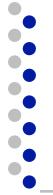


ESE 531: Digital Signal Processing

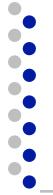
Lec 16: March 26, 2020

Design of FIR Filters, Optimal Filter Design



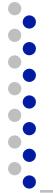
Linear Filter Design

- ❑ Used to be an art
 - Now, lots of tools to design optimal filters
- ❑ For DSP there are two common classes
 - Infinite impulse response IIR
 - Finite impulse response FIR
- ❑ Both classes use finite order of parameters for design
- ❑ Today we will focus on FIR designs



What is a Linear Filter?

- Attenuates certain frequencies
 - Passes certain frequencies
 - Affects both phase and magnitude
-
- IIR (Oppenheim 7.2-7.4)
 - Mostly non-linear phase response
 - Could be linear over a range of frequencies
 - FIR
 - Much easier to control the phase
 - Both non-linear and linear phase



Impulse Invariance

- Let,

$$h[n] = T h_c(nT)$$

$$H_c(j\Omega) = 0, \quad |\Omega| \geq \pi / T$$

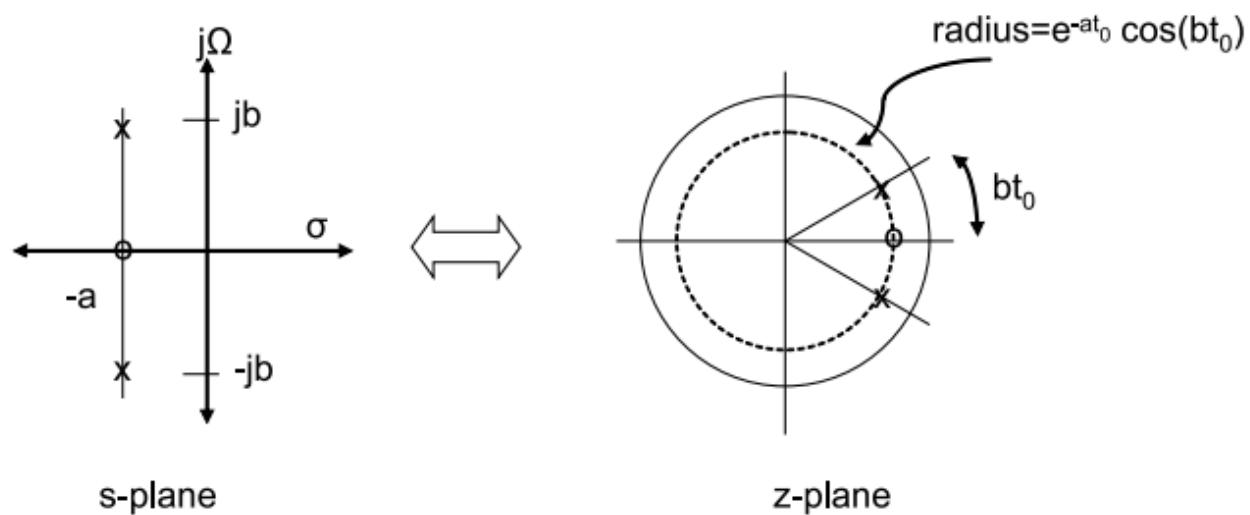
- If sampling at Nyquist Rate then

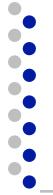
$$\Omega = \omega T$$

$$H(e^{j\omega}) = \sum_{k=-\infty}^{\infty} H_c \left[j \left(\frac{\omega}{T} - \frac{2\pi k}{T} \right) \right]$$

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

S-Plane Mapping to Z-Plane



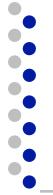


Bilinear Transformation

- The technique uses an algebraic transformation between the variables s and z that maps the entire $j\Omega$ -axis in the s -plane to one revolution of the unit circle in the z -plane.

$$s = \frac{2}{T_d} \left(\frac{1 - z^{-1}}{1 + z^{-1}} \right);$$

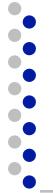
$$H(z) = H_c \left(\frac{2}{T_d} \left(\frac{1 - z^{-1}}{1 + z^{-1}} \right) \right).$$



Bilinear Transformation

$$s = \frac{2}{T_d} \left(\frac{1 - z^{-1}}{1 + z^{-1}} \right);$$

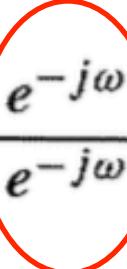
- Substituting $s = j\Omega$ and $z = e^{j\omega}$
 - Just looking at the $j\Omega$ axis in s-plane (i.e the unit circle in the z-plane)

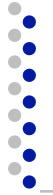


Bilinear Transformation

$$s = \frac{2}{T_d} \left(\frac{1 - z^{-1}}{1 + z^{-1}} \right);$$

- Substituting $s = j\Omega$ and $z = e^{j\omega}$

$$s = \frac{2}{T_d} \left(\frac{1 - e^{-j\omega}}{1 + e^{-j\omega}} \right),$$




Bilinear Transformation

$$s = \frac{2}{T_d} \left(\frac{1 - z^{-1}}{1 + z^{-1}} \right);$$

- Substituting $s = j\Omega$ and $z = e^{j\omega}$

$$s = \frac{2}{T_d} \left(\frac{1 - e^{-j\omega}}{1 + e^{-j\omega}} \right),$$

Euler's formula

$$\frac{2}{T_d} \left[\frac{2e^{-j\omega/2}(j \sin \omega/2)}{2e^{-j\omega/2}(\cos \omega/2)} \right]$$



Bilinear Transformation

$$s = \frac{2}{T_d} \left(\frac{1 - z^{-1}}{1 + z^{-1}} \right);$$

- Substituting $s = j\Omega$ and $z = e^{j\omega}$

$$s = \frac{2}{T_d} \left(\frac{1 - e^{-j\omega}}{1 + e^{-j\omega}} \right),$$

$$s = j\Omega = \frac{2}{T_d} \left[\frac{2e^{-j\omega/2}(j \sin \omega/2)}{2e^{-j\omega/2}(\cos \omega/2)} \right] = \frac{2j}{T_d} \tan(\omega/2).$$



Bilinear Transformation

$$s = \frac{2}{T_d} \left(\frac{1 - z^{-1}}{1 + z^{-1}} \right);$$

- Substituting $s = j\Omega$ and $z = e^{j\omega}$

$$s = \frac{2}{T_d} \left(\frac{1 - e^{-j\omega}}{1 + e^{-j\omega}} \right),$$

$$s = j\Omega = \frac{2}{T_d} \left[\frac{2e^{-j\omega/2}(j \sin \omega/2)}{2e^{-j\omega/2}(\cos \omega/2)} \right] = \frac{2j}{T_d} \tan(\omega/2).$$

$$\Omega = \frac{2}{T_d} \tan(\omega/2),$$

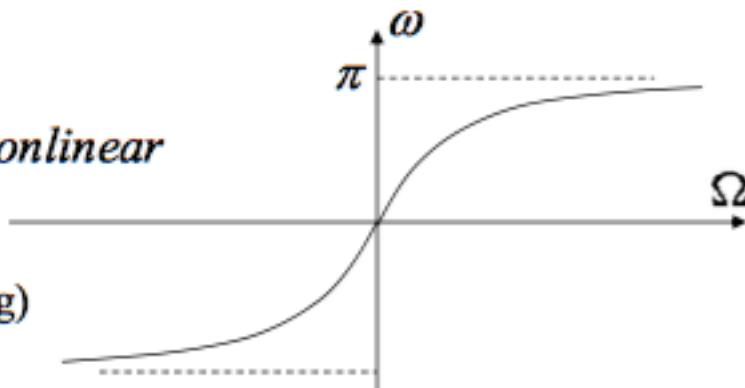
~~$\Omega = \omega T$~~

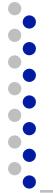
Bilinear Transformation

$$\Omega = \frac{2}{T_d} \tan(\omega/2),$$

$$\omega = 2 \arctan(\Omega T_d / 2).$$

No aliasing, but mapping nonlinear
(Impulse invariance:
linear mapping, but with aliasing)

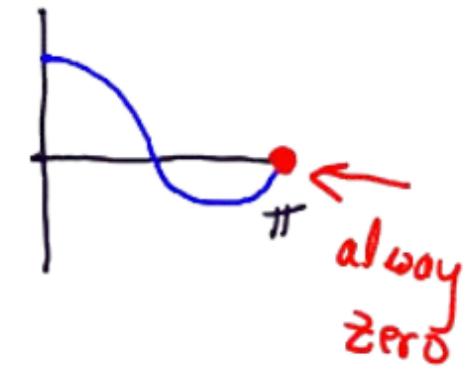
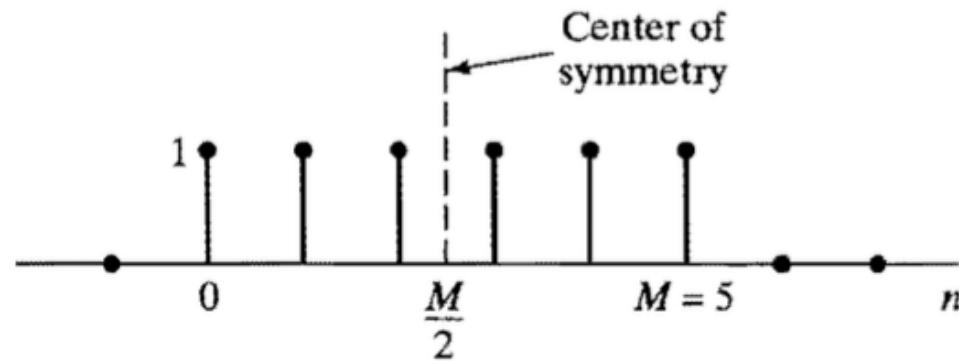
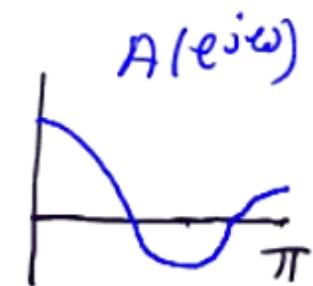
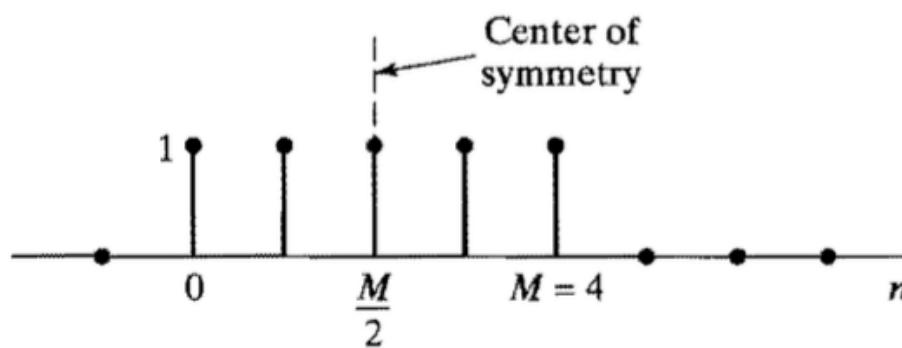




What is a Linear Filter?

