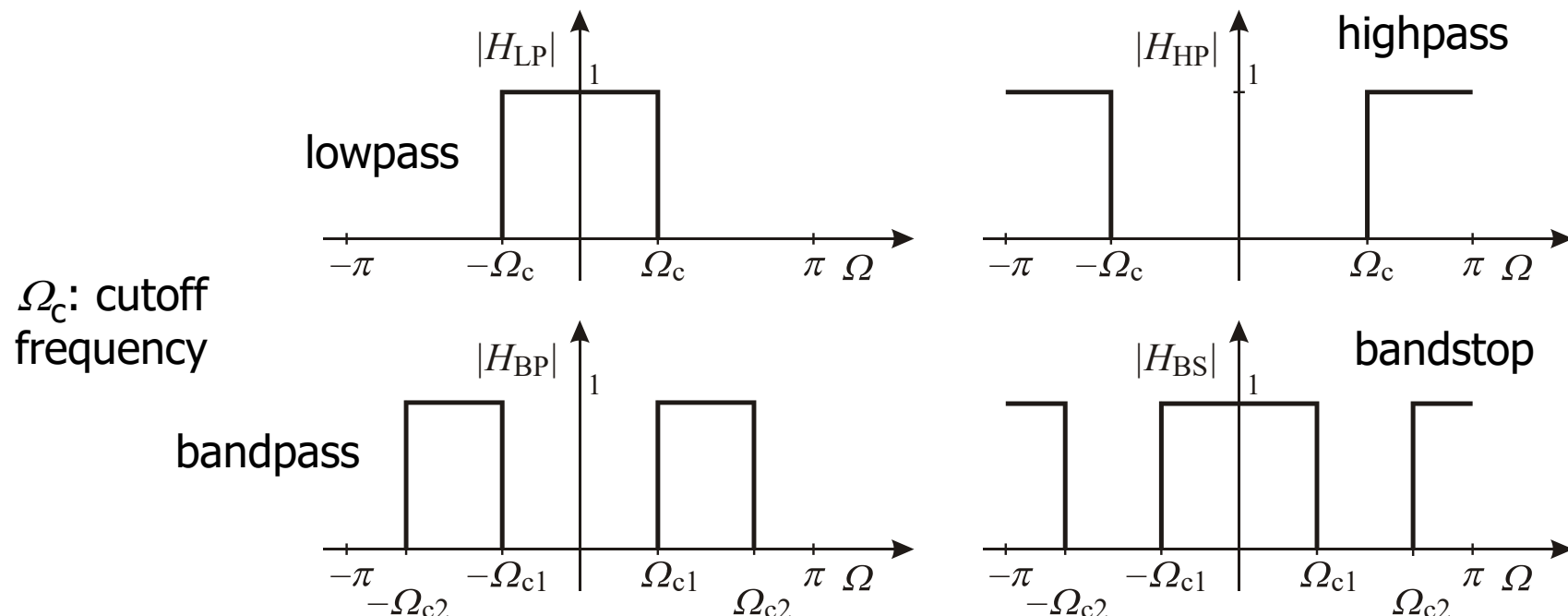
6.2.1 Simple Digital Filters

- 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**

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

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

$x[n] \longrightarrow \boxed{H_1(z)} \longrightarrow y[n]$

$3\text{-dB-cutoff frequency}$

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

1st order
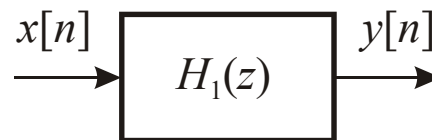
Magnitude vs Ω/π

- Cascading filters improves the frequency selectivity.
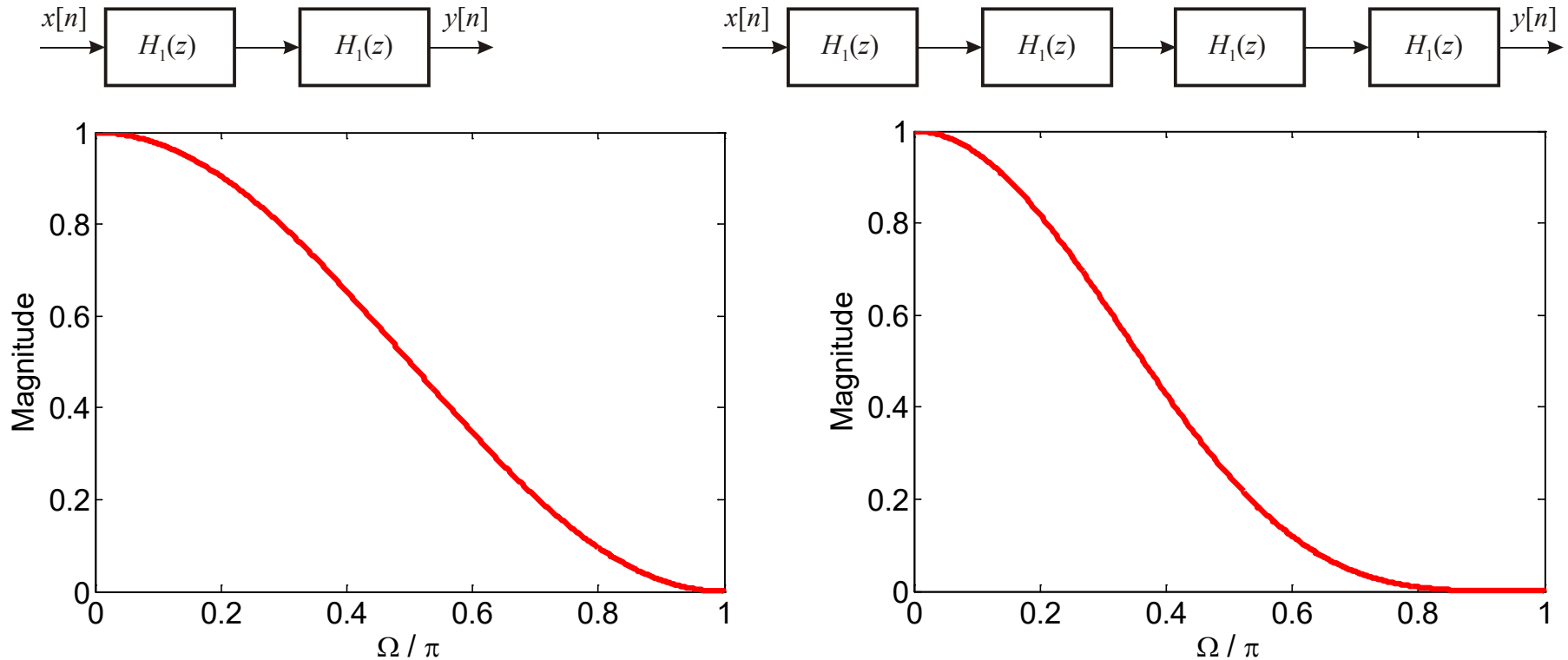
- Higher-order moving-average filters show a different magnitude response.

