
Digital Signal Processing (DSP)

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DIGITAL SIGNAL PROCESSING (DSP)

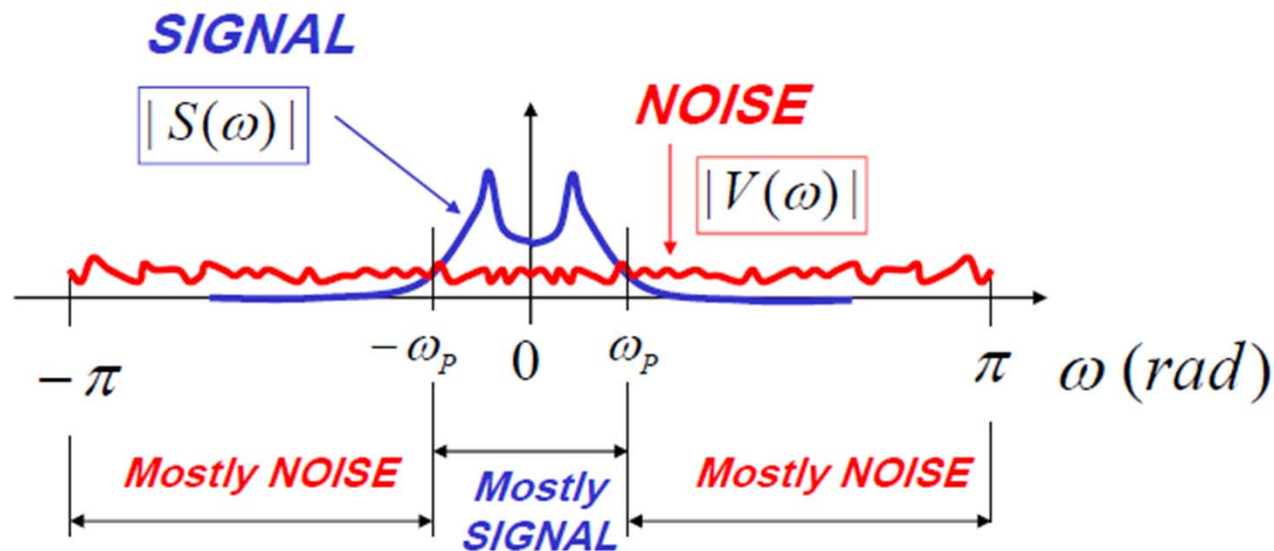
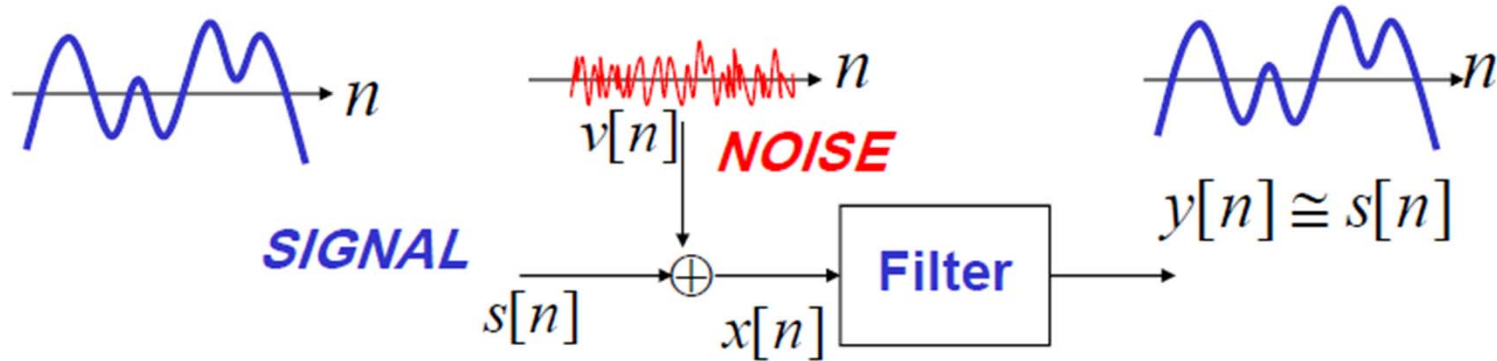
Lecture 8

Filter Design

Filter Design

- Ideal LPF
 - Non-causal
 - Unstable
 - Can not be realized!
- We are going to design realizable filter satisfying a given set of specifications
- Analog filters
- Digital filters