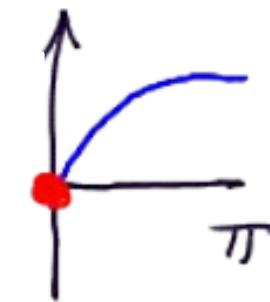
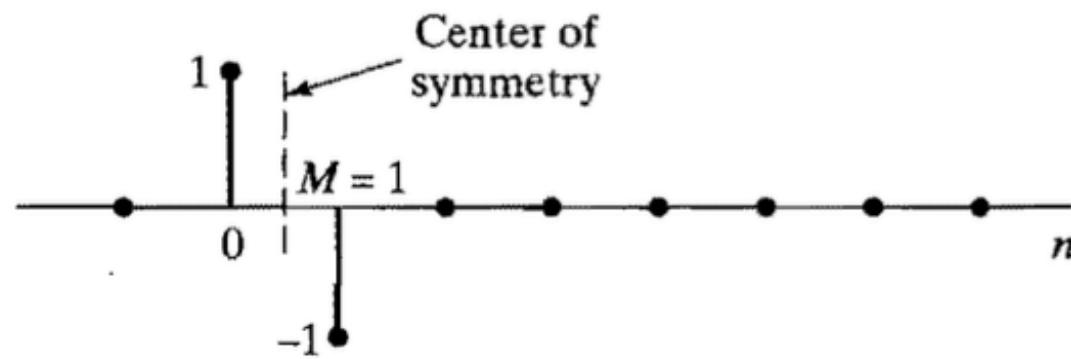
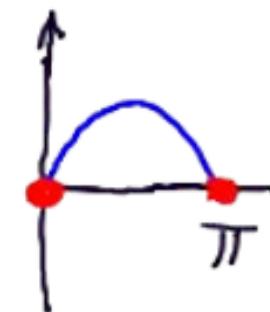
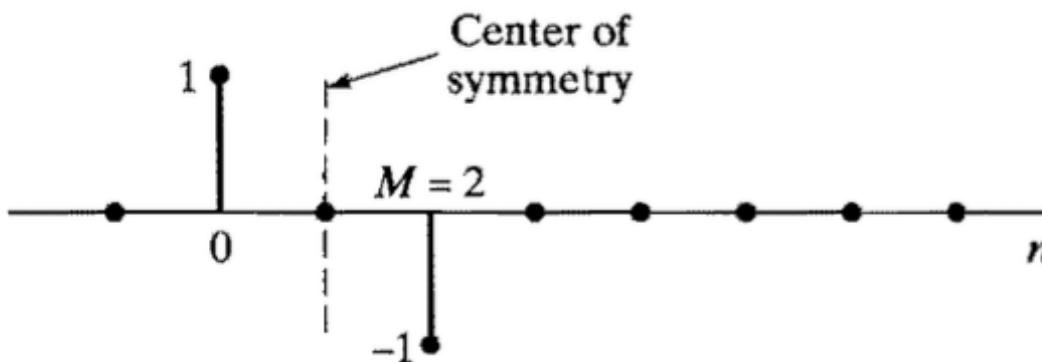
- Attenuates certain frequencies
 - Passes certain frequencies
 - Affects both phase and magnitude
-
- IIR
 - Mostly non-linear phase response
 - Could be linear over a range of frequencies
 - FIR
 - Much easier to control the phase
 - Both non-linear and linear phase

FIR GLP: Type I and II



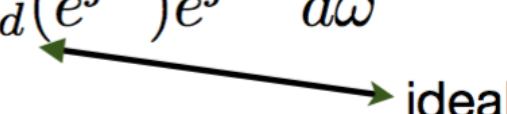


FIR GLP: Type III and IV



FIR Design by Windowing

- Given desired frequency response, $H_d(e^{j\omega})$, find an impulse response

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


- Obtain the M^{th} order causal FIR filter by truncating/windowing it

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



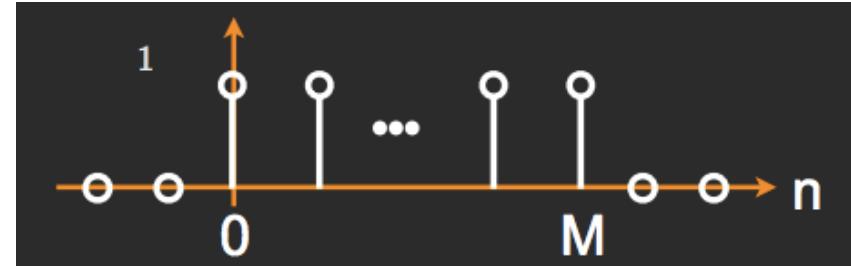
FIR Design by Windowing

- ❑ With multiplication in time property,

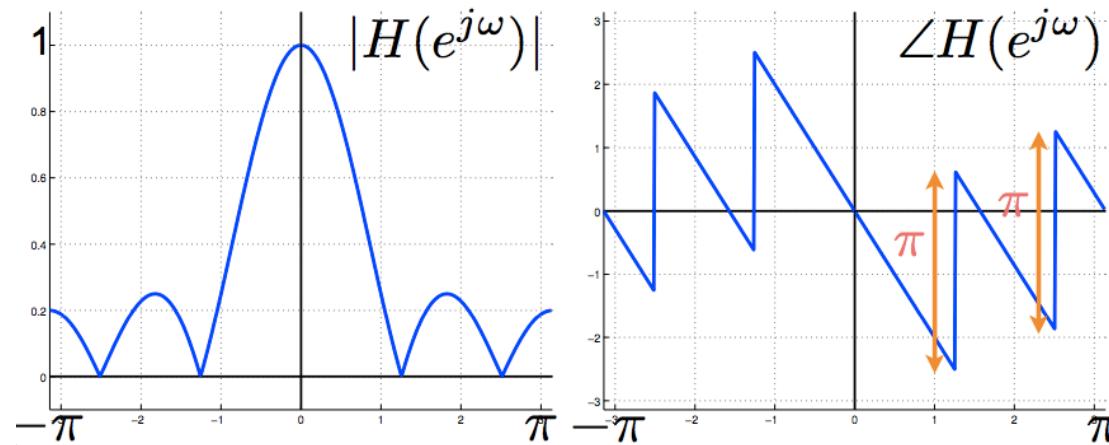
$$H(e^{j\omega}) = H_d(e^{j\omega}) * W(e^{j\omega})$$



Boxcar Window



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



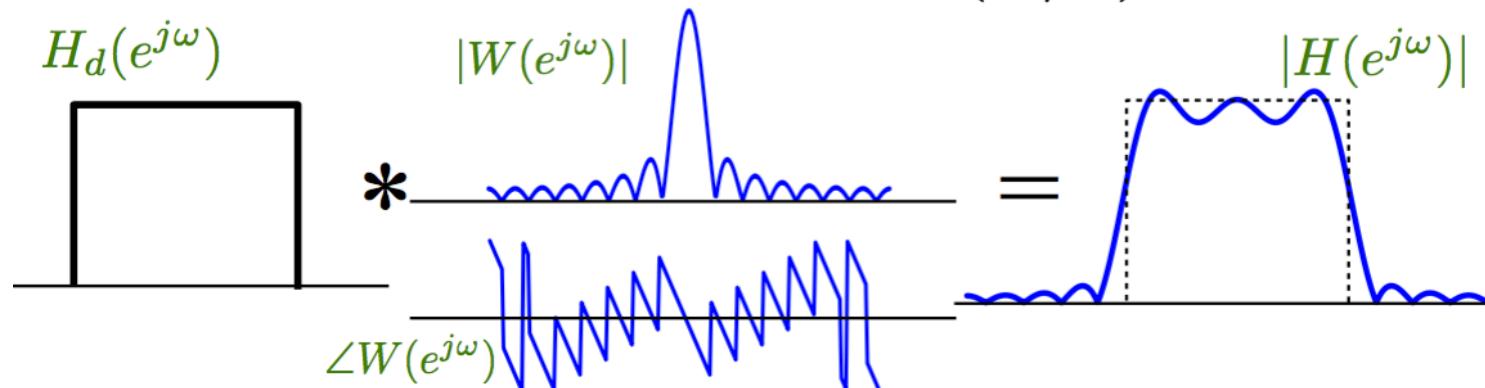
FIR Design by Windowing

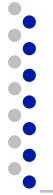
- With multiplication in time property,

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

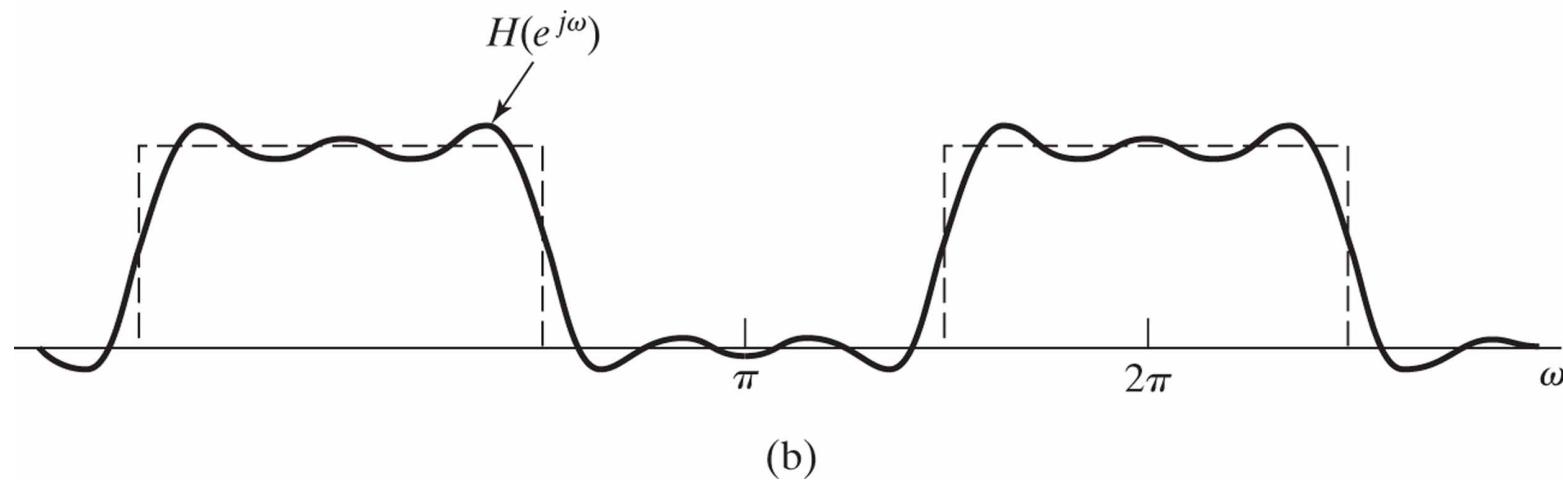
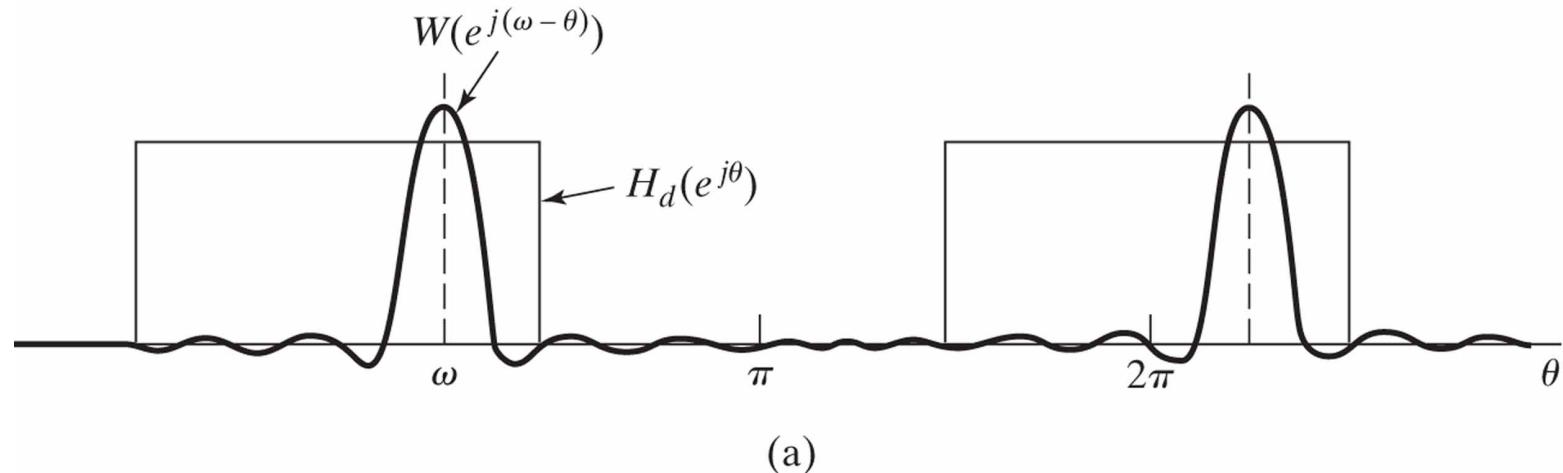
- For Boxcar (rectangular) window