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

- **Highpass FIR digital filters**

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

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

3 - dB cutoff frequency

$$\Omega_{\mathrm{c}} = \frac{\pi}{2} = 90 \text{ degree}$$
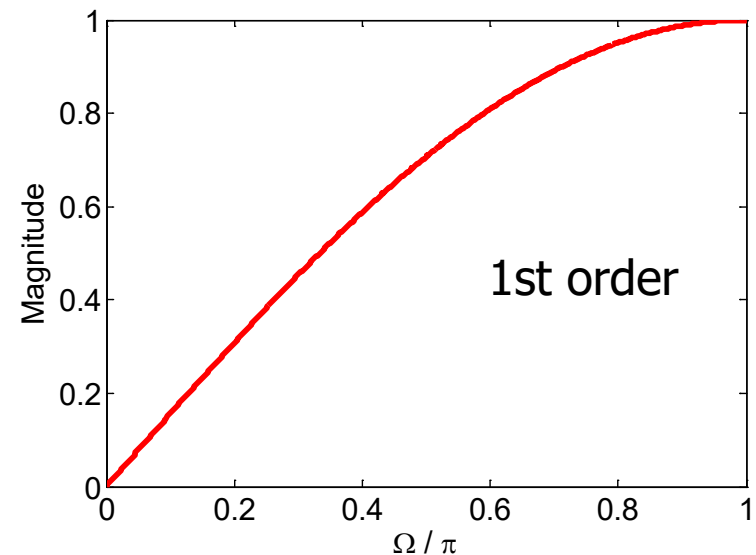
- Higher-order filter

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

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

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

$$H_{\text{Notch}}\left(e^{j\Omega}\right) = 1 - e^{j(\Omega_0 - \Omega)} - e^{-j(\Omega_0 + \Omega)} + e^{-j2\Omega}$$
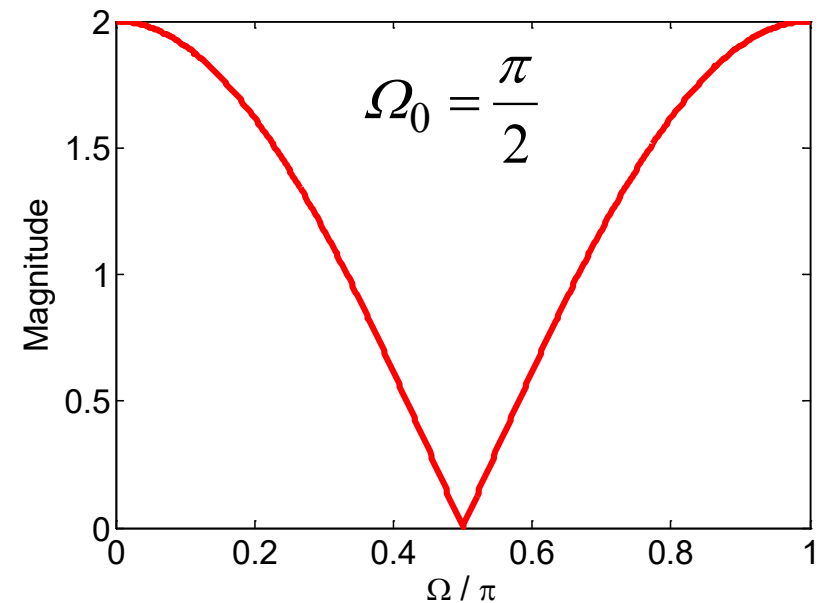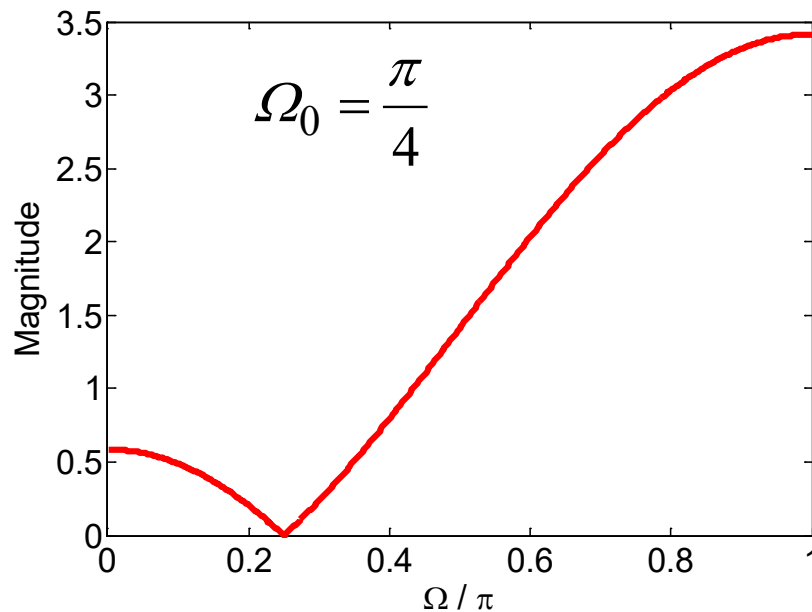
- **FIR bandpass digital filters** ... are not popular

$$H_{BP,2}(z) = \frac{1}{2} \cdot \left(1 - z^{-2}\right) = \frac{1}{2} \cdot \frac{z^2 - 1}{z^2}$$
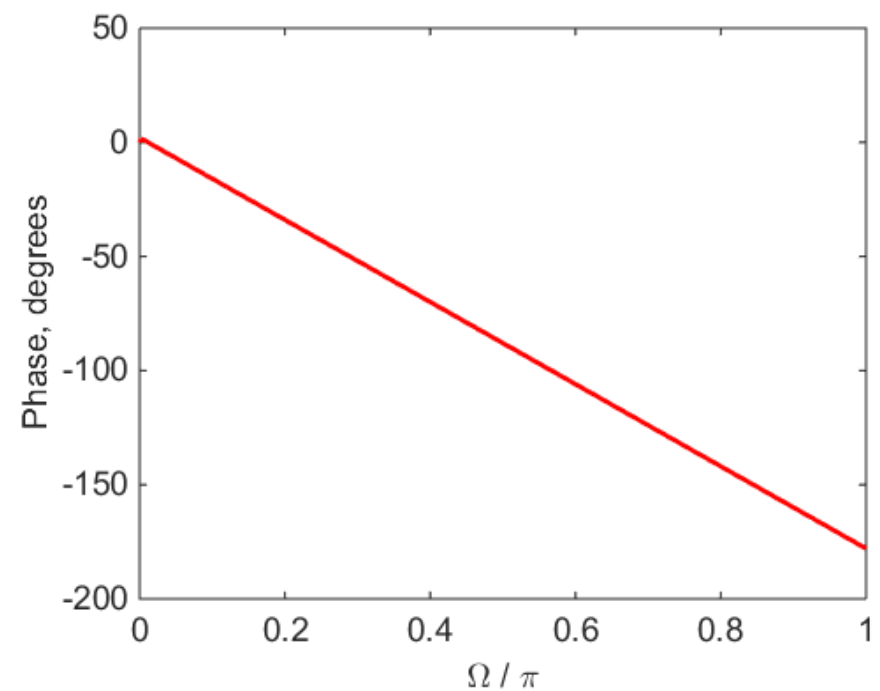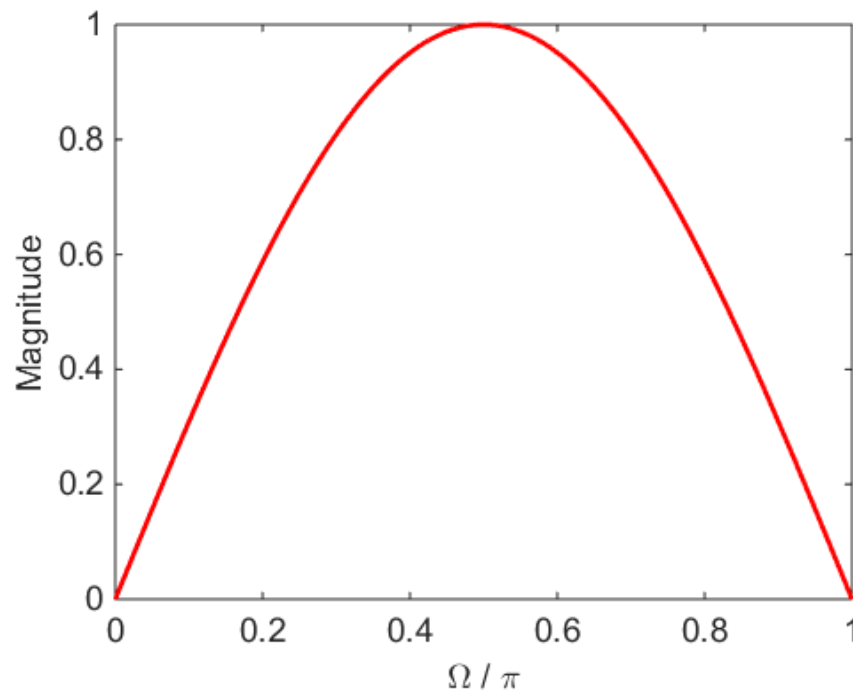
- All FIR digital filters have their poles at the origin. → Only the zeros determine the filter characteristic.