Filter Design

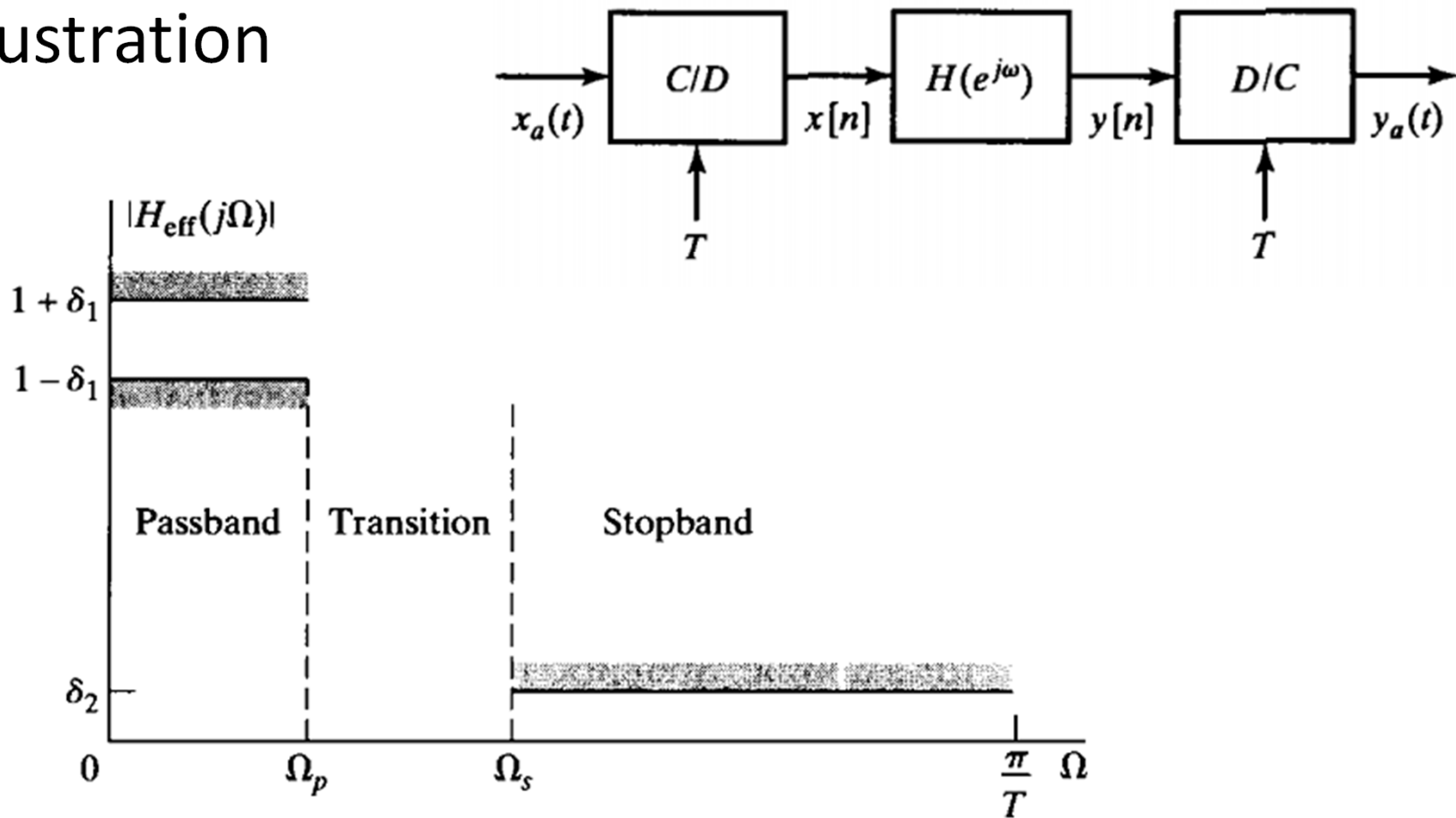


Filter Design

- Design problem: a frequency response (system function) is given. Compute the impulse response/coefficients of the realizable system
- Causal system
- Stable system
- Sometimes linear phase
- Approximate design

Filter Design

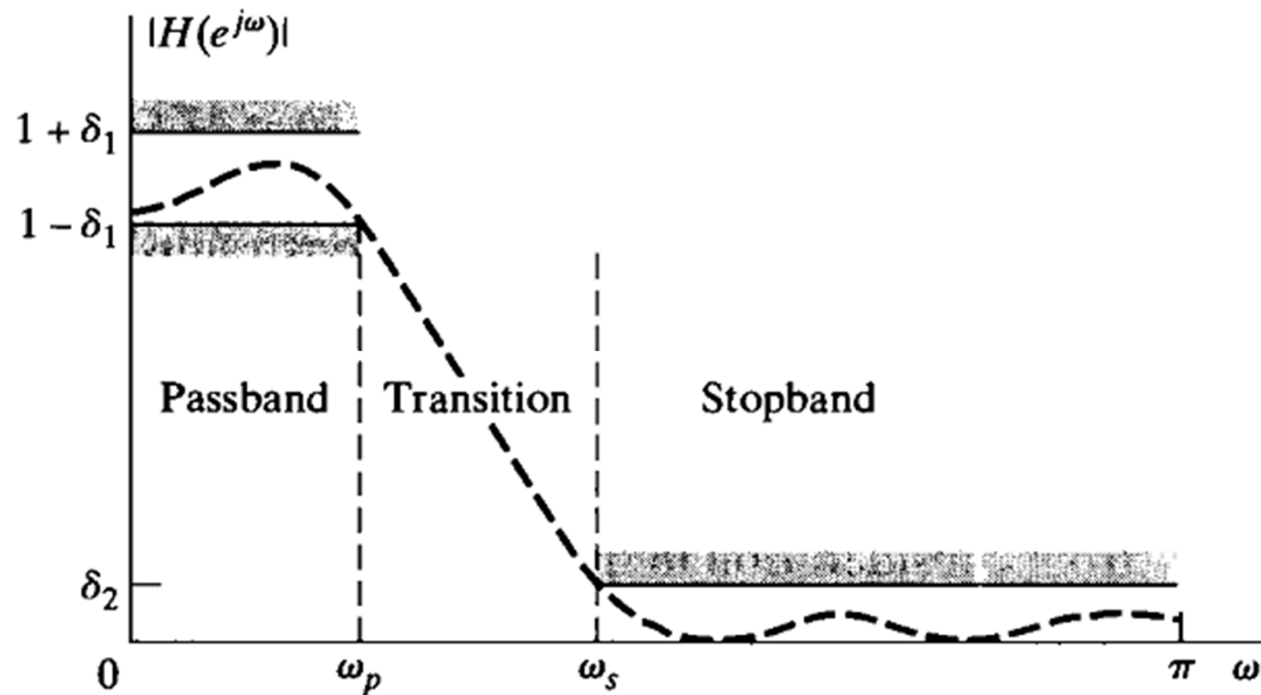
- Illustration



Filter Design

- Cont.

$$H(e^{j\omega}) = H_{\text{eff}}\left(j\frac{\omega}{T}\right), \quad |\omega| < \pi.$$



Filter Design

- A typical set of values

$$\delta_1 = 0.01,$$

$$\delta_2 = 0.001,$$

$$\Omega_p = 2\pi(2000),$$

$$\Omega_s = 2\pi(3000).$$

- Usually given in dB

$$\text{ideal passband gain in decibels} = 20 \log_{10}(1) = 0 \text{ dB}$$

$$\text{maximum passband gain in decibels} = 20 \log_{10}(1.01) = 0.086 \text{ dB}$$

$$\text{maximum stopband gain in decibels} = 20 \log_{10}(0.001) = -60 \text{ dB}$$

Filter Design

- Design constraints

- Passband

$$(1 - \delta_1) \leq |H(e^{j\omega})| \leq (1 + \delta_1), \quad |\omega| \leq \omega_p,$$

- Stopband

$$|H(e^{j\omega})| \leq \delta_2, \quad \omega_s \leq |\omega| \leq \pi$$

- Transition band

Filter Design

- FIR: finite impulse response
- IIR: infinite impulse response
- Remark: generalized linear phase
 - Linear phase

$$\theta(\omega) = \angle H(e^{j\omega}) = -c\omega \quad H(e^{j\omega}) = |H(e^{j\omega})|e^{-j\omega c}$$

- generalized linear phase

$$H(e^{j\omega}) = A(e^{j\omega})e^{-j\omega c + jb}$$

Filter Design

- Selection
 - FIR: causal+ stable+ GLP
 - IIR: causal+ stable+ satisfying spec. with lower complexity