$$W(e^{j\omega}) = e^{-j\omega \frac{M}{2}} \frac{\sin(w(M+1)/2)}{\sin(w/2)}$$

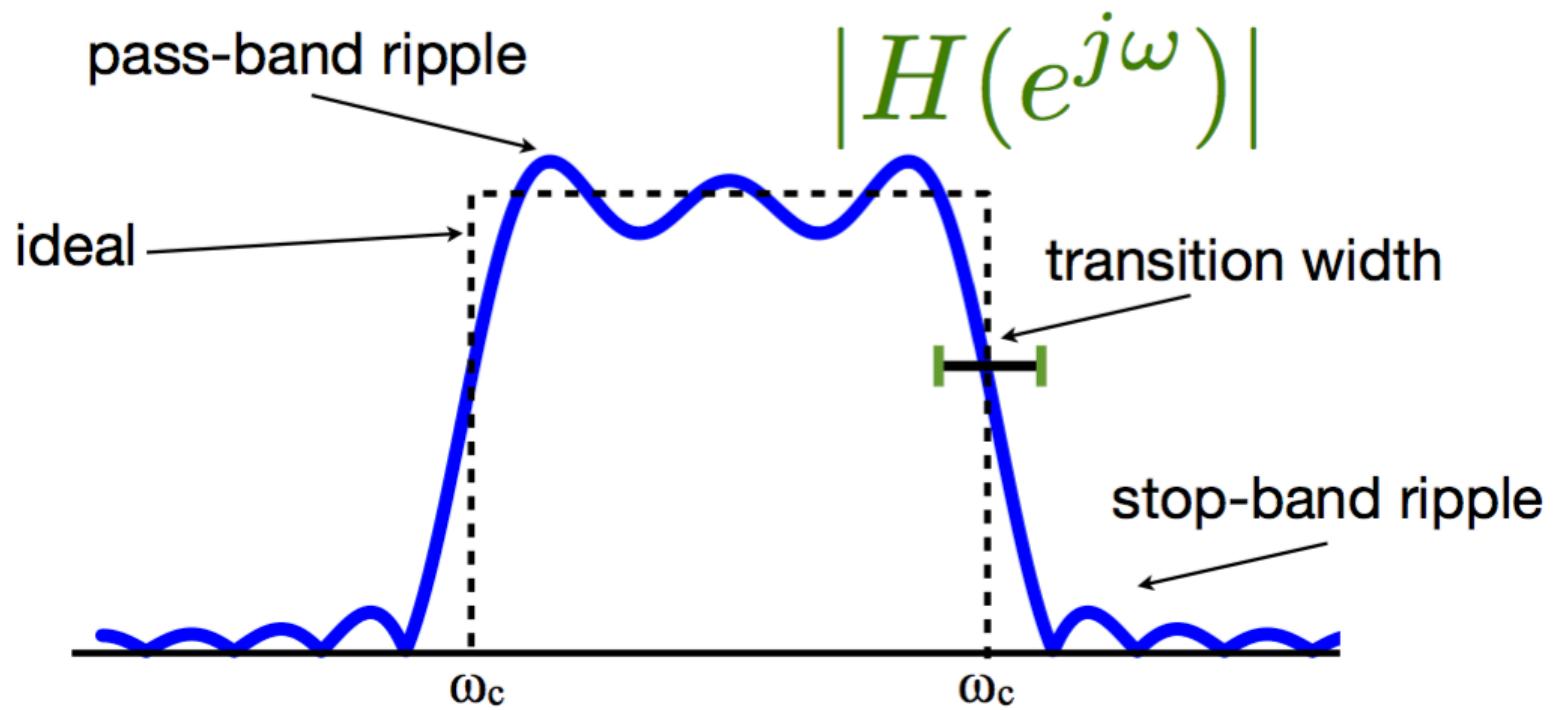




FIR Design by Windowing



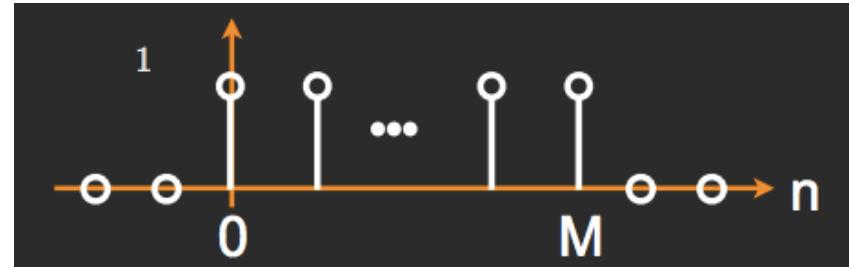
FIR Design by Windowing





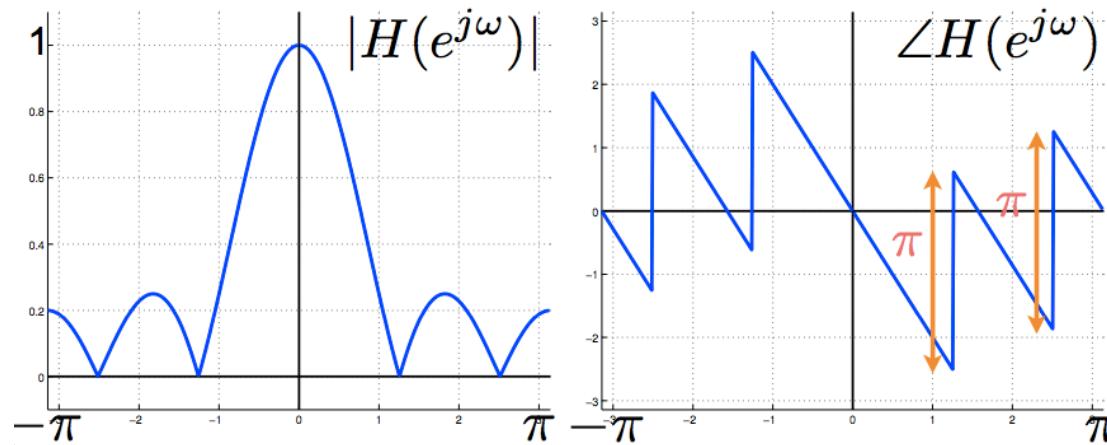
Boxcar Window

Time



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

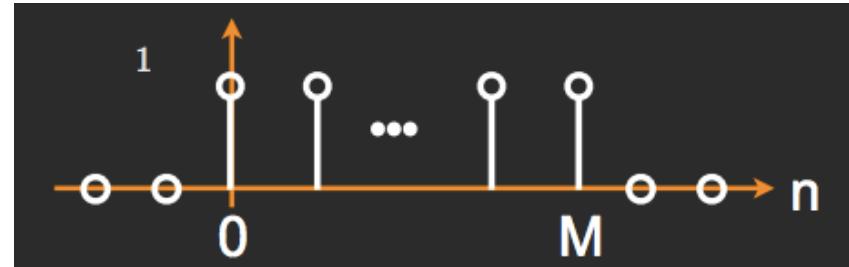
Frequency





Boxcar Window

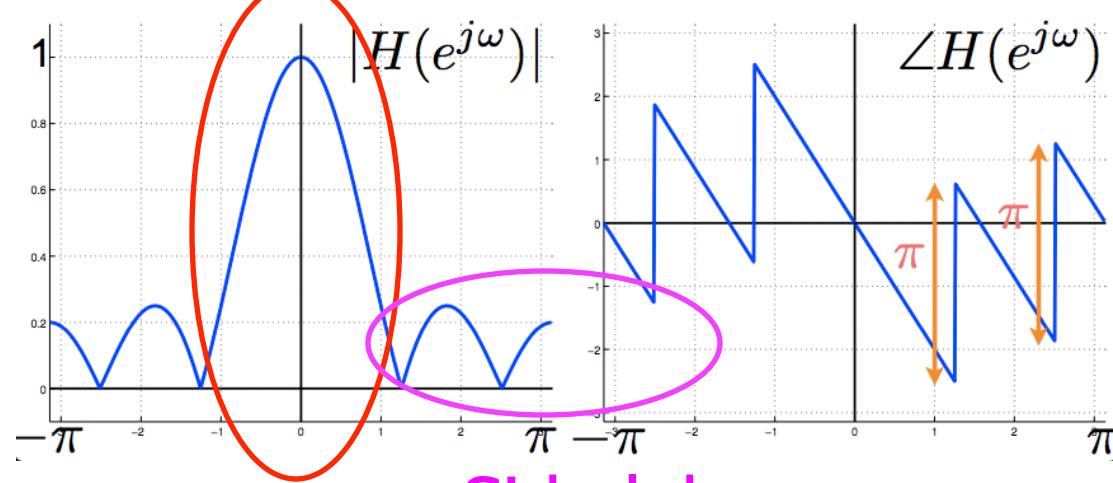
Time



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

Main lobe

Frequency





Ideal “Window”?

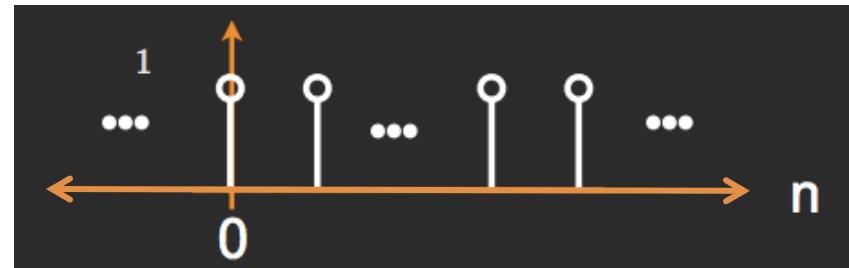
Time

Frequency



Ideal “Window”?

Time



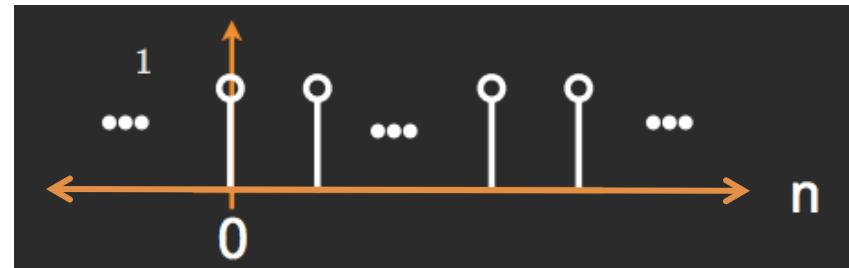
$$w[n] = 1 \Leftrightarrow W(e^{j\omega}) = 2\pi\delta[\omega]$$

Frequency



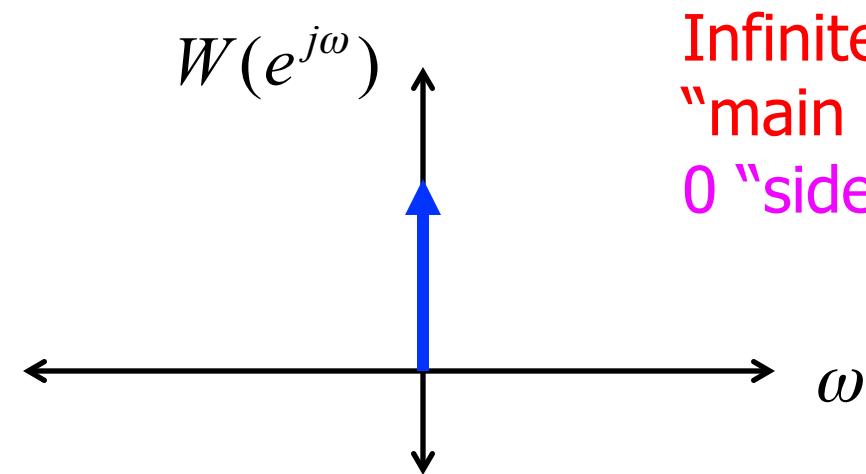
Ideal “Window”?