- The filter characteristic of IIR digital filters are determined by zeros and poles.

- **Lowpass IIR digital filters**

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

$$H_{\mathrm{LP},1}\left(\mathrm{e}^{\mathrm{j}\Omega}\right) = \frac{1-\alpha}{2} \cdot \frac{\mathrm{e}^{\mathrm{j}\Omega}+1}{\mathrm{e}^{\mathrm{j}\Omega}-\alpha}$$
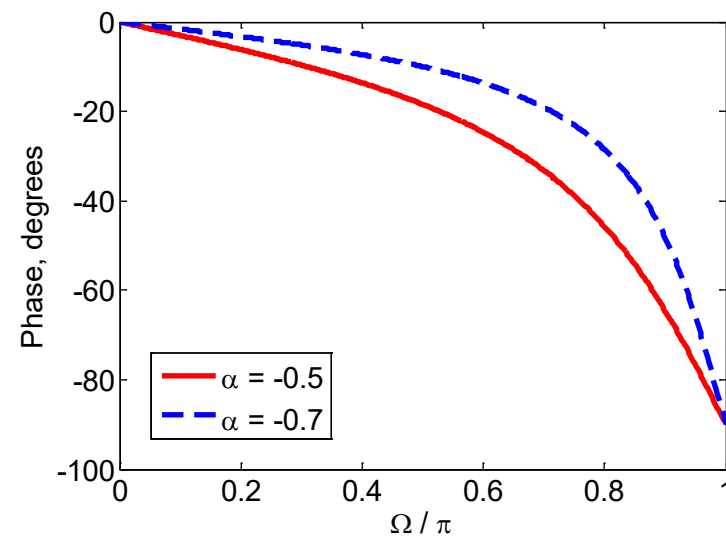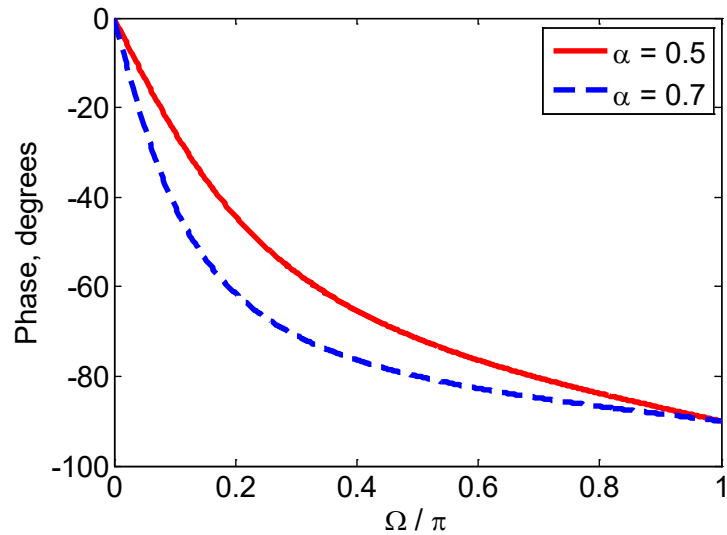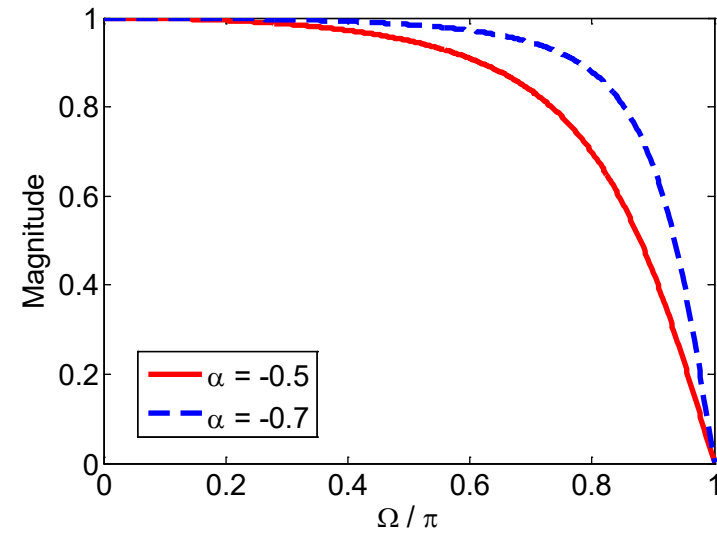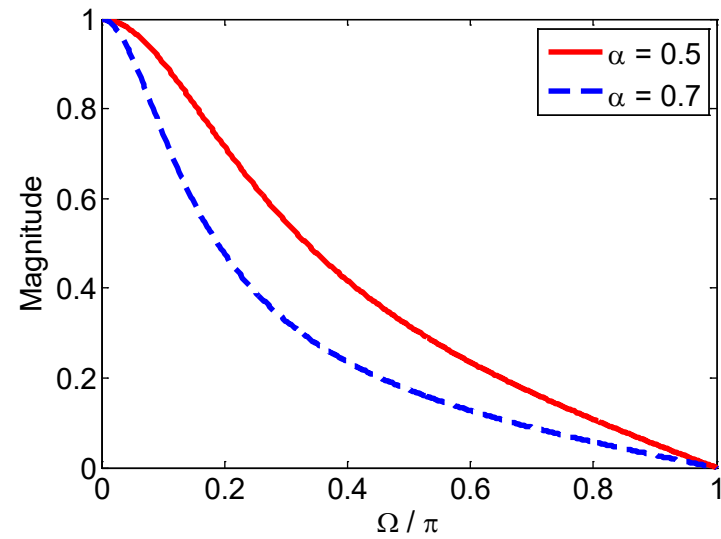
- **Highpass IIR digital filters**

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

$$H_{\mathrm{HP},1}\left(e^{j\Omega}\right) = \frac{1+\alpha}{2} \cdot \frac{e^{j\Omega}-1}{e^{j\Omega}-\alpha}$$
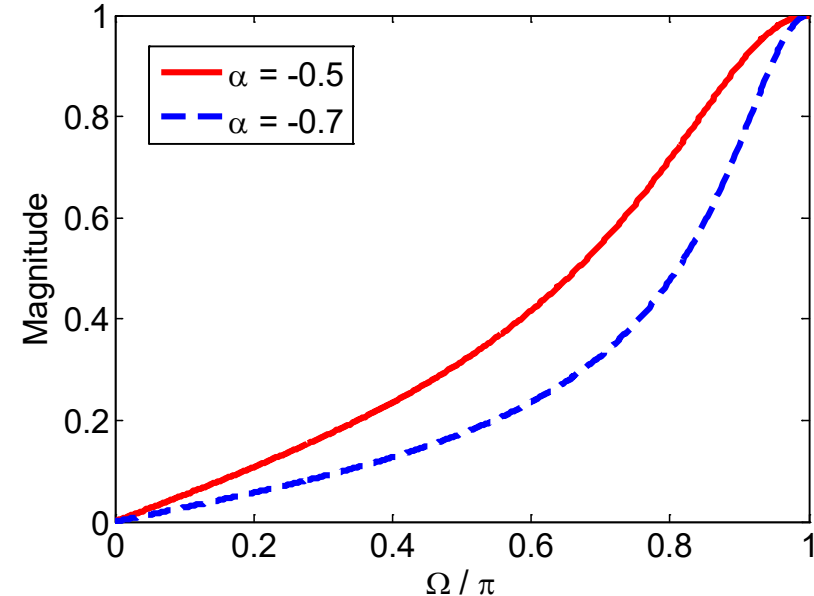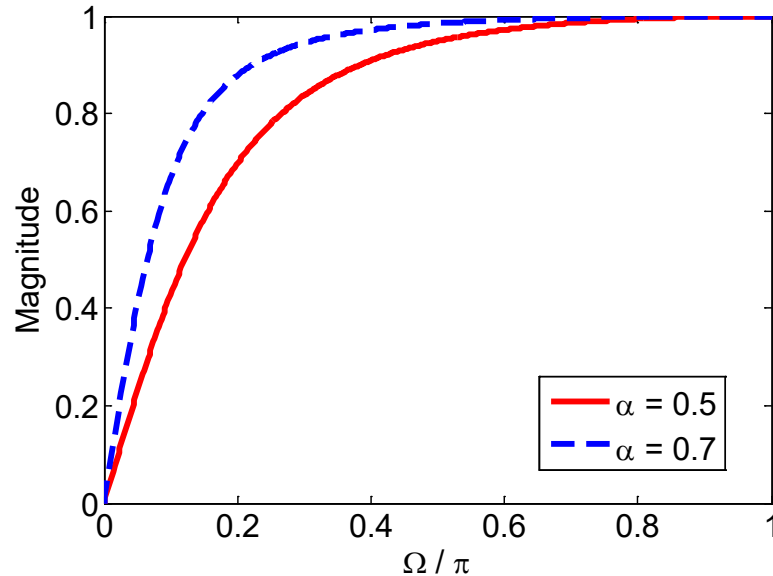
- **Bandpass IIR digital filters**