- A rich literature on analog filter design
 - Butterworth
 - Chebyshev

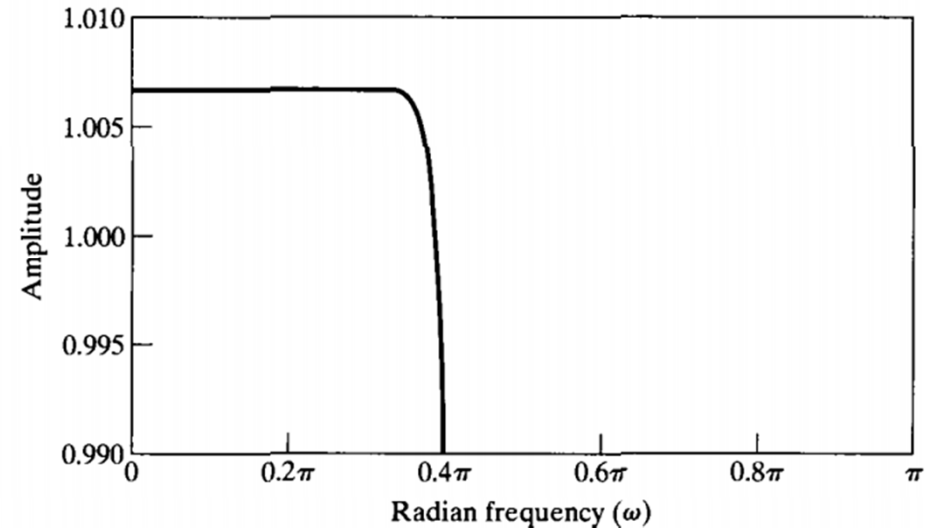
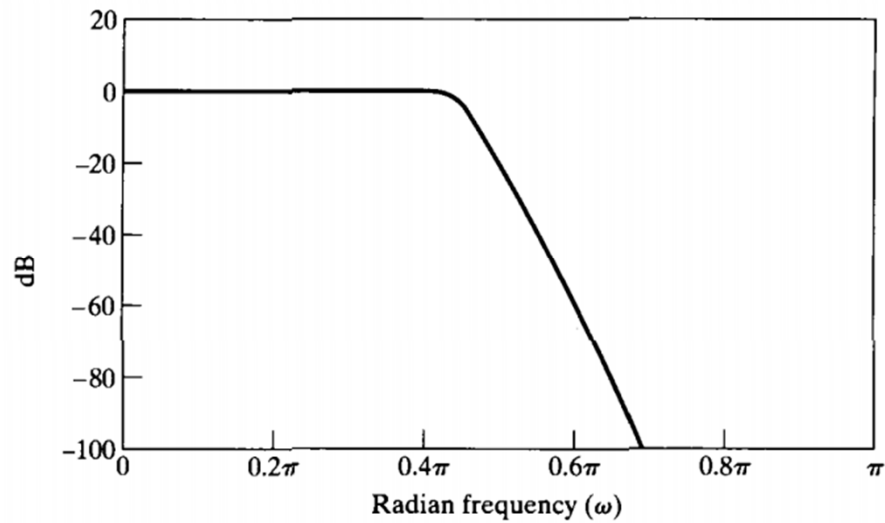
Filter Design

- Idea: digital filter (IIR) design via CT filters
 - 1) Design CT filter $h(t)$ from given frequency response
 - 2) Employ impulse invariance to obtain the DT version of the filter

- Other methods
 - Bilinear transform, etc.

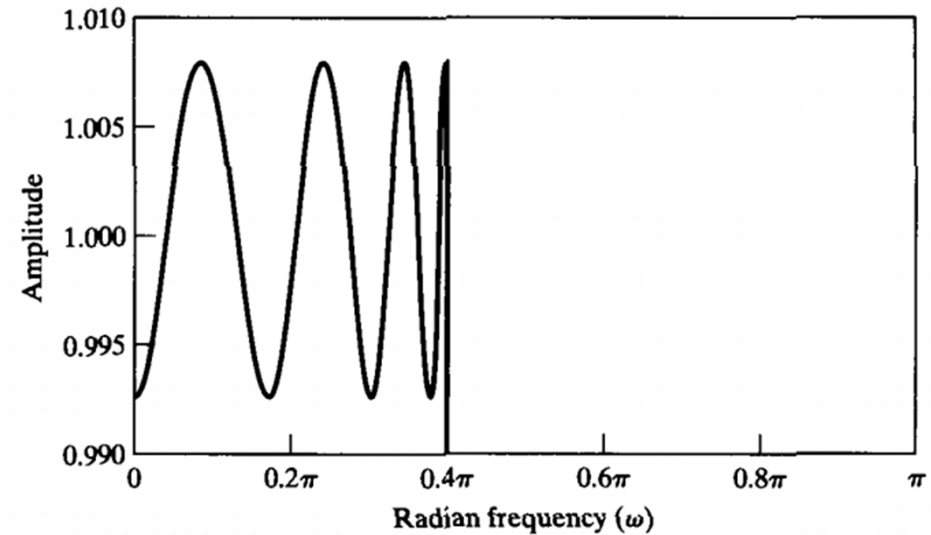
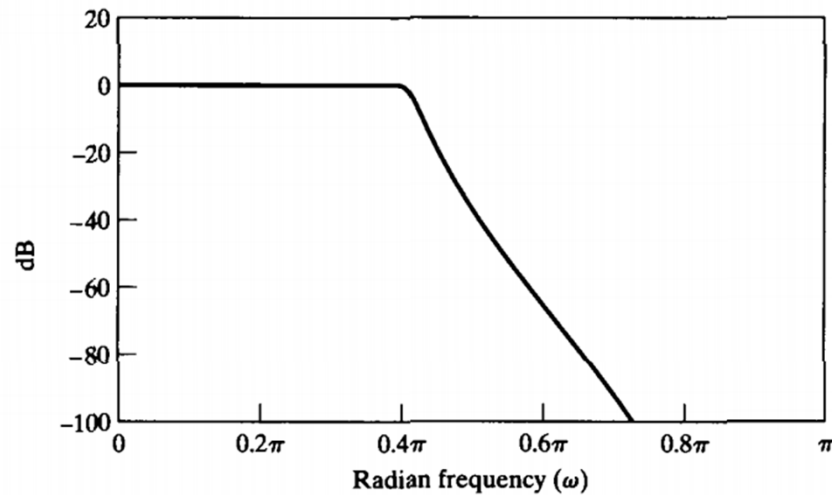
Filter Design

- Butterworth



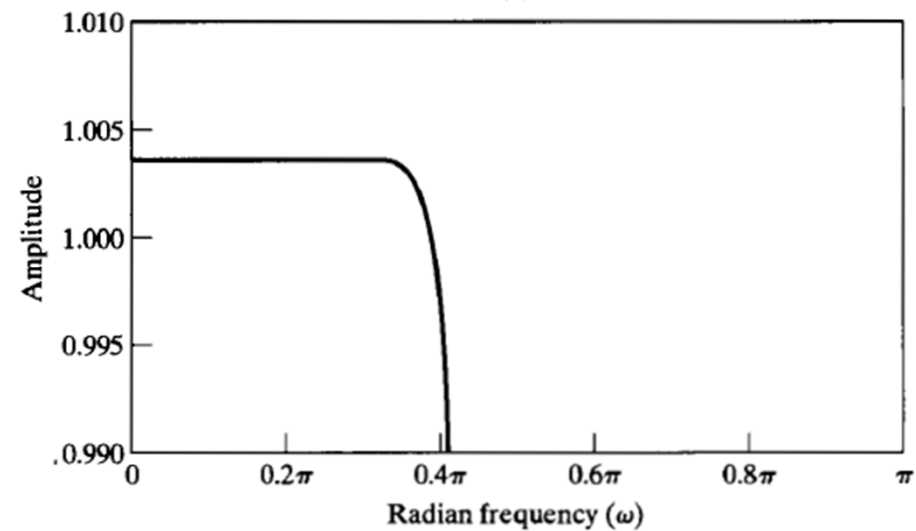
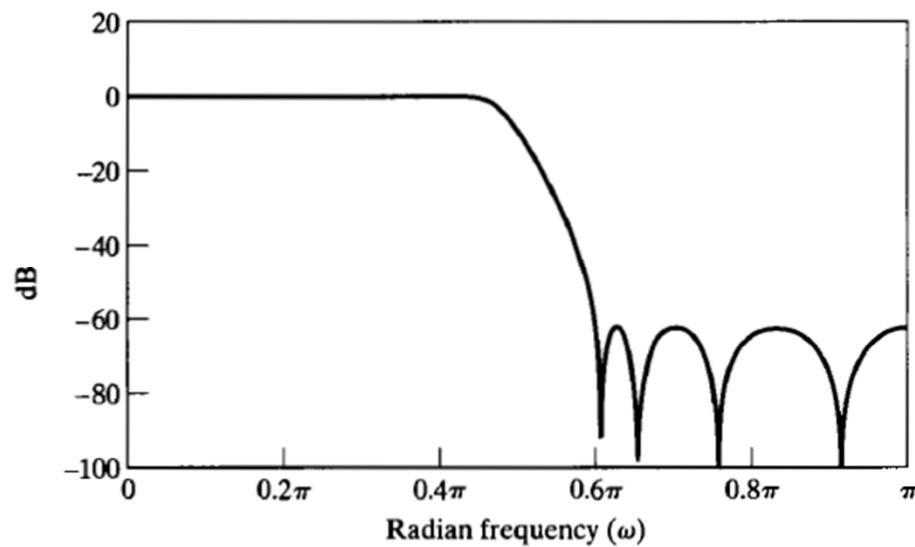
Filter Design