Time



$$w[n] = 1 \Leftrightarrow W(e^{j\omega}) = 2\pi\delta[\omega]$$

Frequency



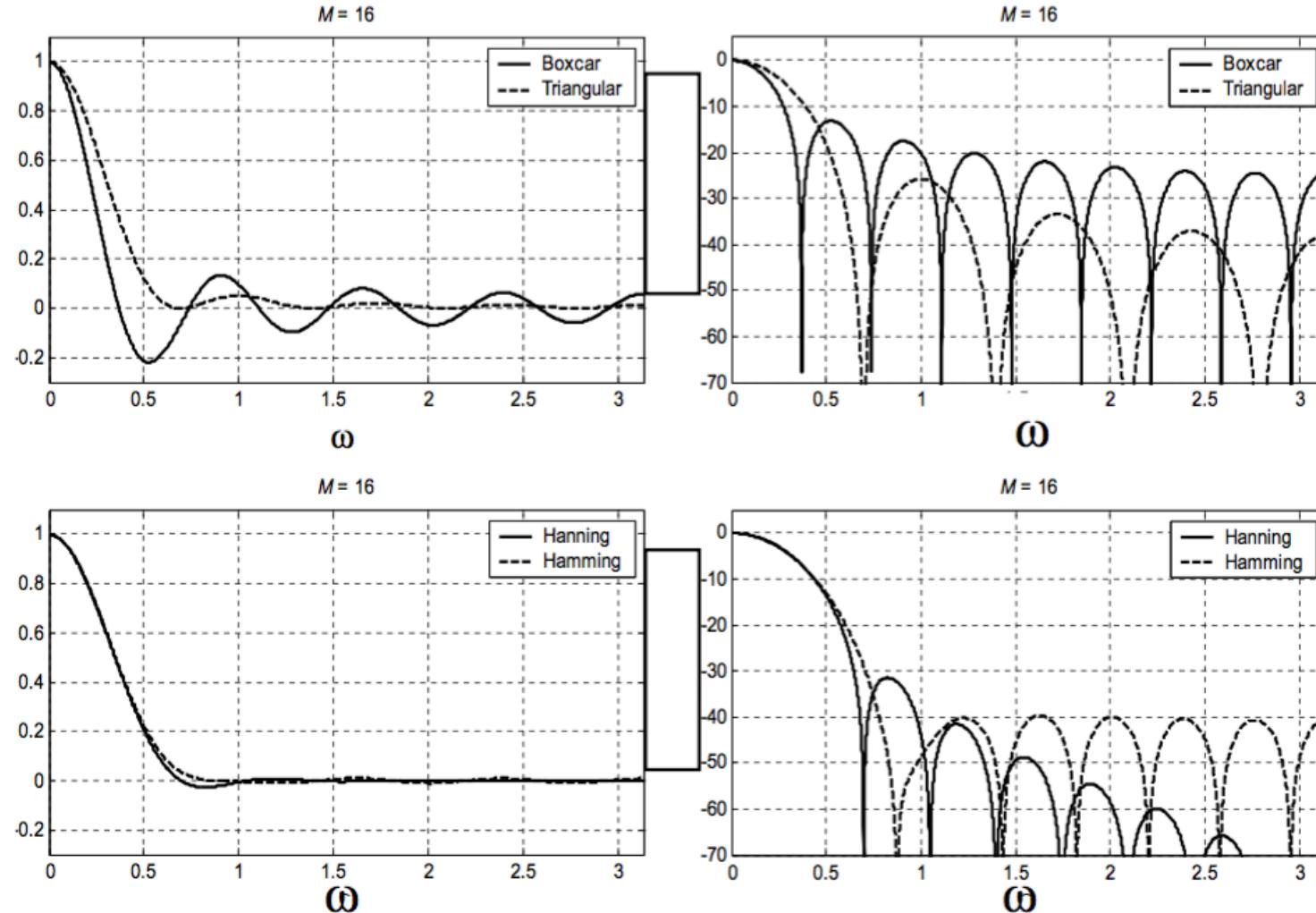
Ininitely narrow
“main lobe” width
0 “side lobe” height



Tapered Windows

Name(s)	Definition	MATLAB Command	Graph ($M = 8$)
Hann	$w[n] = \begin{cases} \frac{1}{2} \left[1 + \cos\left(\frac{\pi n}{M/2}\right) \right] & n \leq M/2 \\ 0 & n > M/2 \end{cases}$	<code>hann(M+1)</code>	
Hanning	$w[n] = \begin{cases} \frac{1}{2} \left[1 + \cos\left(\frac{\pi n}{M/2+1}\right) \right] & n \leq M/2 \\ 0 & n > M/2 \end{cases}$	<code>hanning(M+1)</code>	
Hamming	$w[n] = \begin{cases} 0.54 + 0.46 \cos\left(\frac{\pi n}{M/2}\right) & n \leq M/2 \\ 0 & n > M/2 \end{cases}$	<code>hamming(M+1)</code>	

Tradeoff – Ripple vs. Transition Width



Commonly Used Windows

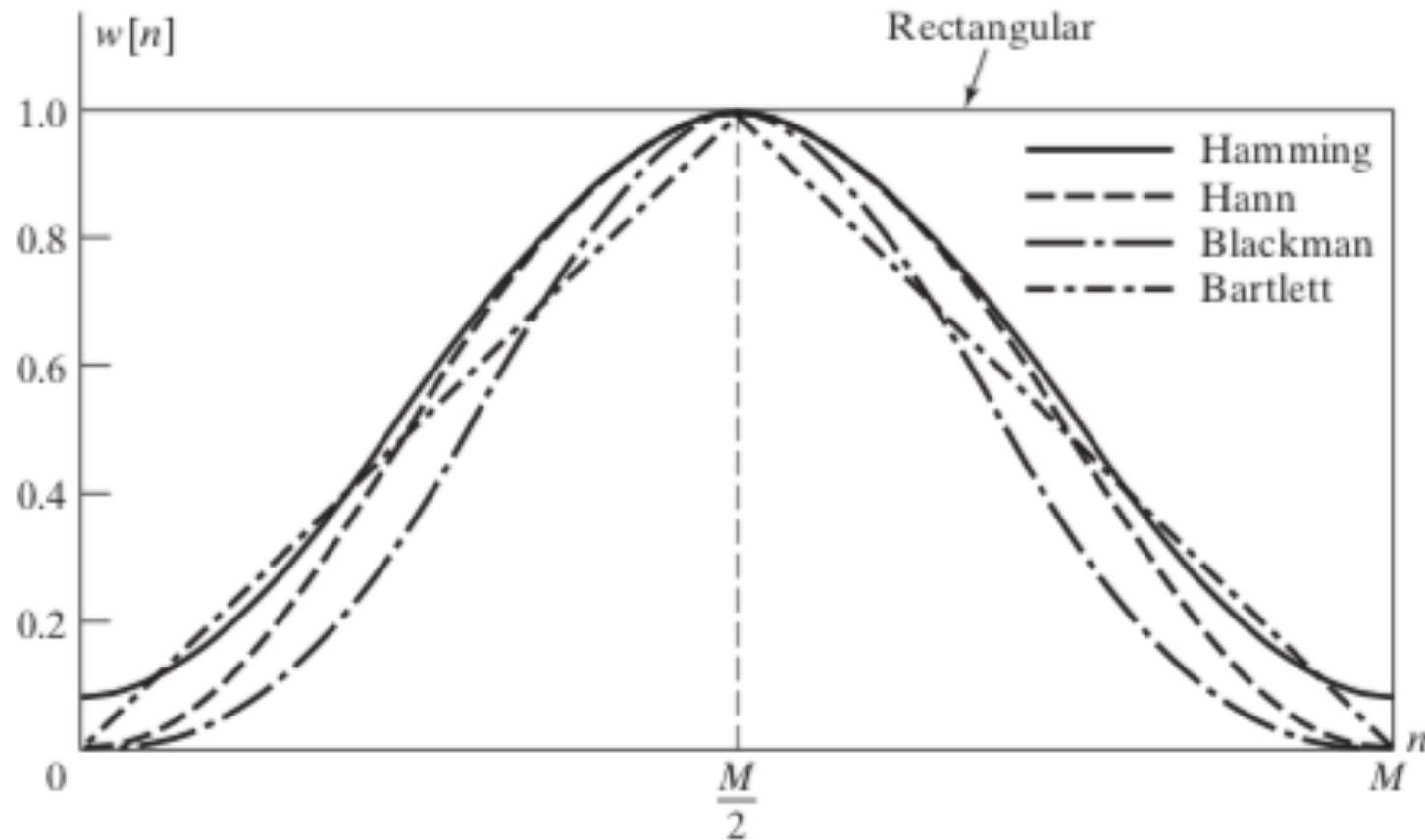
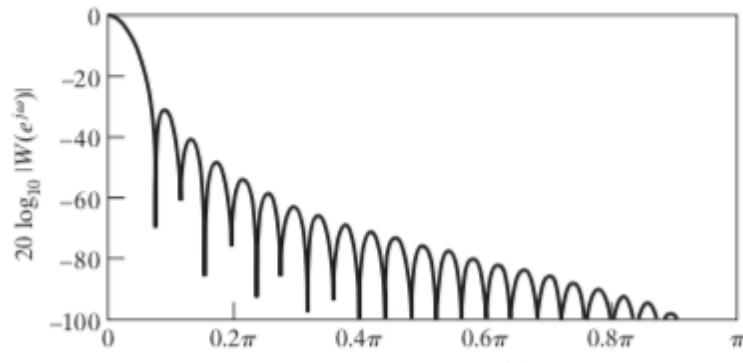


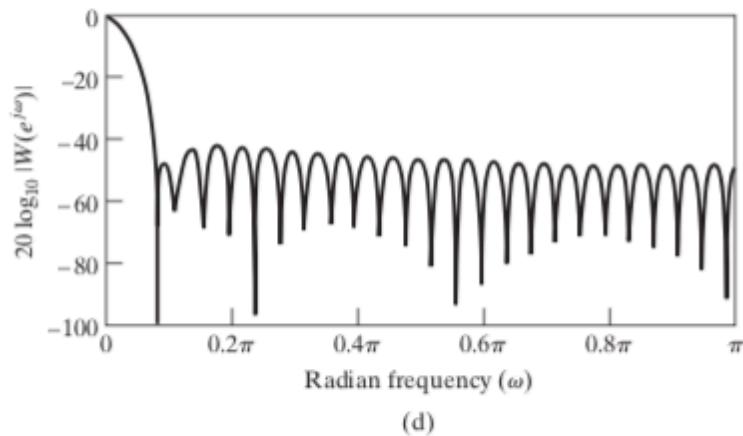
Figure 7.29 Commonly used windows.