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

$\beta = 0.4$
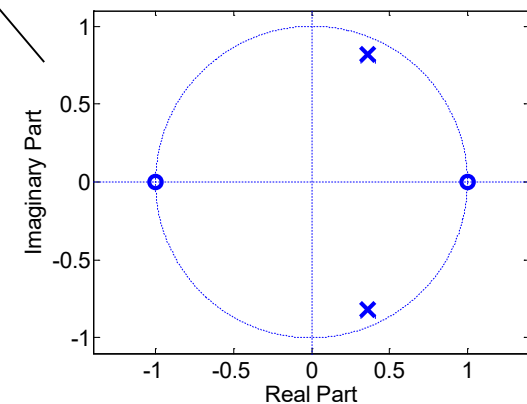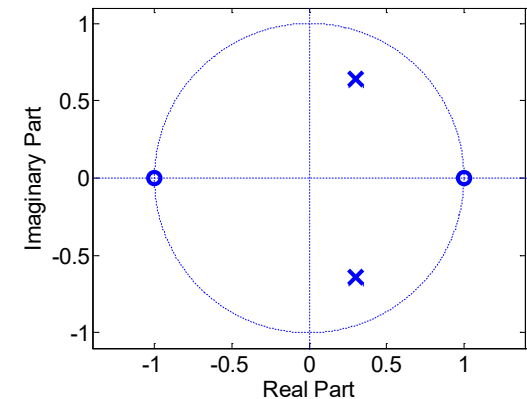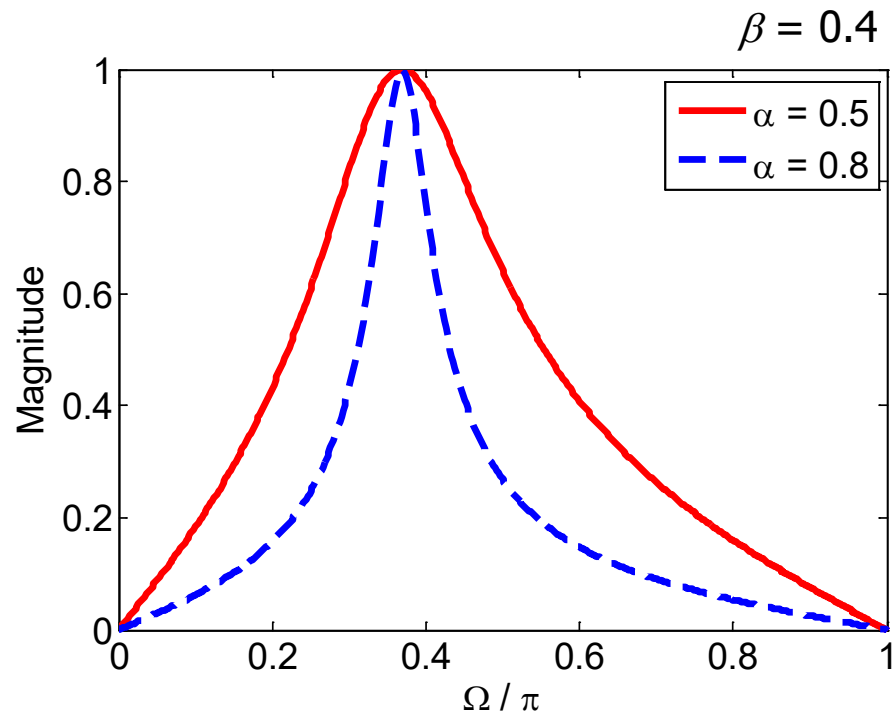
- **Poles:** $\quad p_{1/2} = r \cdot e^{\pm j\phi} \; ; \; r = \sqrt{\alpha} \; ; \; \cos\phi = \dfrac{\beta \cdot (1+\alpha)}{2 \cdot \sqrt{\alpha}}$

- **Maximum:** $\quad \Omega = \Omega_0 \;\; \text{with} \;\; \beta = \cos\Omega_0$
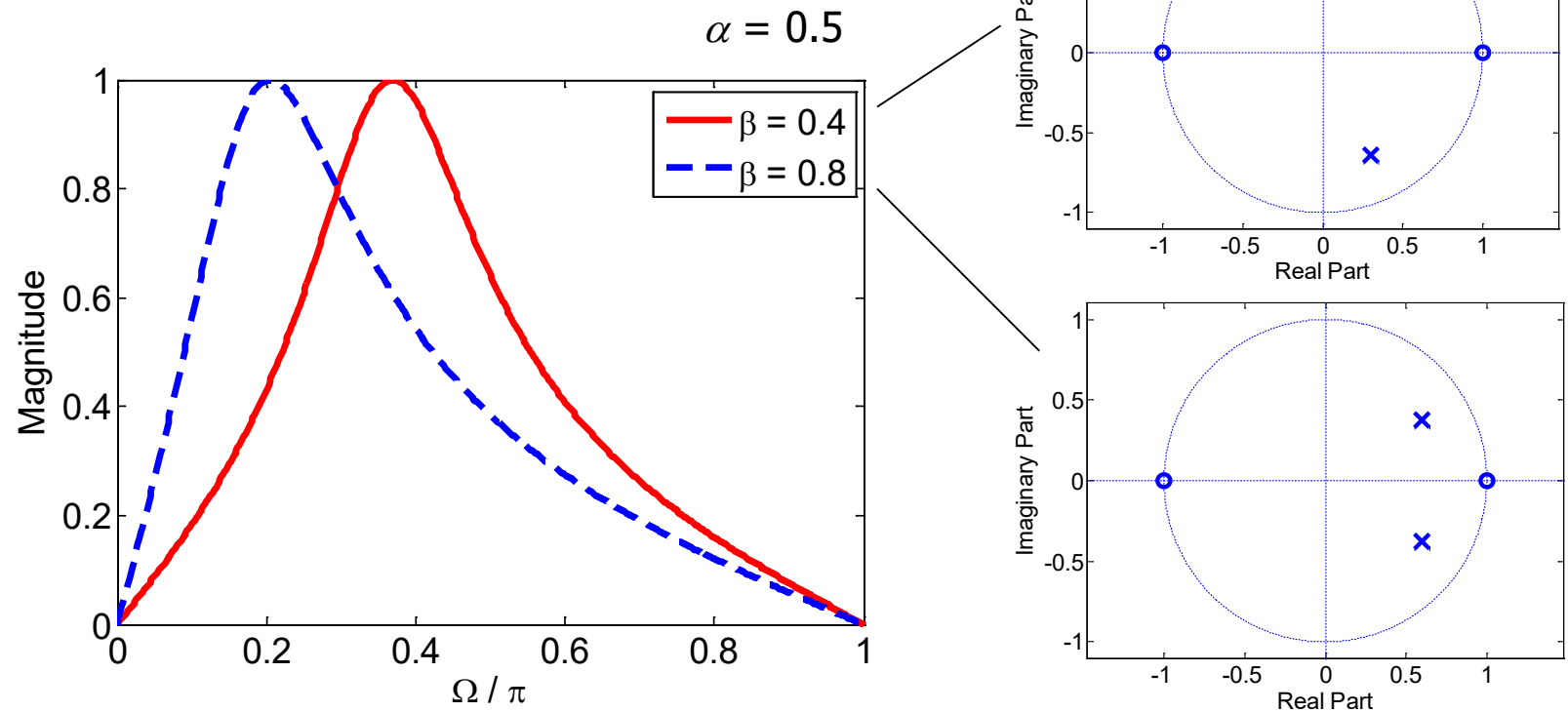
IIR notch digital filters