- Chebyshev (type I): ripple in passband



Filter Design

- Chebyshev (type II): ripple in stopband



Filter Design

- FIR filter design: our main focus
 - Arising in discrete-time implementation
 - Approximation of the response in discrete domain
 - (generalized) linear phase
- Design methods
 - Design by windowing
 - Design by optimal approximation

Filter Design

- Windowing method
 - Given the ideal response

$$H_d(e^{j\omega}) = \sum_{n=-\infty}^{\infty} h_d[n]e^{-j\omega n},$$

- And hence

$$h_d[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} H_d(e^{j\omega})e^{j\omega n}d\omega$$

- Usually non-causal/unstable

Filter Design

- The idea is to truncate the impulse response

$$H_d(e^{j\omega}) = \sum_{n=-\infty}^{\infty} h_d[n]e^{-j\omega n},$$

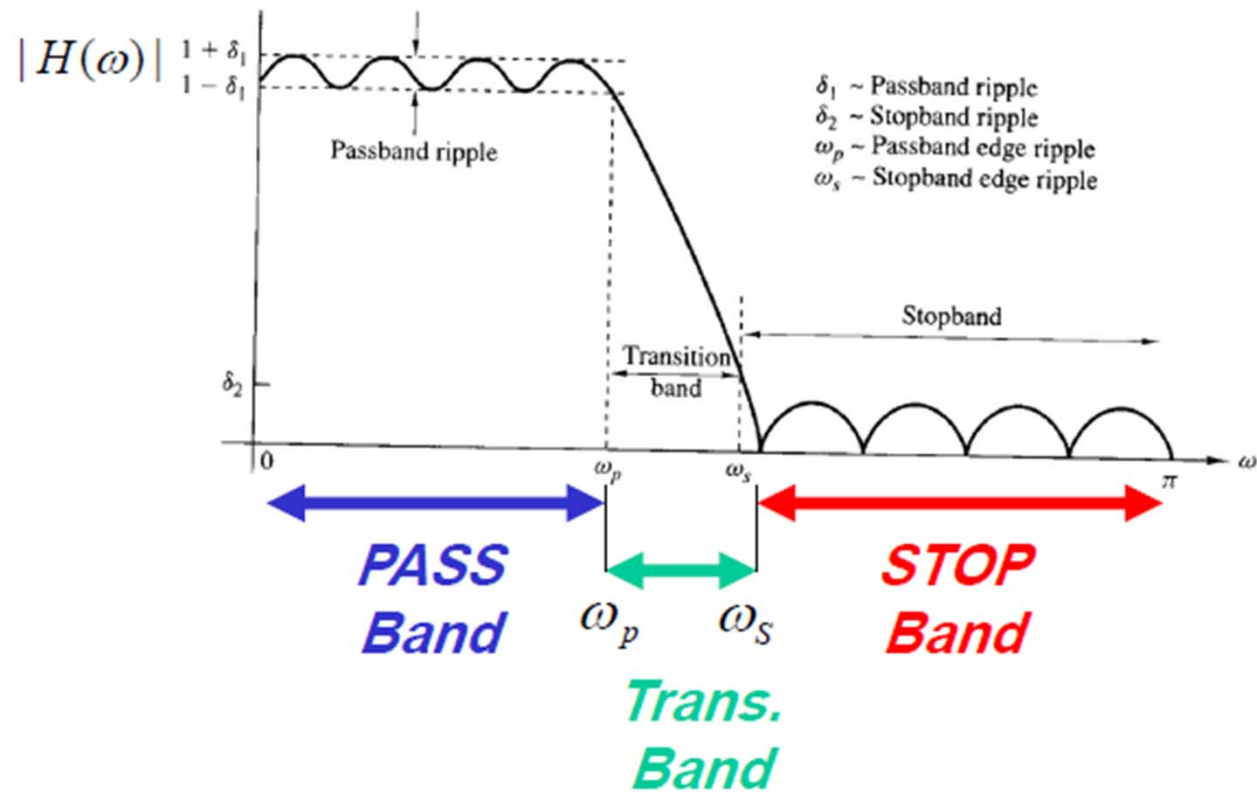
- Actually, the FIR approximation is given by

$$h[n] = \begin{cases} h_d[n], & 0 \leq n \leq M, \\ 0, & \text{otherwise.} \end{cases}$$

- Stable
- Causal

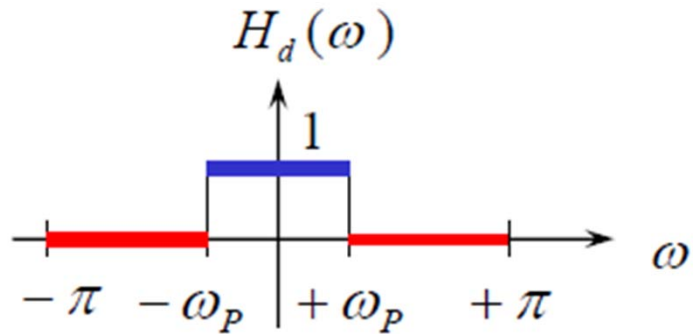
Filter Design

- Example: LPF



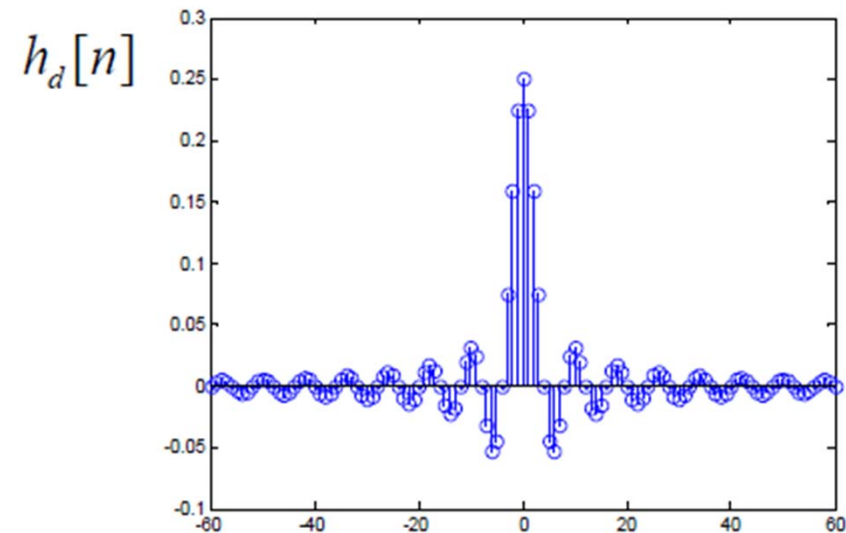
Filter Design

- Ideal form



$$h_d[n] = \frac{\sin(\omega_p n)}{\pi n} = \frac{\omega_p}{\pi} \operatorname{sinc}\left(\frac{\omega_p}{\pi} n\right)$$