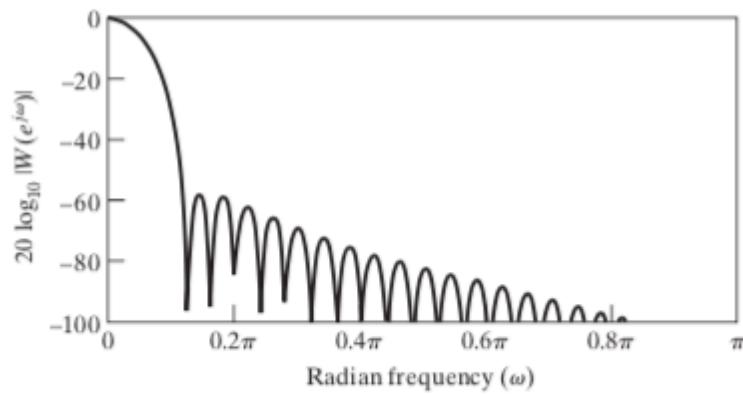
Hann

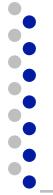


Hamming



Blackman



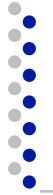


Kaiser Window

- ☐ Near optimal window quantified as the window maximally concentrated around $\omega=0$

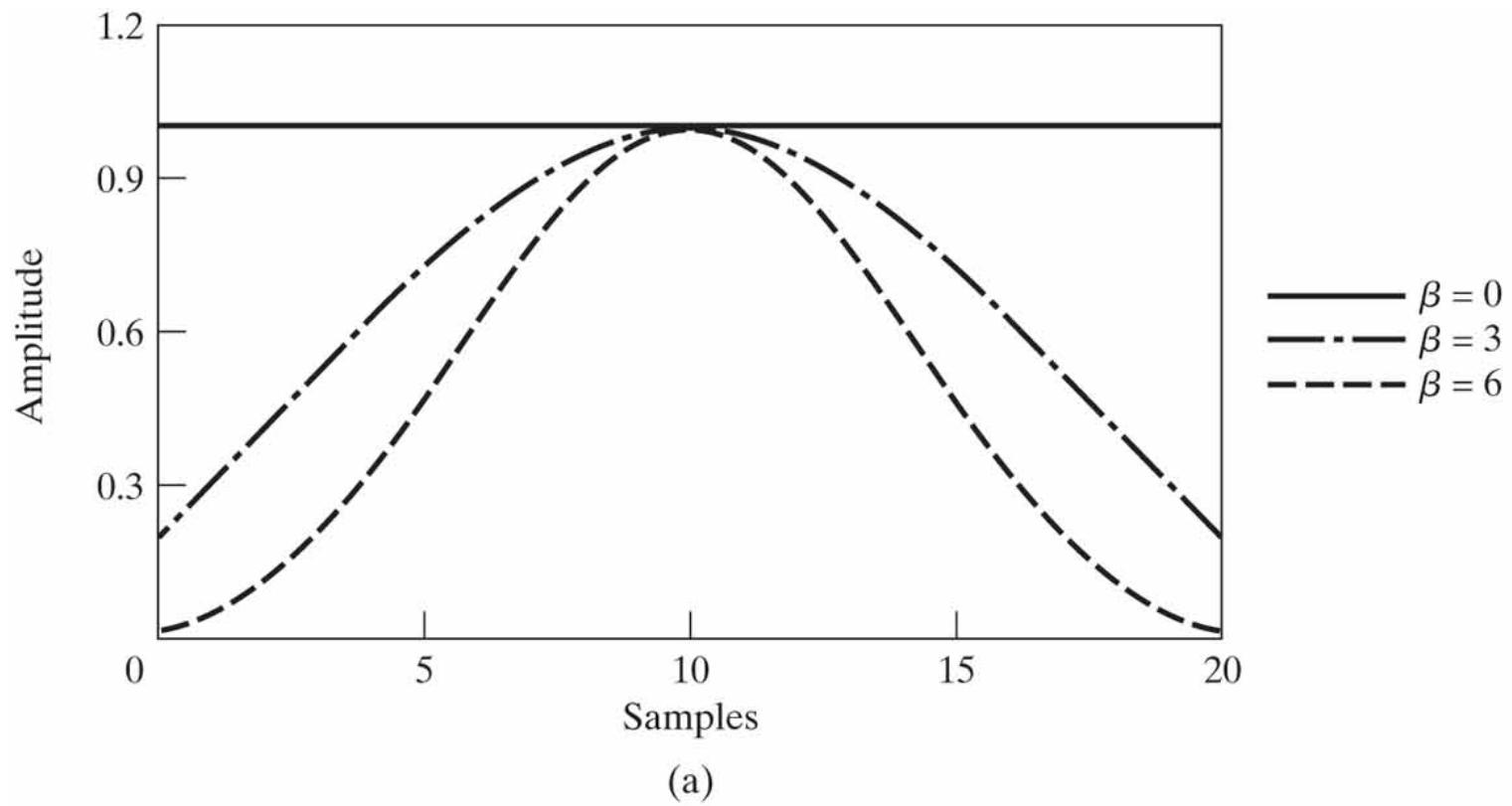
$$w[n] = \begin{cases} \frac{I_0[\beta(1 - [(n - \alpha)/\alpha]^2)^{1/2}]}{I_0(\beta)}, & 0 \leq n \leq M, \\ 0, & \text{otherwise,} \end{cases}$$

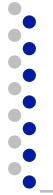
- ☐ Two parameters – M and β
- ☐ $\alpha = M/2$
- ☐ $I_0(x)$ – zeroth order Bessel function of the first kind