$$H_{\mathrm{BP},2}(z) = \frac{1+\alpha}{2} \cdot \frac{1 - 2\cdot\beta\cdot z^{-1} + z^{-2}}{1 - \beta\cdot(1+\alpha)\cdot z^{-1} + \alpha\cdot z^{-2}} = \frac{1-\alpha}{2} \cdot \frac{z^2 - 2\cdot\beta\cdot z + 1}{z^2 - \beta\cdot(1+\alpha)\cdot z + \alpha}$$
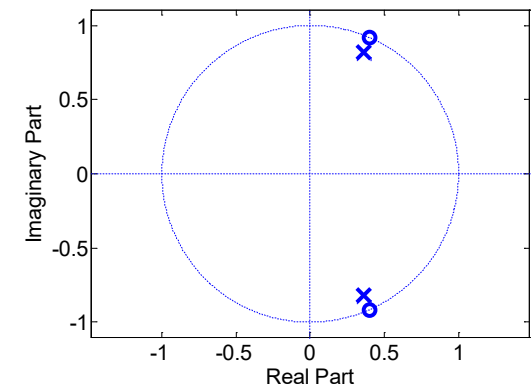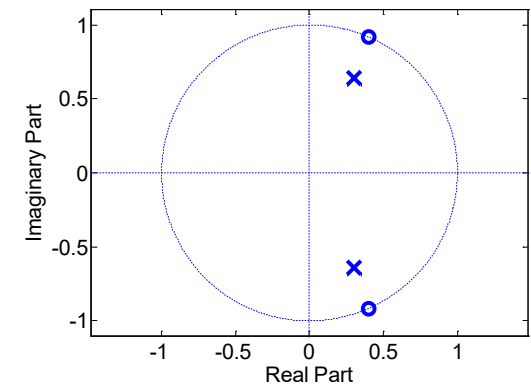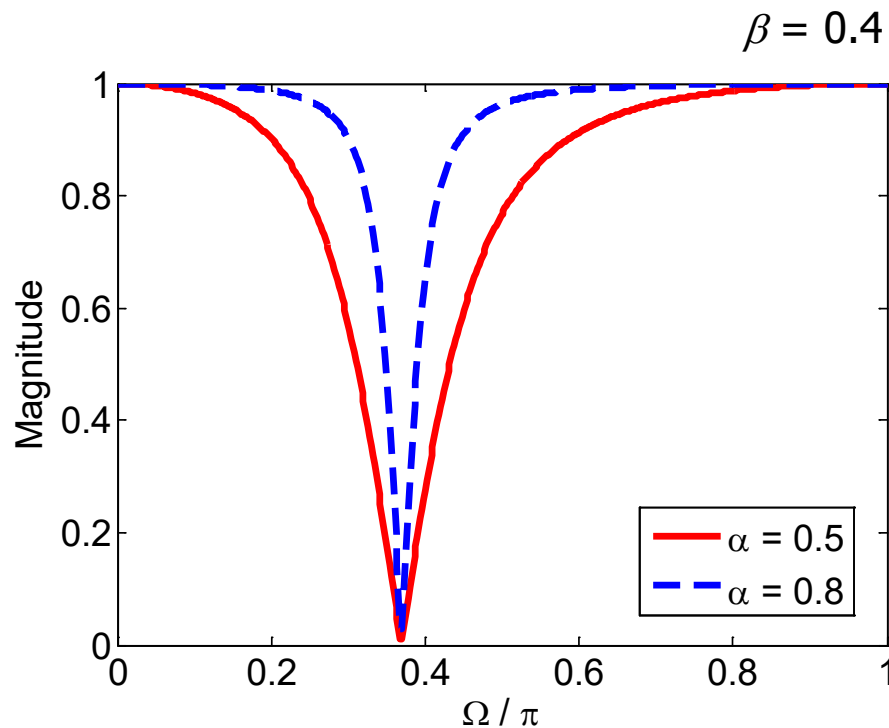
- Zeros: $z_{1/2} = r \cdot \mathrm{e}^{\pm \mathrm{j}\phi}$; $r = 1$; $\cos\phi = \beta$

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

- Notch: $\Omega = \Omega_0$ with $\beta = \cos\Omega_0$

$\alpha = 0.5$
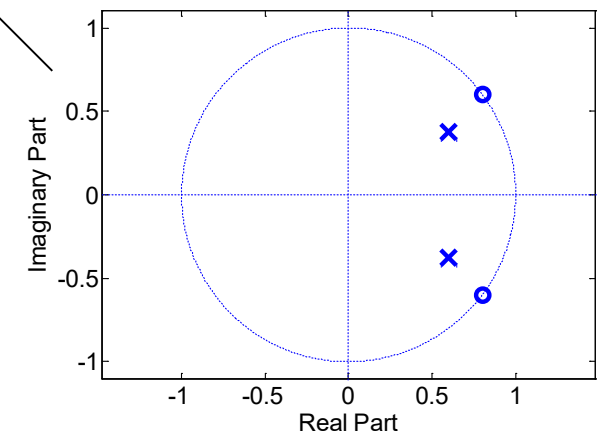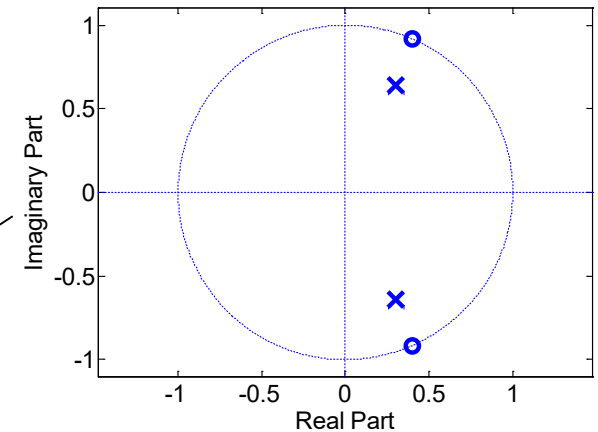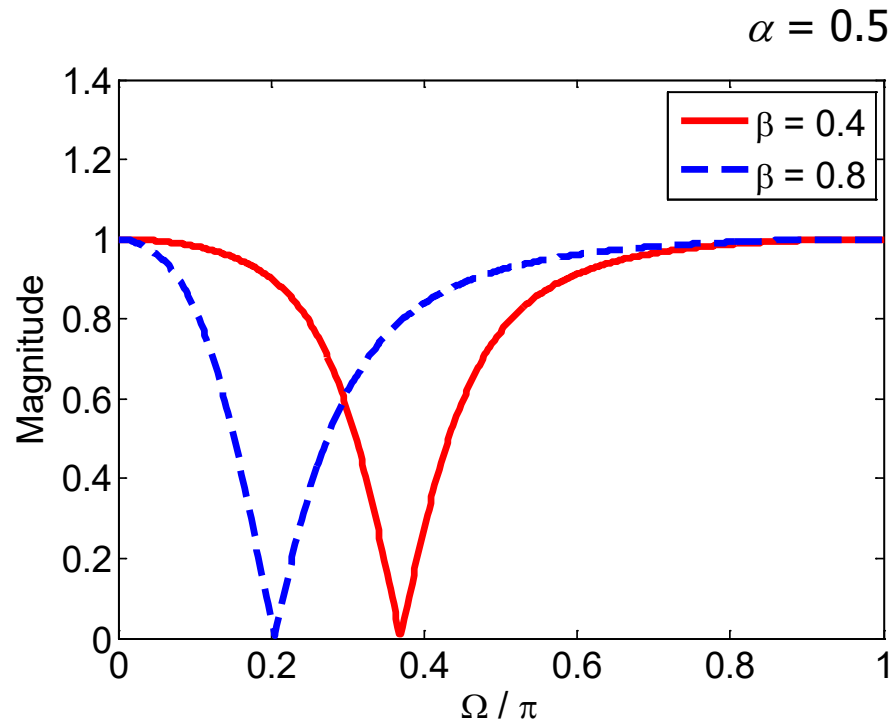
- **Extended Definition**