$$H_d(\omega) = \sum_{n=-\infty}^{+\infty} h_d[n] e^{-j\omega n}$$



Filter Design

- Noting that

$$\lim_{n \rightarrow \pm\infty} h_d[n] = 0$$

- We use the following approximation

$$H_d(\omega) \cong \sum_{n=-L}^{+L} h_d[n] e^{-j\omega n}$$

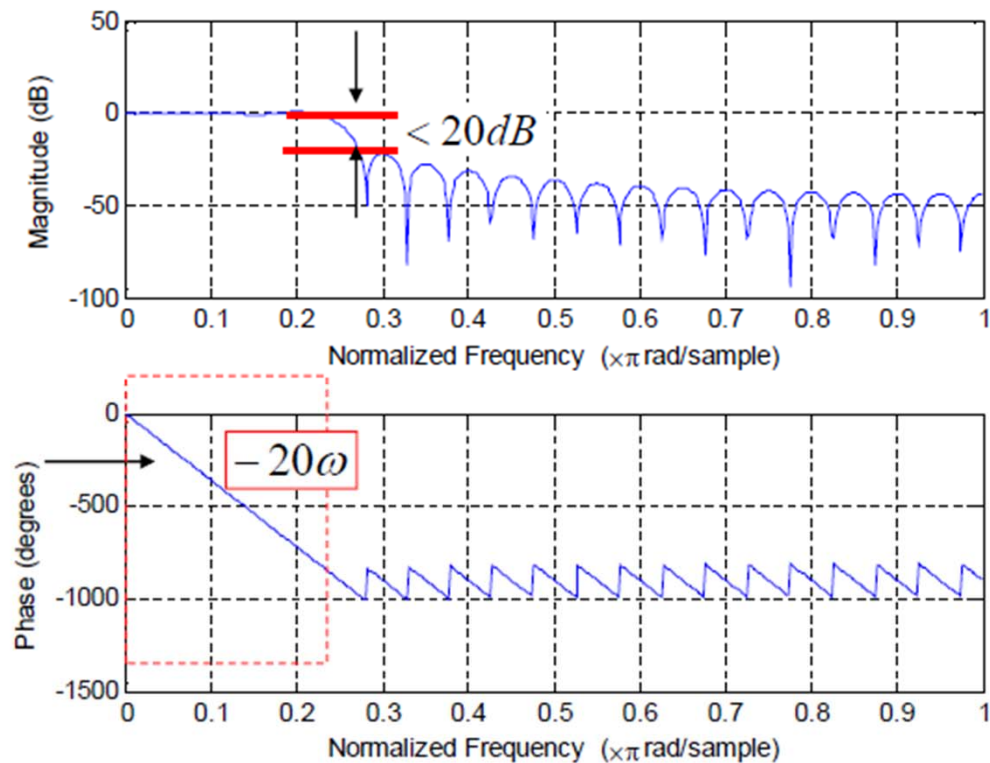
- And select the below filter to be causal

$$h[n] = h_d[n - L]$$

Filter Design

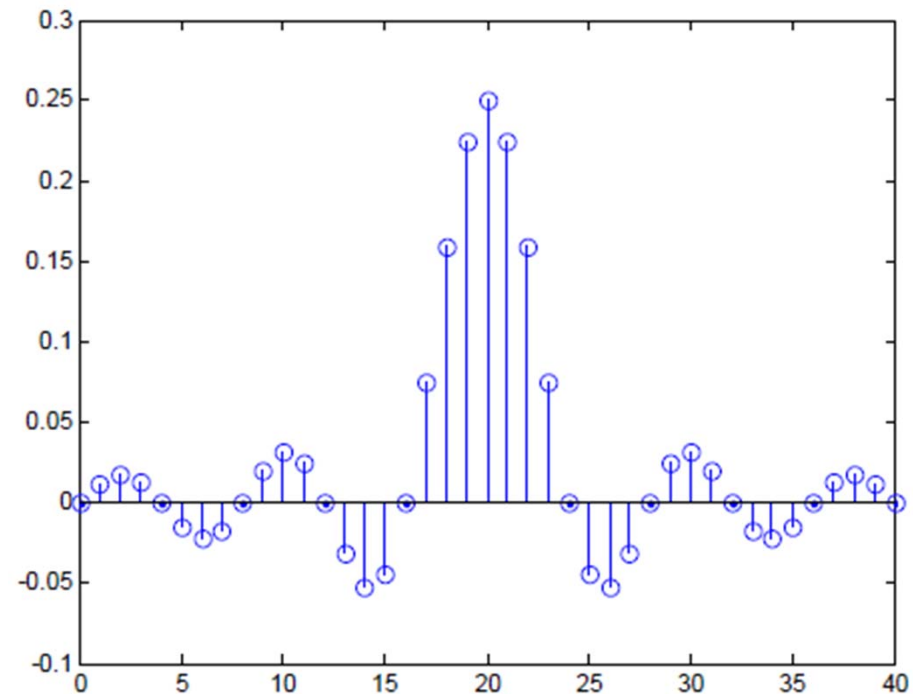
- Example (cont.)

$$\omega_p = \pi/4 \quad L = 20, \quad N = 40$$



Filter Design

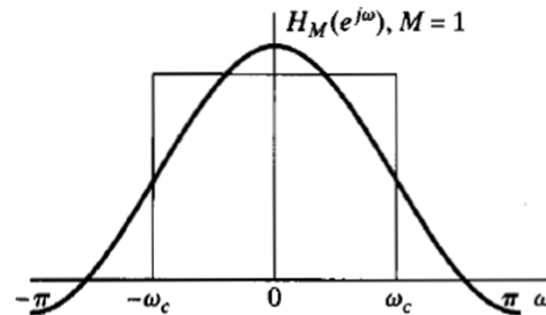
- And in the time domain
 - Causal
 - Stable



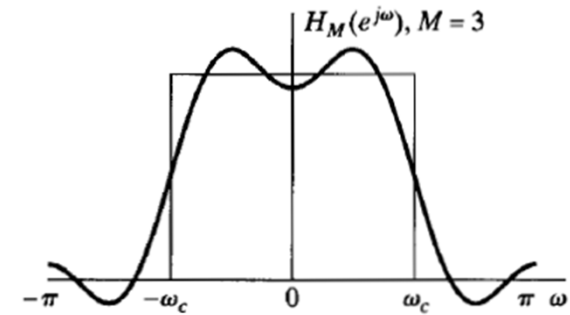
Filter Design

- Remember: Gibbs phenomenon

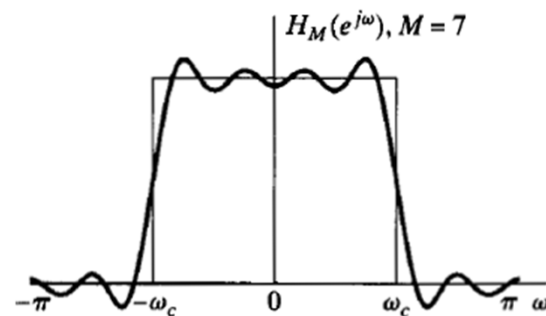
$$H_M(e^{j\omega}) = \sum_{n=-M}^M \frac{\sin \omega_c n}{\pi n} e^{-j\omega n}$$