Kaiser Window

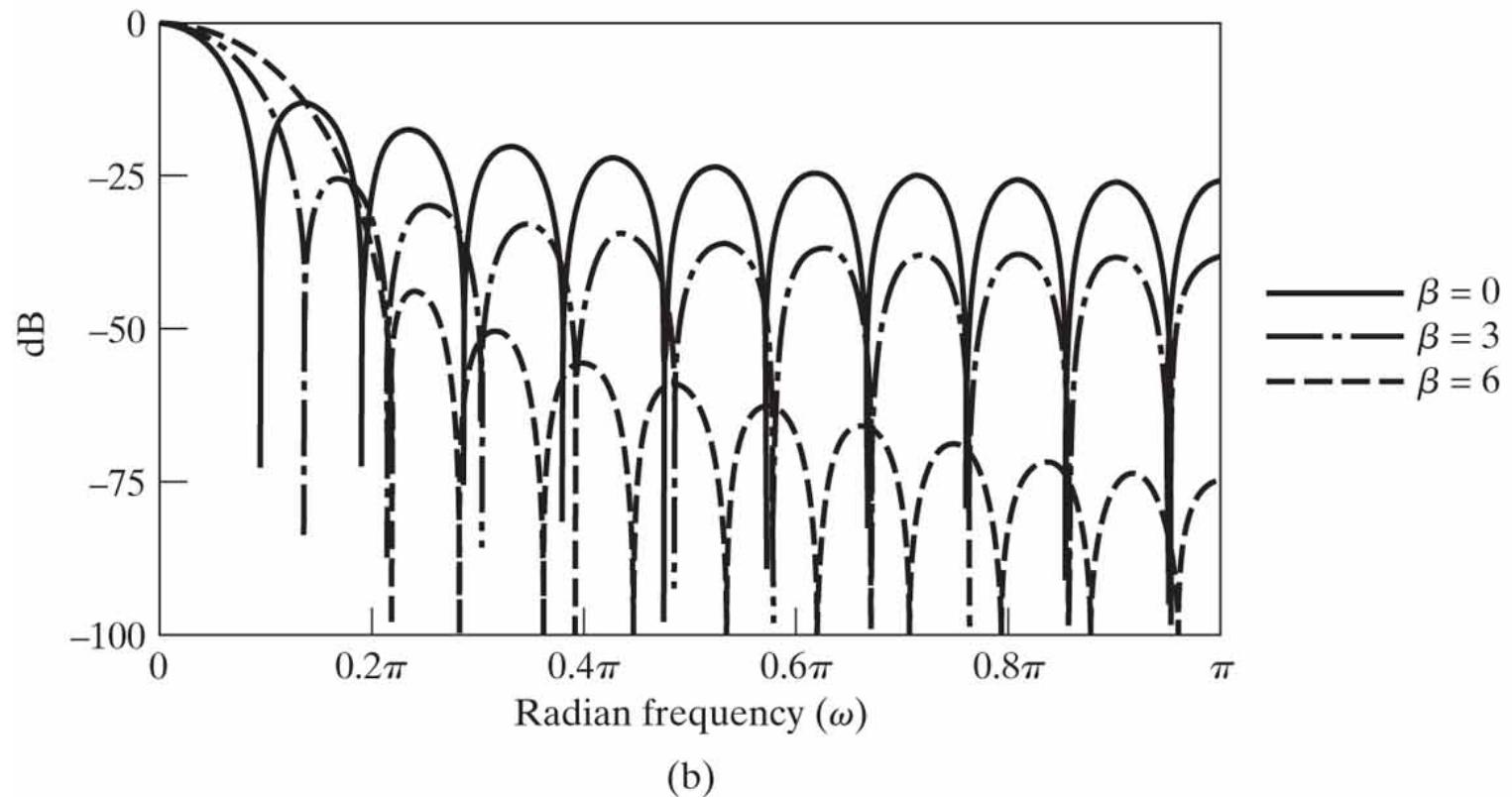
□ $M=20$





Kaiser Window

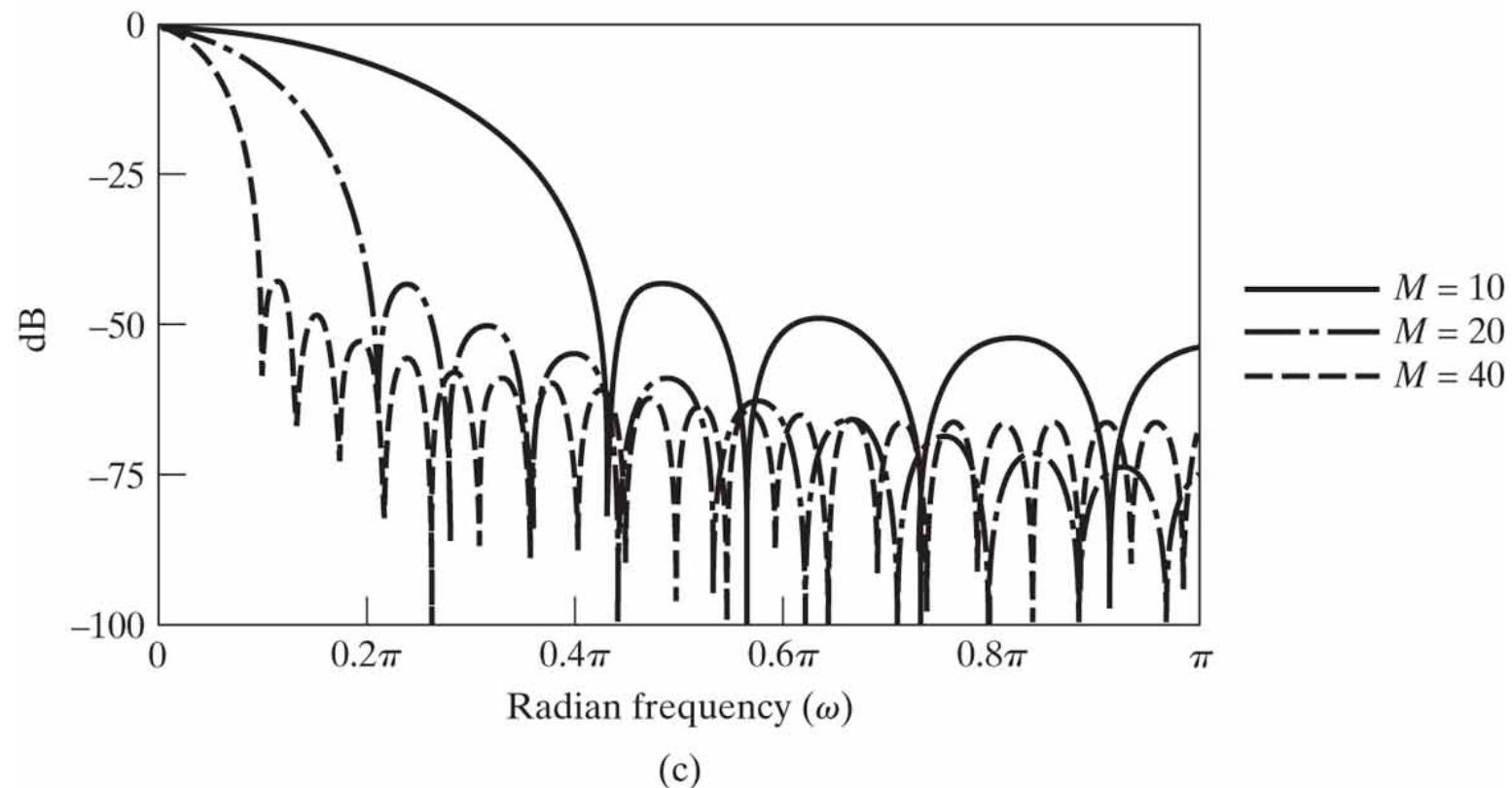
□ M=20



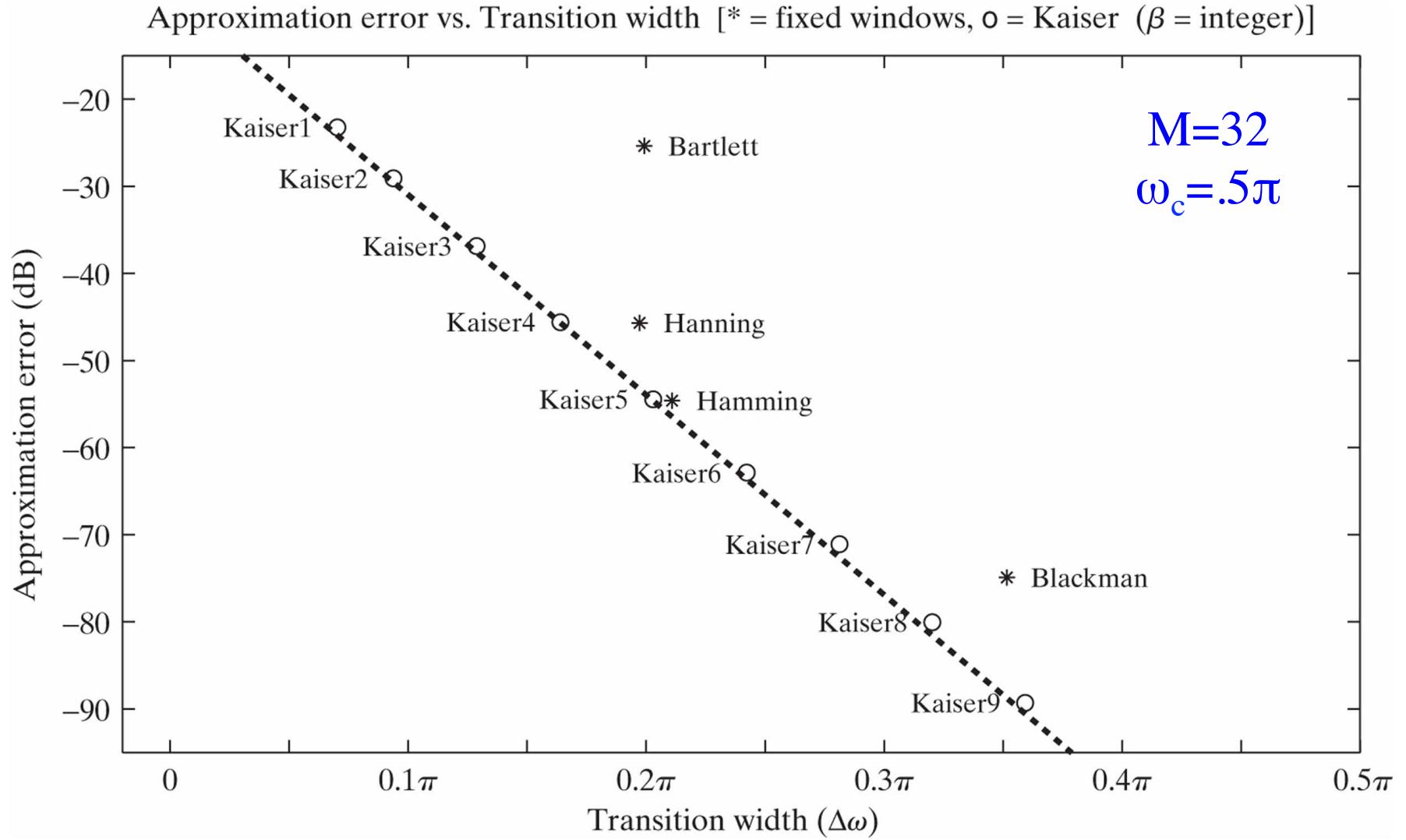


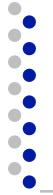
Kaiser Window

◻ $\beta = 6$



Approximation Error



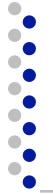


FIR Filter Design

- Choose a desired frequency response $H_d(e^{j\omega})$
 - non causal (zero-delay), and infinite imp. response
 - If derived from C.T, choose T and use:

$$H_d(e^{j\omega}) = H_c(j \frac{\Omega}{T})$$

- Window:
 - Length $M+1 \Leftrightarrow$ affects transition width
 - Type of window \Leftrightarrow transition-width/ ripple



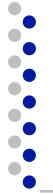
FIR Filter Design

- Choose a desired frequency response $H_d(e^{j\omega})$
 - non causal (zero-delay), and infinite imp. response
 - If derived from C.T, choose T and use:

$$H_d(e^{j\omega}) = H_c(j \frac{\Omega}{T})$$

- Window:
 - Length $M+1 \Leftrightarrow$ affects transition width
 - Type of window \Leftrightarrow transition-width/ ripple
 - Modulate to shift impulse response
 - Force causality

$$H_d(e^{j\omega})e^{-j\omega \frac{M}{2}}$$



FIR Filter Design

- Determine truncated impulse response $h_1[n]$

$$h_1[n] = \begin{cases} \frac{1}{2\pi} \int_{-\pi}^{\pi} H_d(e^{j\omega}) e^{-j\omega \frac{M}{2}} e^{j\omega n} & 0 \leq n \leq M \\ 0 & \text{otherwise} \end{cases}$$

- Apply window

$$h_w[n] = w[n]h_1[n]$$

- Check:

- Compute $H_w(e^{j\omega})$, if does not meet specs increase M or change window

Example: FIR Low-Pass Filter Design

$$H_d(e^{j\omega}) = \begin{cases} 1 & |\omega| \leq \omega_c \\ 0 & \text{otherwise} \end{cases}$$

Choose M \Rightarrow Window length and set

$$H_1(e^{j\omega}) = H_d(e^{j\omega})e^{-j\omega \frac{M}{2}}$$

$$h_1[n] = \begin{cases} \frac{\sin(\omega_c(n-M/2))}{\pi(n-M/2)} & 0 \leq n \leq M \\ 0 & \text{otherwise} \end{cases}$$

$\frac{\omega_c}{\pi} \text{sinc}\left(\frac{\omega_c}{\pi}(n - M/2)\right)$

Example: FIR Low-Pass Filter Design

- The result is a windowed sinc function

$$h_w[n] = w[n]h_1[n]$$

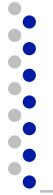
$$\frac{\omega_c}{\pi} \text{sinc}\left(\frac{\omega_c}{\pi}(n - M/2)\right)$$

- High Pass Design:

- Design low pass
- Transform to $h_w[n](-1)^n = h_w[n]e^{j\pi n}$

- General bandpass

- Transform to $2h_w[n]\cos(\omega_0 n)$ or $2h_w[n]\sin(\omega_0 n)$



Design through FFT

- ❑ To design order M filter:
- ❑ Over-Sample/discretize the frequency response at P points where $P \gg M$ ($P=15M$ is good)

$$H_1(e^{j\omega_k}) = H_d(e^{j\omega_k})e^{-j\omega_k \frac{M}{2}}$$

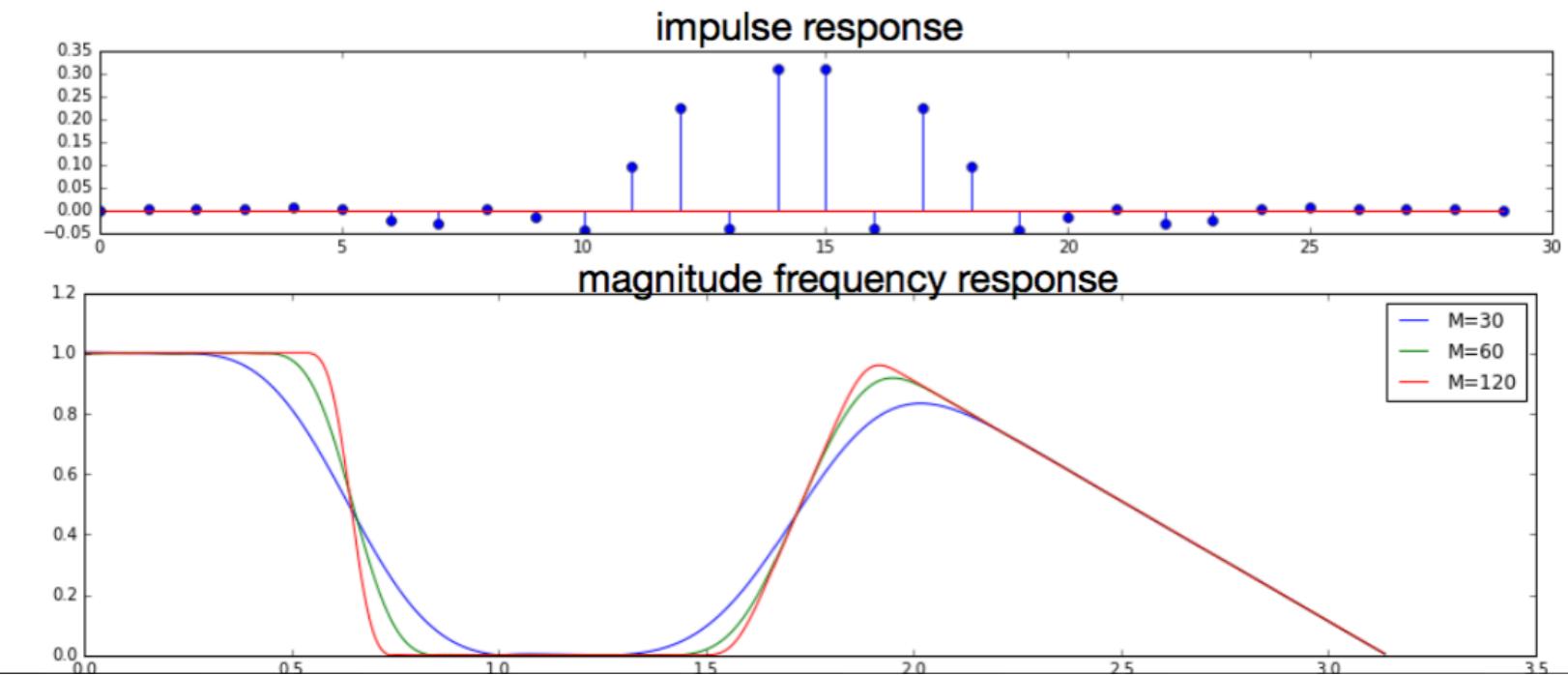
- ❑ Sampled at: $\omega_k = k \frac{2\pi}{P}$ $|k = [0, \dots, P - 1]$
- ❑ Compute $h_1[n] = \text{IDFT}_P(H_1[k])$
- ❑ Apply $M+1$ length window:

$$h_w[n] = w[n]h_1[n]$$



Example

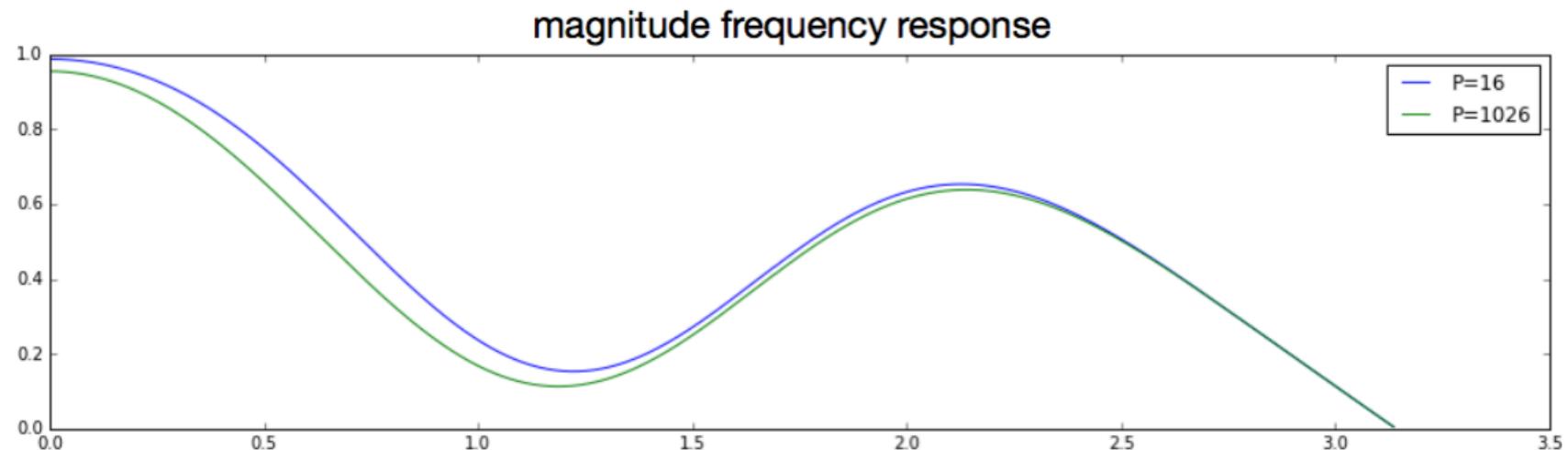
- `signal.firwin2(M+1,omega_vec/pi, amp_vec)`
- `taps1 = signal.firwin2(30, [0.0,0.2,0.21,0.5, 0.6, 1.0], [1.0, 1.0, 0.0,0.0,1.0,0.0])`

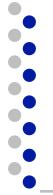




Example

- For $M+1=14$
 - $P = 16$ and $P = 1026$





Optimal Filter Design

- Window method

- Design Filters heuristically using windowed sinc functions
- Choose order and window type
- Check DTFT to see if filter specs are met

- Optimal design

- Design a filter $h[n]$ with $H(e^{j\omega})$
- Approximate $H_d(e^{j\omega})$ with some optimality criteria - or satisfies specs.



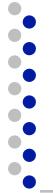
Mathematical Optimization

(mathematical) optimization problem

$$\begin{aligned} & \text{minimize} && f_0(x) \\ & \text{subject to} && f_i(x) \leq b_i, \quad i = 1, \dots, m \end{aligned}$$

- $x = (x_1, \dots, x_n)$: optimization variables
- $f_0 : \mathbf{R}^n \rightarrow \mathbf{R}$: objective function
- $f_i : \mathbf{R}^n \rightarrow \mathbf{R}, i = 1, \dots, m$: constraint functions

optimal solution x^* has smallest value of f_0 among all vectors that satisfy the constraints



Examples

portfolio optimization

- variables: amounts invested in different assets
- constraints: budget, max./min. investment per asset, minimum return
- objective: overall risk or return variance

device sizing in electronic circuits

- variables: device widths and lengths
- constraints: manufacturing limits, timing requirements, maximum area
- objective: power consumption

data fitting

- variables: model parameters
- constraints: prior information, parameter limits
- objective: measure of misfit or prediction error



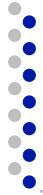
Solving Optimization Problems

general optimization problem

- very difficult to solve
- methods involve some compromise, *e.g.*, very long computation time, or not always finding the solution

exceptions: certain problem classes can be solved efficiently and reliably

- least-squares problems
- linear programming problems
- convex optimization problems



Least-Squares Optimization

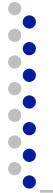
$$\text{minimize} \quad \|Ax - b\|_2^2$$

solving least-squares problems

- analytical solution: $x^* = (A^T A)^{-1} A^T b$
- reliable and efficient algorithms and software
- computation time proportional to $n^2 k$ ($A \in \mathbb{R}^{k \times n}$); less if structured
- a mature technology

using least-squares

- least-squares problems are easy to recognize
- a few standard techniques increase flexibility (e.g., including weights, adding regularization terms)



Linear Programming

$$\begin{aligned} & \text{minimize} && c^T x \\ & \text{subject to} && a_i^T x \leq b_i, \quad i = 1, \dots, m \end{aligned}$$

solving linear programs

- no analytical formula for solution
- reliable and efficient algorithms and software
- computation time proportional to n^2m if $m \geq n$; less with structure
- a mature technology

using linear programming

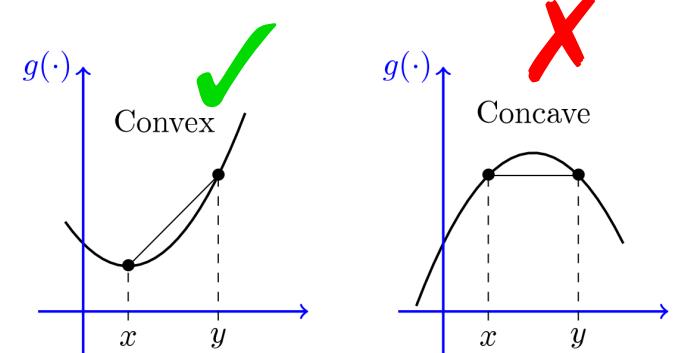
- not as easy to recognize as least-squares problems
- a few standard tricks used to convert problems into linear programs
(e.g., problems involving ℓ_1 - or ℓ_∞ -norms, piecewise-linear functions)



Convex Optimization

$$\begin{aligned} & \text{minimize} && f_0(x) \\ & \text{subject to} && f_i(x) \leq b_i, \quad i = 1, \dots, m \end{aligned}$$

- objective and constraint functions are convex:

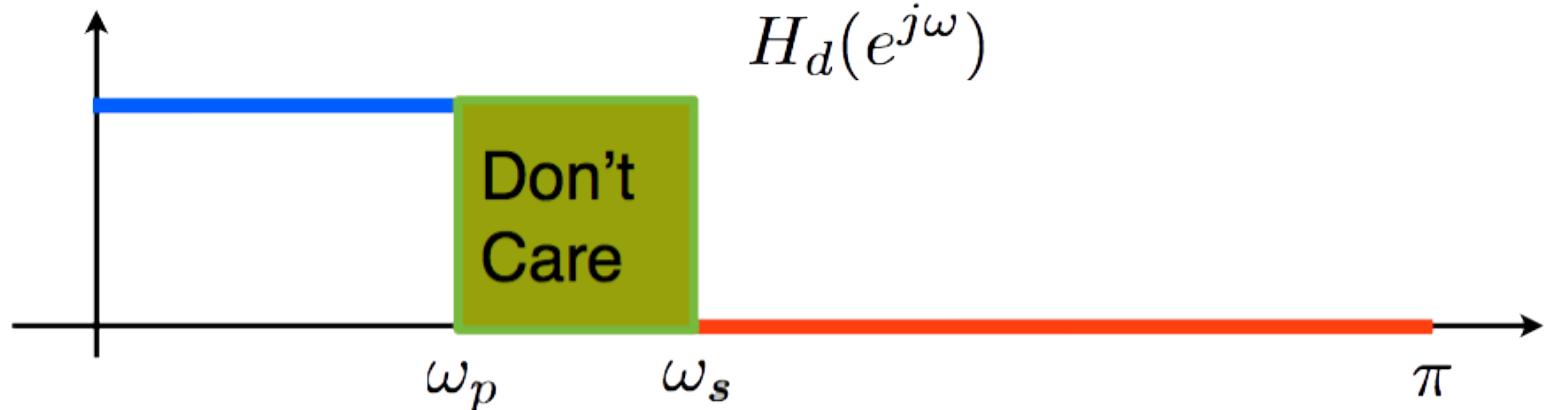


$$f_i(\alpha x + \beta y) \leq \alpha f_i(x) + \beta f_i(y)$$

if $\alpha + \beta = 1$, $\alpha \geq 0$, $\beta \geq 0$

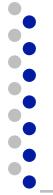
- includes least-squares problems and linear programs as special cases

Optimality – Least Squares



- Least Squares:

$$\text{minimize} \quad \int_{\omega \in \text{care}} |H(e^{j\omega}) - H_d(e^{j\omega})|^2 d\omega$$



Design Through Optimization

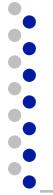
- ❑ Idea: Sample/discretize the frequency response

$$H(e^{j\omega}) \Rightarrow H(e^{j\omega_k})$$

- ❑ Sample points are fixed $\omega_k = k \frac{\pi}{P}$

$$-\pi \leq \omega_1 < \cdots < \omega_p \leq \pi$$

- ❑ M+1 is the filter order
- ❑ P >> M + 1 (rule of thumb P=15M)
- ❑ Yields a (good) approximation of the original problem



Example: Least Squares

- ❑ Target: Design $M+1 = 2N+1$ filter
- ❑ First design non-causal $\tilde{H}(e^{j\omega})$ and hence $\tilde{h}[n]$



Example: Least Squares

- ❑ Target: Design $M+1 = 2N+1$ filter
- ❑ First design non-causal $\tilde{H}(e^{j\omega})$ and hence $\tilde{h}[n]$

- ❑ Then, shift to make causal

$$h[n] = \tilde{h}[n - M/2]$$

$$H(e^{j\omega}) = e^{-j\frac{M}{2}} \tilde{H}(e^{j\omega})$$



Example: Least Squares

$$\tilde{h} = [\tilde{h}[-N], \tilde{h}[-N+1], \dots, \tilde{h}[N]]^T$$

$$b = [H_d(e^{j\omega_1}), \dots, H_d(e^{j\omega_P})]^T$$

$$A = \begin{bmatrix} e^{-j\omega_1(-N)} & \dots & e^{-j\omega_1(+N)} \\ \vdots & & \vdots \\ e^{-j\omega_P(-N)} & \dots & e^{-j\omega_P(+N)} \end{bmatrix}$$

$$\operatorname{argmin}_{\tilde{h}} \|A\tilde{h} - b\|^2$$



Least-Squares

$$\operatorname{argmin}_{\tilde{h}} \quad ||A\tilde{h} - b||^2$$

Solution:

$$\tilde{h} = (A^* A)^{-1} A^* b$$

- Result will generally be non-symmetric and complex valued.
- However, if $\tilde{H}(e^{j\omega})$ is real, $\tilde{h}[n]$ should have symmetry!