$$H(\mathrm{e}^{\mathrm{j}\Omega}) = \widetilde{H}(\Omega) \cdot \mathrm{e}^{\mathrm{j}\varphi} \text{ with } \varphi = a \cdot \Omega + b$$

$$\widetilde{H}(\Omega): \text{ Zero - phase response}$$

phase delay $\tau_{\mathrm{p}} = -\dfrac{\varphi}{\Omega} = -a - \dfrac{b}{\Omega}$ 　　　 group delay $\tau_{\mathrm{g}} = -\dfrac{\mathrm{d}\varphi}{\mathrm{d}\Omega} = -a$

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

Order N of transfer function can be even or odd.

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

  $$h[n] = h[N - n] \ \text{ with } \ 0 \le n \le N \,; \ b = 0 \text{ or } \pi$$

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

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

- General form of frequency response

  $$H(\mathrm{e}^{\mathrm{j}\Omega}) = \widetilde{H}(\Omega) \cdot \mathrm{e}^{\mathrm{j}b} \cdot \mathrm{e}^{-\mathrm{j}N\Omega/2}$$

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

- **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

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

- **Type IV:** Antisymmetric impulse response, N odd

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

6.2.2.2 Zero Locations of Linear-Phase FIR Filter

- **Type I and II**

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

- **Type III and IV**

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

- **Requirements for zeros**

  - Real zero ($z \neq \pm 1$): $\qquad\qquad\qquad z = r$ <u>and</u> $z = 1/r$ (2 zeros)

  - Zero on unit circle ($z \neq \pm 1$): $\qquad z = \mathrm{e}^{\mathrm{j}\phi}$ <u>and</u> $z = \mathrm{e}^{-\mathrm{j}\phi}$ (2 zeros)

  - Complex zero ($|z| \neq 1$, not real): $\quad z = r \cdot \mathrm{e}^{\pm\mathrm{j}\phi}$ <u>and</u> $z = \dfrac{1}{r} \cdot \mathrm{e}^{\pm\mathrm{j}\phi}$ (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 $z = -1$. → 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

$$N \approx \frac{-20 \cdot \log_{10}\left(\sqrt{\delta_1 \cdot \delta_2}\right) - 13}{14.6 \cdot (\Omega_2 - \Omega_1)/(2\pi)}$$

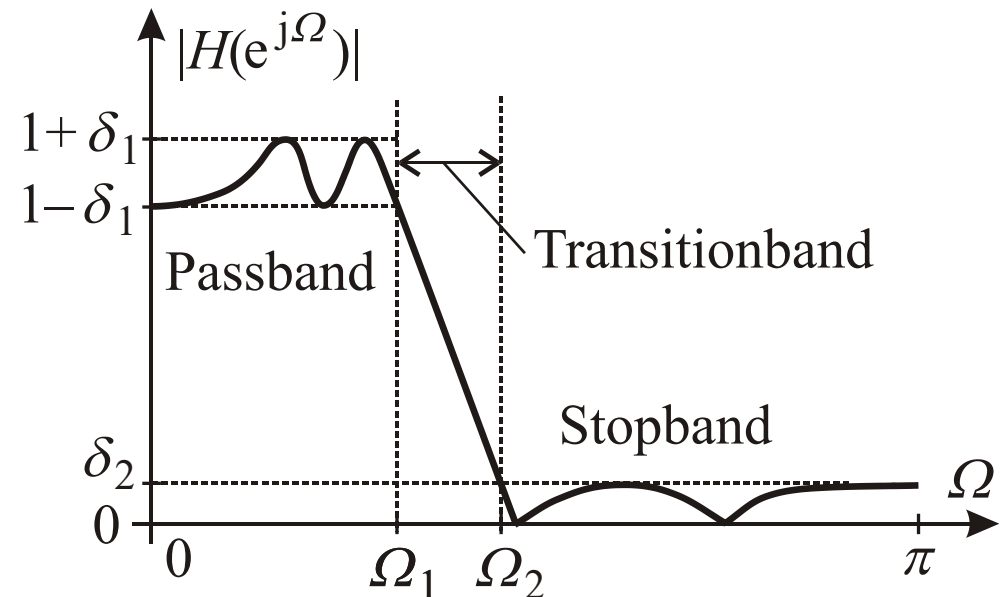- BELLANGER's formula

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

- 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

Example: Positioning system

closed-loop system

simplified block
diagram

negative feedback
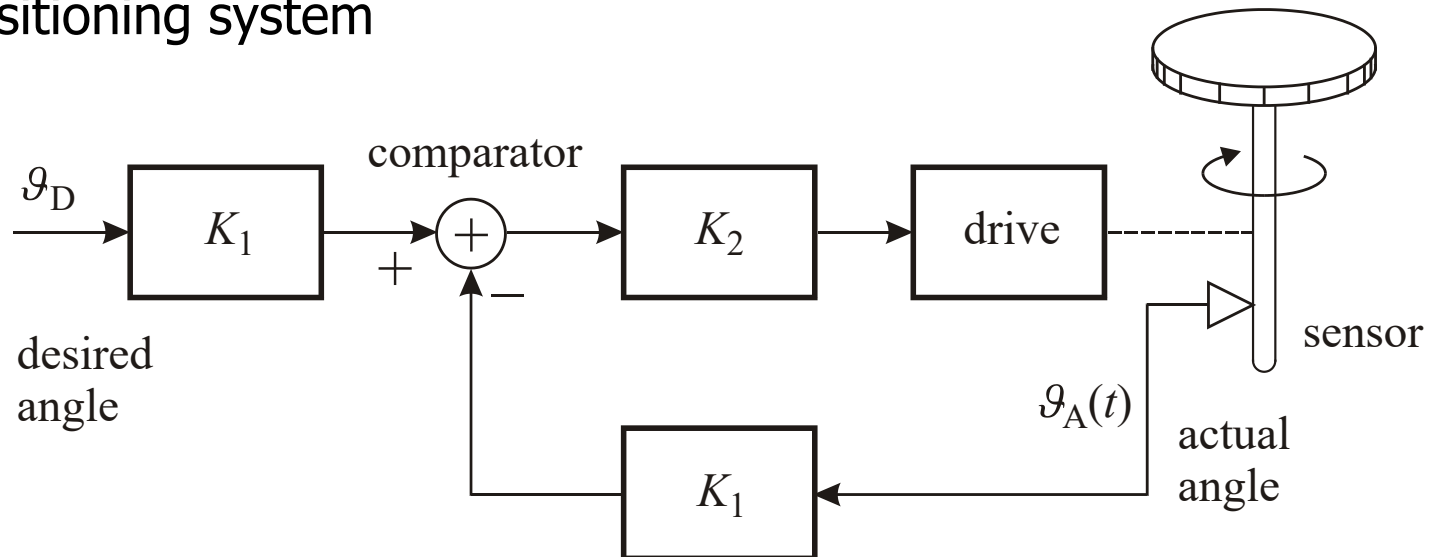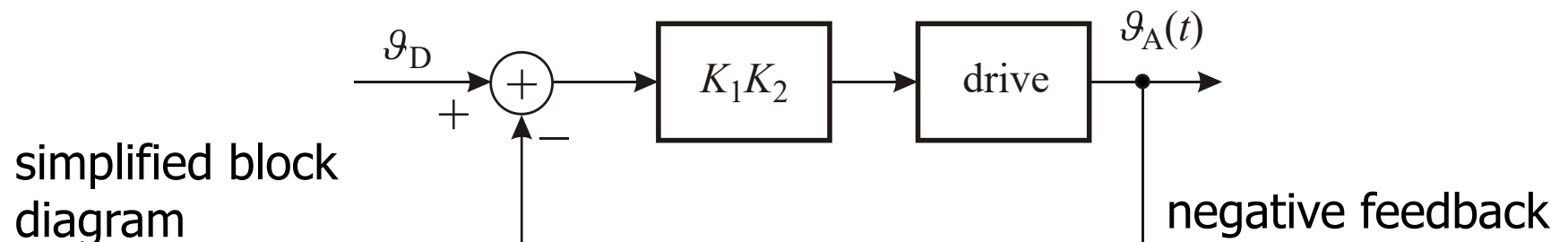
6.3.1 Linear Feedback Systems