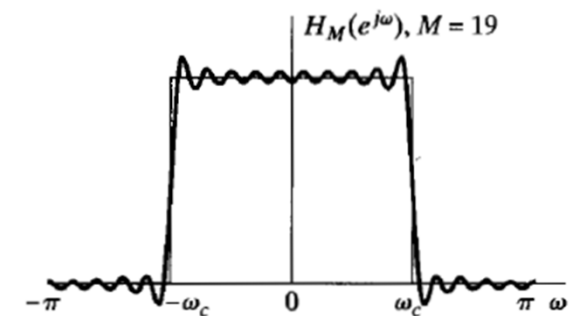
(a)



(b)



(c)



(d)

Filter Design

- The name of the method: windowing!!
 - Generally, we can use a window function $w[n]$:

$$h[n] = h_d[n]w[n],$$

- Win the previous case (rectangular window)

$$w[n] = \begin{cases} 1, & 0 \leq n \leq M, \\ 0, & \text{otherwise.} \end{cases}$$

Filter Design

- What happens by using window function?

$$h[n] = h_d[n]w[n],$$

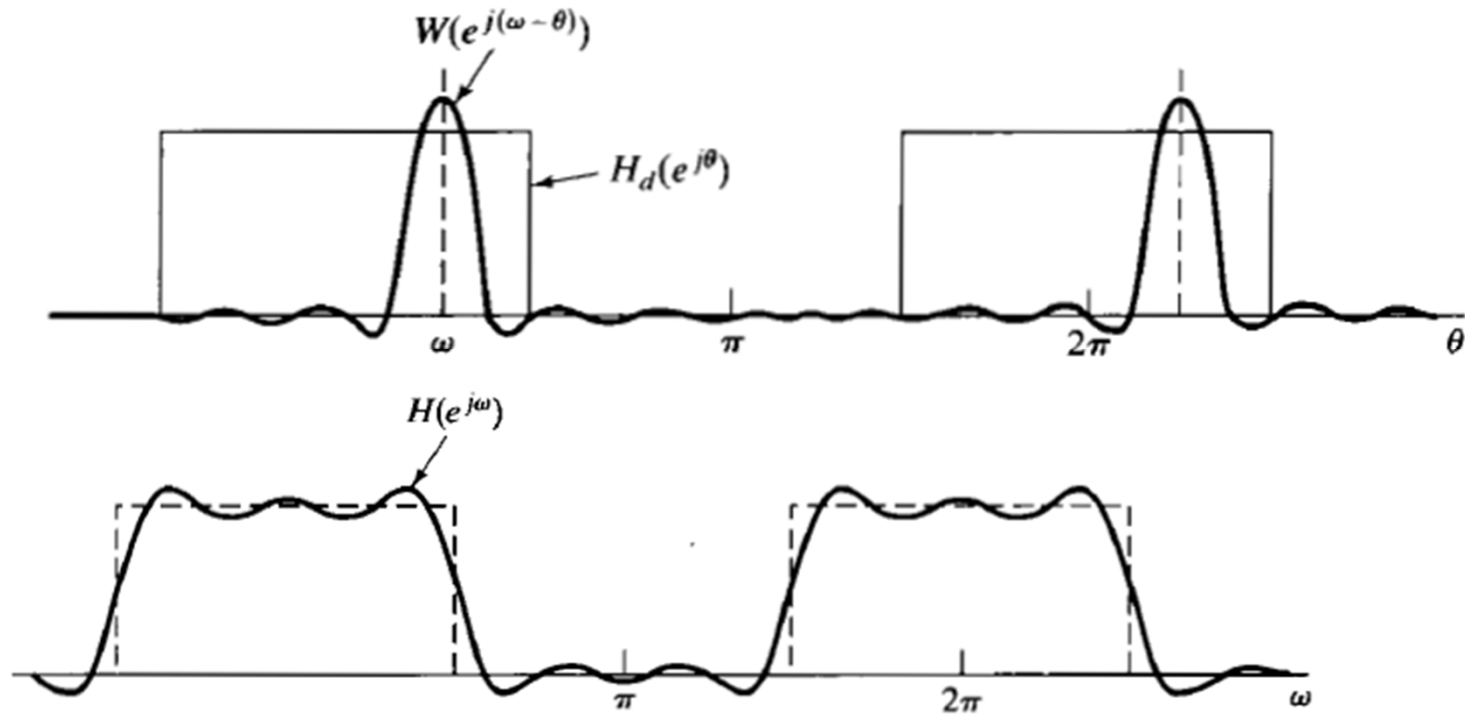
- In frequency domain

$$H(e^{j\omega}) = \frac{1}{2\pi} \int_{-\pi}^{\pi} H_d(e^{j\theta})W(e^{j(\omega-\theta)})d\theta$$

- The output spectrum is affected by the spectral content of the window

Filter Design

- Illustration



Filter Design

- Special case: $w[n]=1$ for all n
 - The resulting DTFT is given by impulse train with period; hence

$$H(e^{j\omega}) = H_d(e^{j\omega})$$

- (*)Observation: narrowband window functions are desired
 - Approximately reproducing the desired ideal filter

Filter Design

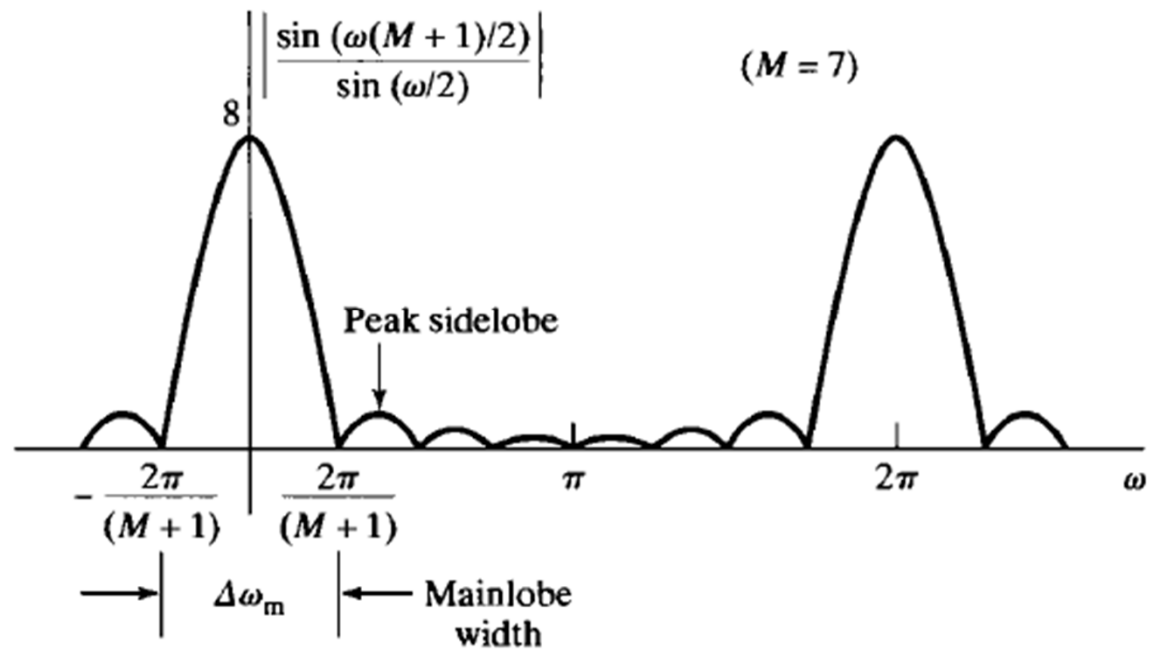
- (**)Implementation issue: shrinking the window function to reduce computational burden as much as possible
- Items * and ** are conflicting requirements
- Illustration of the conflict for rectangular window