- **Continuous-time LTI feedback system**
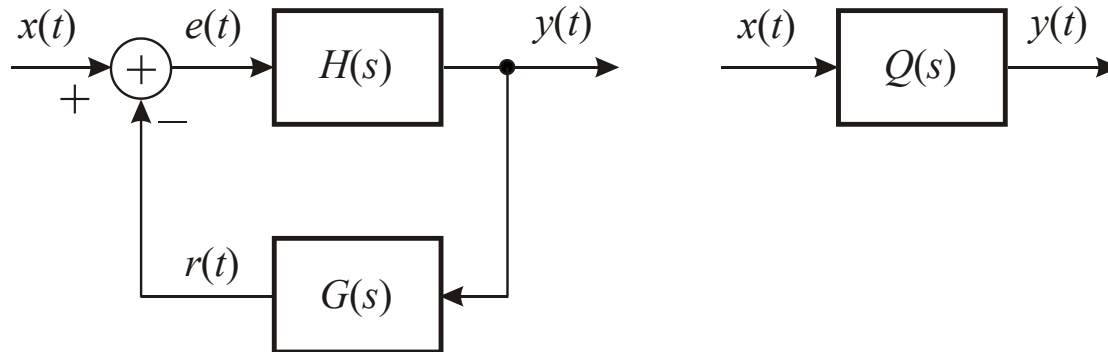
$$Q(s) = \frac{H(s)}{1 + G(s) \cdot H(s)}$$

- **Discrete-time LTI feedback system**
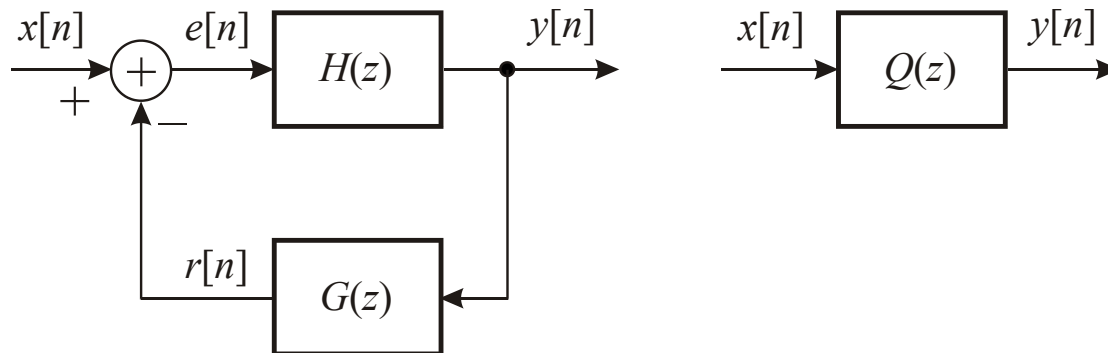
$$Q(z) = \frac{H(z)}{1 + G(z) \cdot H(z)}$$

- $H(s)$, $H(z)$: 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

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

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

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

- Discrete-time system: **population growth**

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

$y[n]$: population at generation n

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

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

$r[n]$: 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
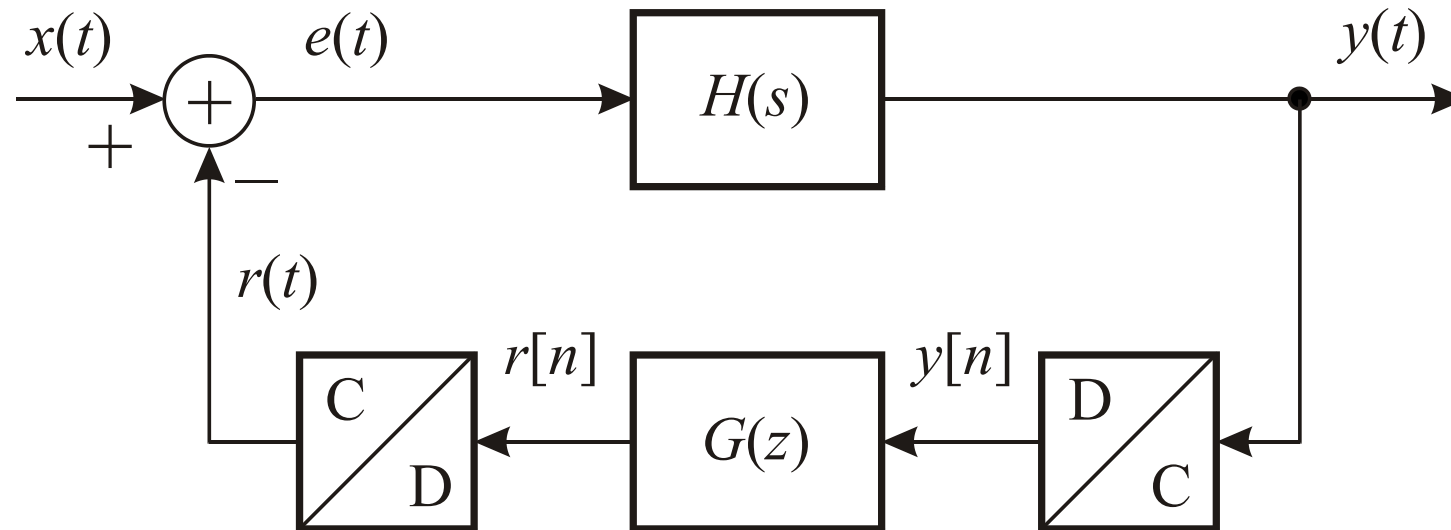
- Assumption