$$W(e^{j\omega}) = \sum_{n=0}^M e^{-j\omega n} = \frac{1 - e^{-j\omega(M+1)}}{1 - e^{-j\omega}} = e^{-j\omega M/2} \frac{\sin[\omega(M+1)/2]}{\sin(\omega/2)}$$

- Linear phase

Filter Design

- Spectrum of the rectangular window



Filter Design

- Observations from the figure
 - Larger M leads to narrower mainlobe
 - The region between the first zero-crossing of the sides

$$\Delta\omega_m = 4\pi / (M + 1)$$

- Larger M is associated with higher computational load
- The energy of the sidelobes do not change with M
- Oscillation due to sidelobes: Gibbs phenomenon

Filter Design

- The choice of the window type is important
 - Mainlobe width: sharpness
 - Sidelobes energy: oscillation
- For a given length of the window, various shapes can be used
 - Example: smoothing the rectangular window for sidelobe reduction

Filter Design

- Window selection/design
 - Commonly used window functions

Rectangular

$$w[n] = \begin{cases} 1, & 0 \leq n \leq M, \\ 0, & \text{otherwise} \end{cases}$$

Bartlett (triangular)

$$w[n] = \begin{cases} 2n/M, & 0 \leq n \leq M/2, \\ 2 - 2n/M, & M/2 < n \leq M, \\ 0, & \text{otherwise} \end{cases}$$

Filter Design

- Common window (cont.)

Hanning

$$w[n] = \begin{cases} 0.5 - 0.5 \cos(2\pi n/M), & 0 \leq n \leq M, \\ 0, & \text{otherwise} \end{cases}$$

Hamming

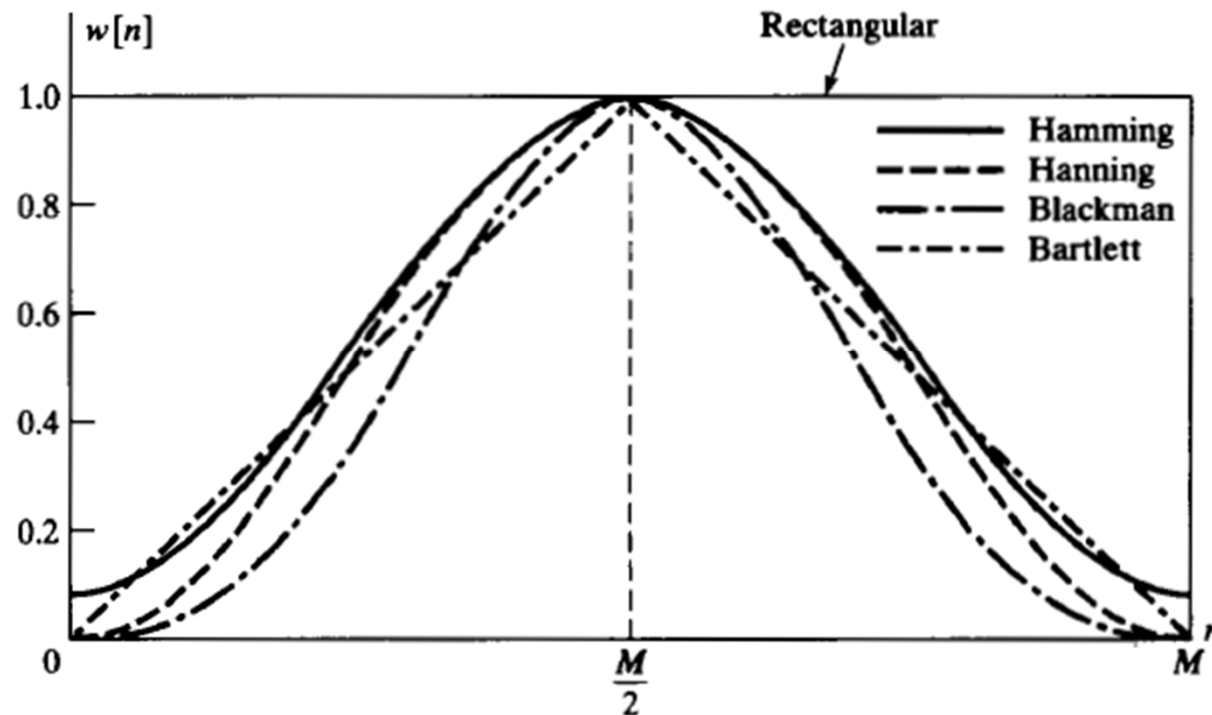
$$w[n] = \begin{cases} 0.54 - 0.46 \cos(2\pi n/M), & 0 \leq n \leq M, \\ 0, & \text{otherwise} \end{cases}$$

Blackman

$$w[n] = \begin{cases} 0.42 - 0.5 \cos(2\pi n/M) + 0.08 \cos(4\pi n/M), & 0 \leq n \leq M, \\ 0, & \text{otherwise} \end{cases}$$

Filter Design

- Time domain behavior



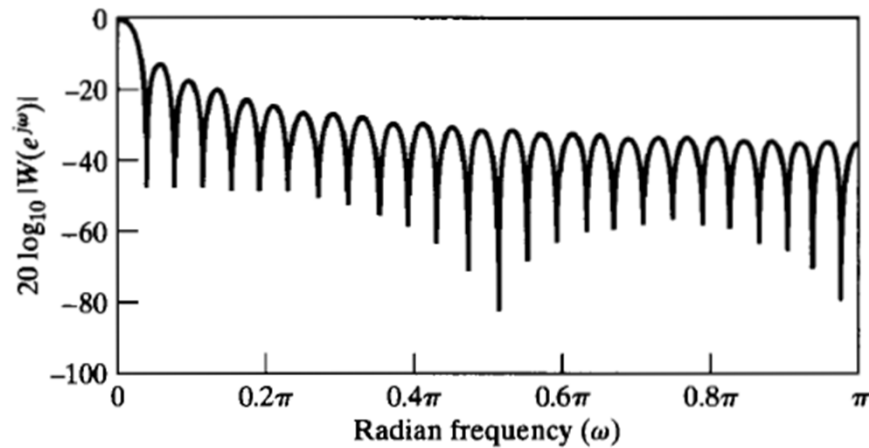
Filter Design

- Application
 - FIR filter design
 - Spectral analysis

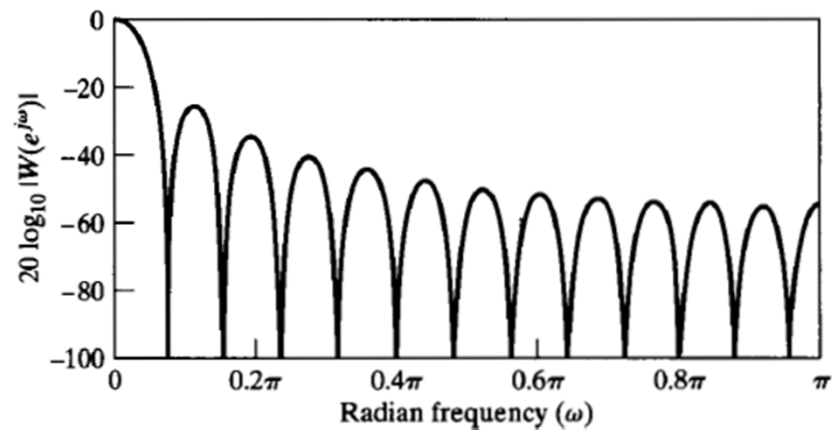
- Properties
 - Simple functional form
 - Computation
 - Spectra are concentrated around zero frequency

Filter Design

- Spectra



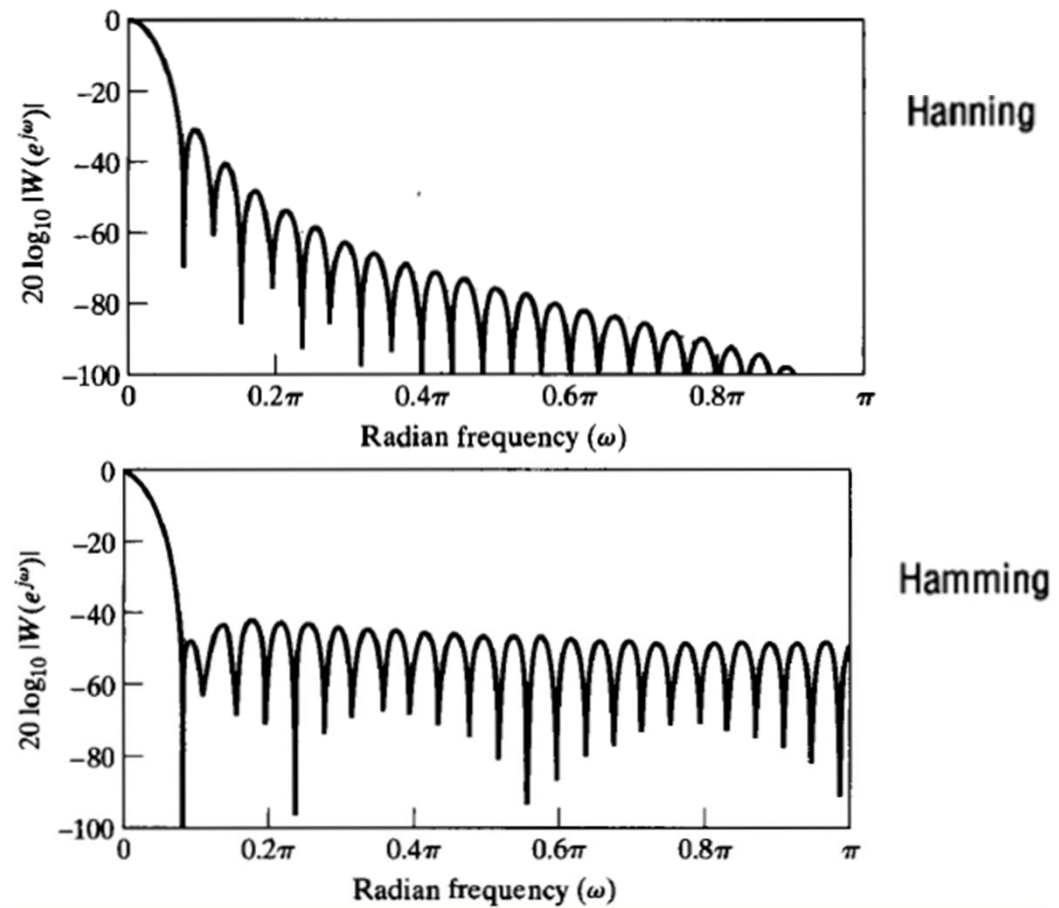
Rectangular.



Bartlett

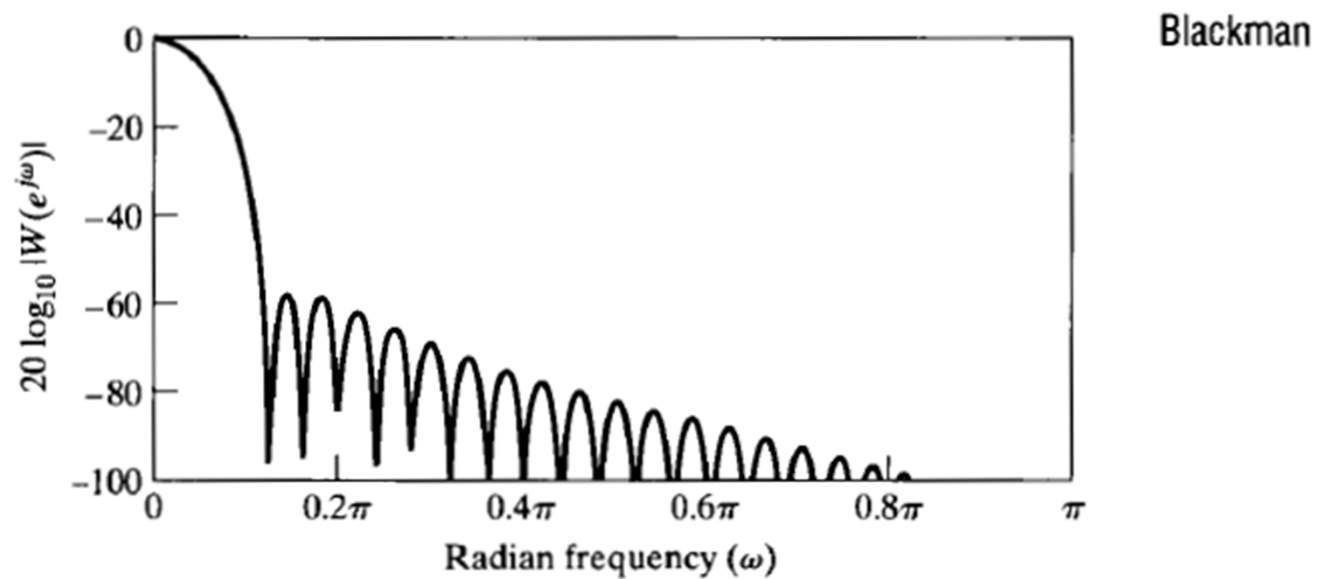
Filter Design

- Spectra



Filter Design

- Spectra



Filter Design

- Summary of windows

TABLE **COMPARISON**

Type of Window	Peak Side-Lobe Amplitude (Relative)	Approximate Width of Main Lobe
Rectangular	-13	$4\pi/(M + 1)$
Bartlett	-25	$8\pi/M$
Hanning	-31	$8\pi/M$
Hamming	-41	$8\pi/M$
Blackman	-57	$12\pi/M$

- Cost of lower sidelobe level: wider mainlobe