$$x(t) = x[n] \text{ for } n \cdot T_s \leq t < (n+1) \cdot T_s$$

- Resultant is a completely discrete-time feedback system.
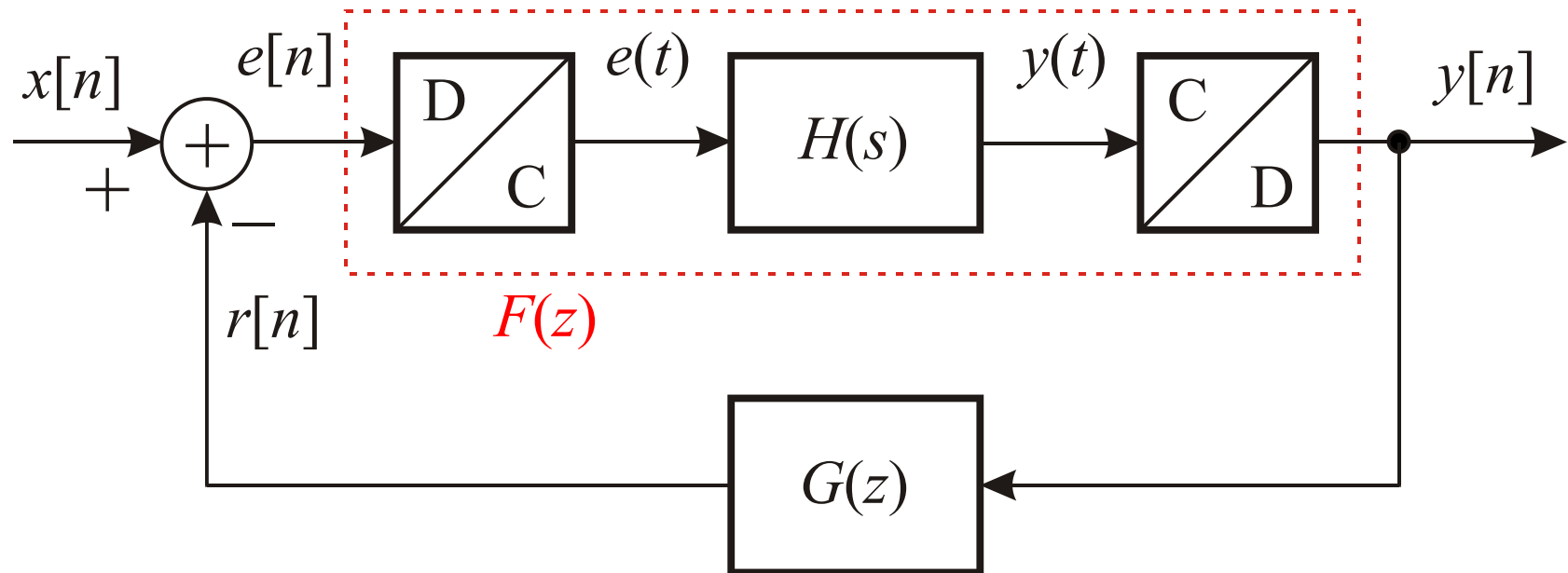
6.3.4 Root-Locus Analysis

- The feedback systems are supplemented by an adjustable gain element.
- **Continuous-time systems**
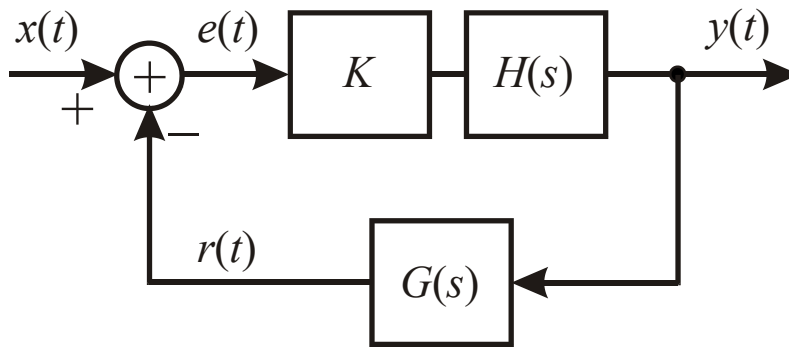
$$Q(s) = \frac{K \cdot H(s)}{1 + K \cdot G(s) \cdot H(s)}$$

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

- **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

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

- $K \to 0 \implies G(\dots) \cdot H(\dots) \to -\infty$ $\leftarrow$ poles of $G(\dots) \cdot H(\dots)$

- $K \to \pm\infty \implies G(\dots) \cdot H(\dots) \to 0$ $\leftarrow$ zeros of $G(\dots) \cdot H(\dots)$

- $\arg\{G(\dots) \cdot H(\dots)\} = n \cdot \pi$; n: integer
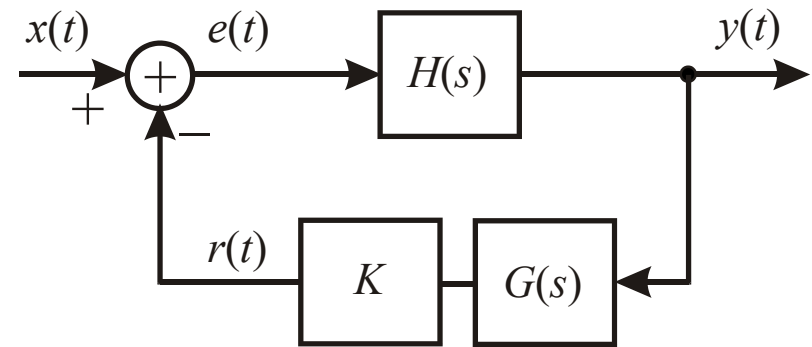
  - $\arg\{G(\dots) \cdot H(\dots)\} = k \cdot \pi$; k: odd

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

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

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

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

The root-locus curve

- starts at zeros of $G(\dots) \cdot H(\dots)$ for $K \to -\infty$,

- equals poles of $G(\dots) \cdot H(\dots)$ for $K = 0$,

- ends at zeros of $G(\dots) \cdot H(\dots)$ 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

$$W(u) = \frac{u - z_1}{u - p_1} = \frac{N_1 \cdot e^{j\beta_1}}{D_1 \cdot e^{j\alpha_1}}$$
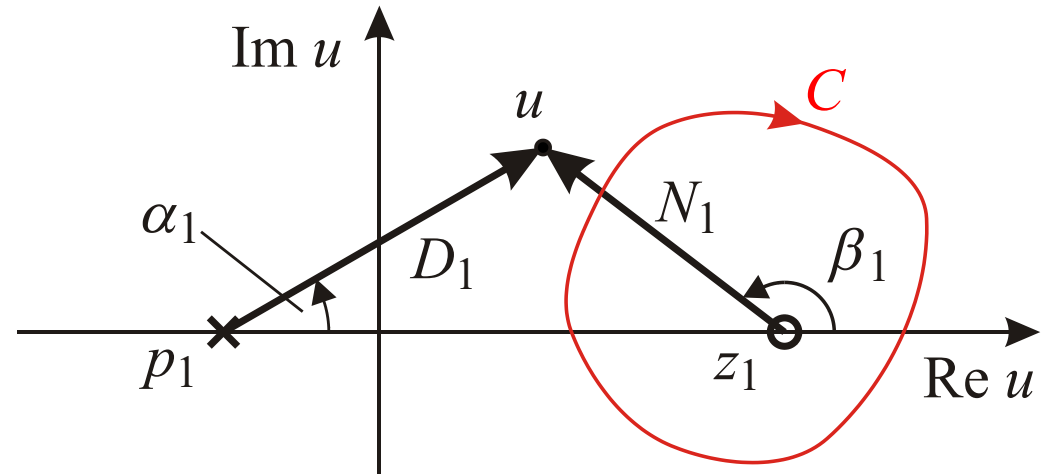
If u traverses along a closed contour C in the u-plane <u>once</u> in the clockwise direction,

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 $H(\mathrm{j}\omega) \cdot G(\mathrm{j}\omega)$ 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 $H(\mathrm{j}\omega) \cdot G(\mathrm{j}\omega)$ must equal the number of right-half-plane poles of $H(s) \cdot G(s)$.
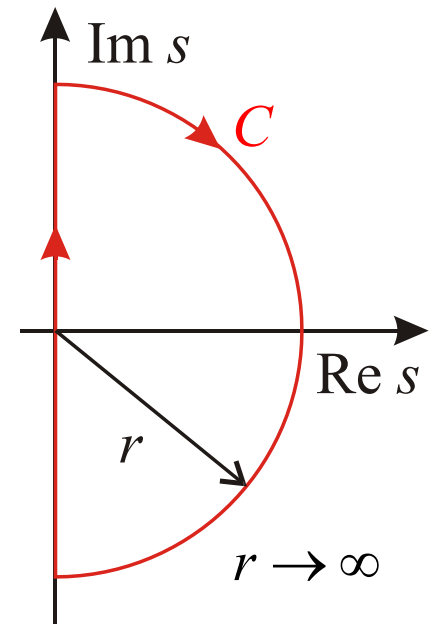
Discrete-Time NYQUIST Stability Criterion

- **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 $H\left(e^{j\Omega}\right) \cdot G\left(e^{j\Omega}\right)$ as Ω varies from 0 to 2π is called the NYQUIST plot.

- The number of clockwise encirclements of the point $-1/K$ by the NYQUIST plot equals the number of closed-loop poles outside the unit circle minus the number of poles of $H(z) \cdot G(z)$ outside the unit circle.

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