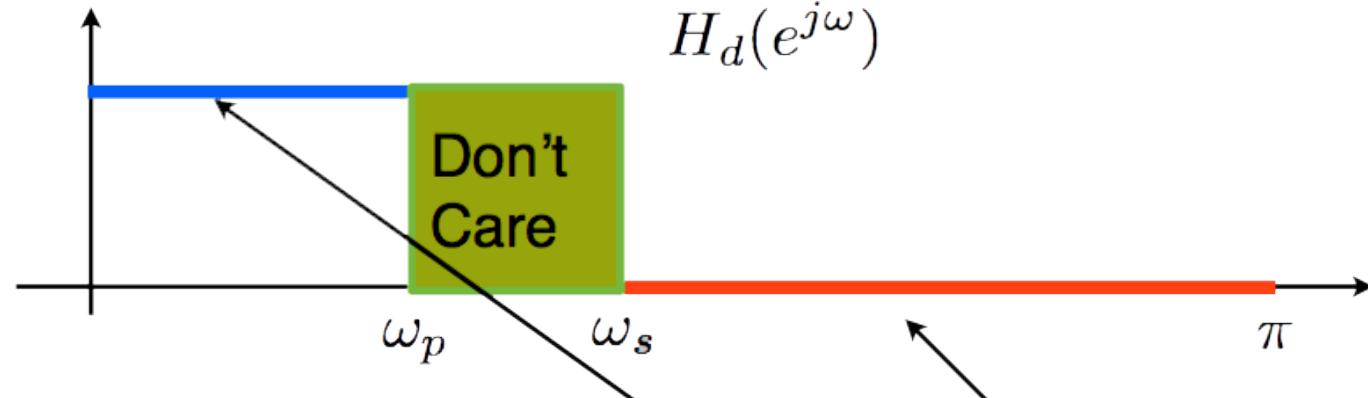


Design of Linear-Phase L.P Filter

- Suppose:
 - $\tilde{H}(e^{j\omega})$ is real-symmetric
 - M is even (M+1 length)
- Then:
 - $\tilde{h}[n]$ is real-symmetric around midpoint
- So:

$$\begin{aligned}\tilde{H}(e^{j\omega}) &= \tilde{h}[0] + \tilde{h}[1]e^{-j\omega} + \tilde{h}[-1]e^{+j\omega} \\ &\quad + \tilde{h}[2]e^{-j2\omega} + \tilde{h}[-2]e^{+j2\omega} \dots \\ &= \tilde{h}[0] + 2\cos(\omega)\tilde{h}[1] + 2\cos(2\omega)\tilde{h}[2] + \dots\end{aligned}$$

Least-Squares Linear Phase Filter

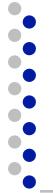


Given M , ω_P , ω_S find the best LS filter:

$$A = \begin{bmatrix} 1 & \dots & 2 \cos\left(\frac{M}{2}\omega_1\right) \\ \vdots & & \vdots \\ 1 & \dots & 2 \cos\left(\frac{M}{2}\omega_p\right) \\ 1 & \dots & 2 \cos\left(\frac{M}{2}\omega_s\right) \\ \vdots & & \vdots \\ 1 & \dots & 2 \cos\left(\frac{M}{2}\omega_P\right) \end{bmatrix}$$

$b = [1, 1, \dots, 1, 0, 0, \dots, 0]^T$

Capital P



Least-Squares Linear Phase Filter

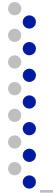
Given M , ω_P , ω_s find the best LS filter:

$$A = \begin{bmatrix} 1 & \dots & 2 \cos\left(\frac{M}{2}\omega_1\right) \\ \vdots & & \\ 1 & \dots & 2 \cos\left(\frac{M}{2}\omega_p\right) \\ 1 & \dots & 2 \cos\left(\frac{M}{2}\omega_s\right) \\ \vdots & & \\ 1 & \dots & 2 \cos\left(\frac{M}{2}\omega_P\right) \end{bmatrix} \quad b = [1, 1, \dots, 1, 0, 0, \dots, 0]^T$$

$$\tilde{h}_+ = [\tilde{h}[0], \dots, \tilde{h}\left[\frac{M}{2}\right]]^T = (A^* A)^{-1} A^* b$$

$$\tilde{h} = \begin{cases} \tilde{h}_+[n] & n \geq 0 \\ \tilde{h}_+[-n] & n < 0 \end{cases}$$

$$h[n] = \tilde{h}[n - M/2]$$



Extension:

- ❑ LS has no preference for pass band or stop band
- ❑ Use weighting of LS to change ratio

want to solve the discrete version of:

$$\text{minimize} \quad \int_{-\pi}^{\pi} W(\omega) |H(e^{j\omega}) - H_d(e^{j\omega})|^2 d\omega$$

where $W(\omega)$ is δ_p in the pass band and δ_s in stop band

Similarly: $W(\omega)$ is 1 in the pass band and δ_p/δ_s in stop band



Weighted Least-Squares

$$\operatorname{argmin}_{\tilde{h}_+} (A\tilde{h}_+ - b)^* W^2 (A\tilde{h}_+ - b)$$

Solution:

$$\tilde{h}_+ = (A^* W^2 A)^{-1} W^2 A^* b$$

$$W = \begin{bmatrix} 1 & & & & 0 \\ & 1 & & & \\ & & \dots & & \\ & & & \frac{\delta_p}{\delta_s} & \\ & & & & \dots \\ 0 & & & & & \frac{\delta_p}{\delta_s} \end{bmatrix}$$



Optimality – min-max

- Chebychev Design (min-max)

$$\text{minimize}_{\omega \in \text{care}} \quad \max |H(e^{j\omega}) - H_d(e^{j\omega})|$$

- Parks-McClellan algorithm - equiripple
- Also known as Remez exchange algorithms (`signal.remez`)
- Can also use convex optimization

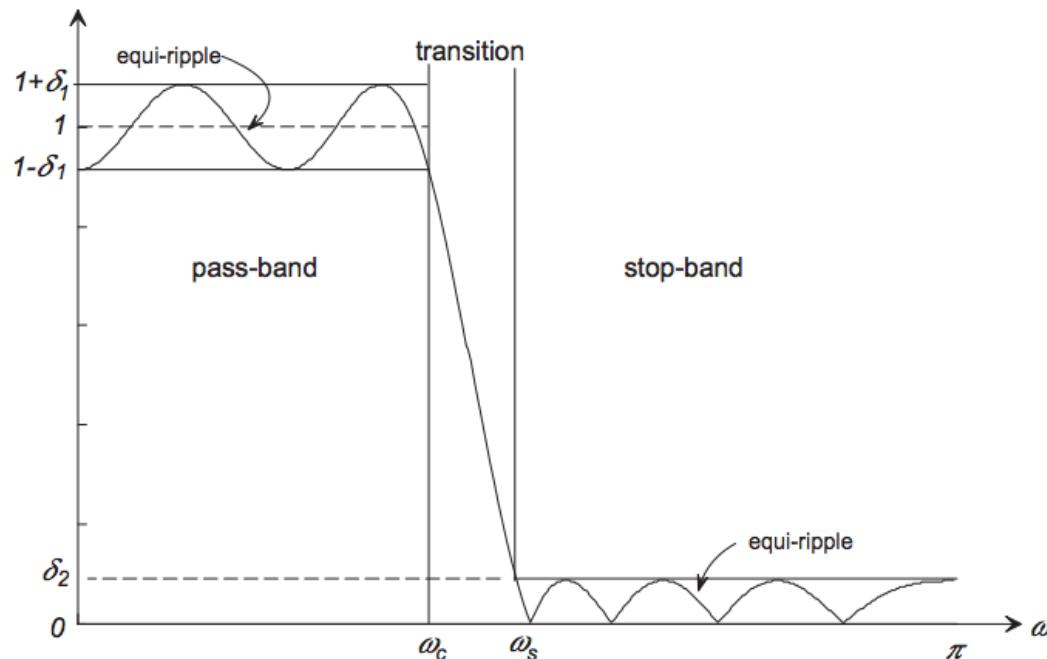


Parks-McClellan

- Allows for multiple pass- and stop-bands.
- Is an equi-ripple design in the pass- and stop-bands, but allows independent weighting of the ripple in each band.
- Allows specification of the band edges.

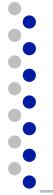


Parks-McClellan: LP Filter



- For the low-pass filter shown above the specifications are

$$\begin{aligned} 1 - \delta_1 &< H(e^{j\omega}) &< 1 + \delta_1 & \text{in the pass-band } 0 < \omega \leq \omega_c \\ -\delta_2 &< H(e^{j\omega}) &< \delta_2 & \text{in the stop-band } \omega_s < \omega \leq \pi. \end{aligned}$$



Parks-McClellan: LP Filter

- ❑ Need to determine $M+1$ (length of the filter) and the filter coefficients $\{h_n\}$



Parks-McClellan: LP Filter

- Need to determine $M+1$ (length of the filter) and the filter coefficients $\{h_n\}$
- If we assume M even and even symmetry FIR filter (Type I), then

$$A_e(e^{j\omega}) = h_e[0] + \sum_{n=1}^L 2h_e[n] \cos(\omega n).$$

$$H(e^{j\omega}) = A_e(e^{j\omega})e^{-j\omega M/2}.$$

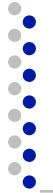


Parks-McClellan: LP Filter

- Define approximation error function

$$E(\omega) = W(\omega)[H_d(e^{j\omega}) - A_e(e^{j\omega})],$$

$$W(e^{j\omega}) = \begin{cases} \delta_2/\delta_1 & \text{in the pass-band} \\ 1 & \text{in the stop-band} \\ 0 & \text{in the transition band.} \end{cases}$$



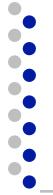
Parks-McClellan: LP Filter

- Define approximation error function

$$E(\omega) = W(\omega)[H_d(e^{j\omega}) - A_e(e^{j\omega})],$$

- Apply min-max or Chebyshev criteria

$$\min_{\{h_e[n]: 0 \leq n \leq L\}} \left(\max_{\omega \in F} |E(\omega)| \right),$$



Min-Max Filter Design

- Constraints:

- min-max pass-band ripple

$$1 - \delta_p \leq |H(e^{j\omega})| \leq 1 + \delta_p, \quad 0 \leq w \leq \omega_p$$

- min-max stop-band ripple

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

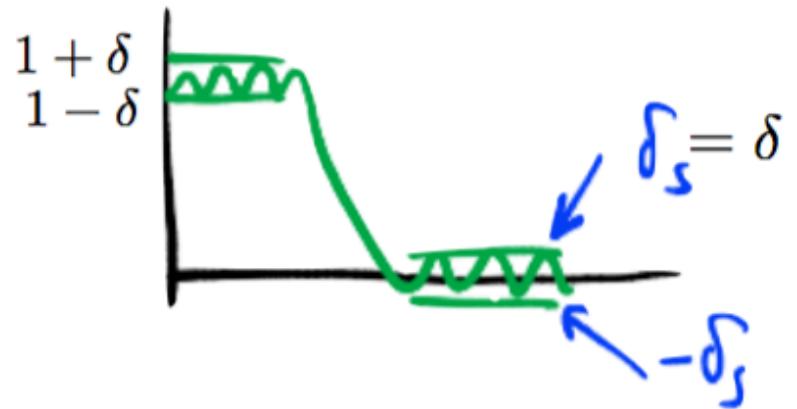
Min-Max Ripple Design

- Given ω_p, ω_s, M , find δ, \tilde{h}_+

minimize δ

Subject to :

$$\begin{aligned}1 - \delta &\leq \tilde{H}(e^{j\omega_k}) \leq 1 + \delta & 0 \leq \omega_k \leq \omega_p \\-\delta &\leq \tilde{H}(e^{j\omega_k}) \leq \delta & \omega_s \leq \omega_k \leq \pi \\\delta &> 0\end{aligned}$$



- Formulation is a linear program with solution δ, \tilde{h}_+
- A well studied class of problems with good solvers

Min-Max Ripple via LP

minimize δ

subject to :

$$1 - \delta \leq A_p \tilde{h}_+ \leq 1 + \delta$$

$$-\delta \leq A_s \tilde{h}_+ \leq \delta$$

$$\delta > 0$$

$$A_p = \begin{bmatrix} 1 & 2 \cos(\omega_1) & \cdots & 2 \cos\left(\frac{M}{2}\omega_1\right) \\ & \vdots & & \\ 1 & 2 \cos(\omega_p) & \cdots & 2 \cos\left(\frac{M}{2}\omega_p\right) \end{bmatrix}$$
$$A_s = \begin{bmatrix} 1 & 2 \cos(\omega_s) & \cdots & 2 \cos\left(\frac{M}{2}\omega_s\right) \\ & \vdots & & \\ 1 & 2 \cos(\omega_{\textcolor{red}{P}}) & \cdots & 2 \cos\left(\frac{M}{2}\omega_{\textcolor{red}{P}}\right) \end{bmatrix}$$

capital P

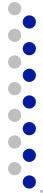


Parks-McClellan

- ❑ The method is based on reformulating the problem as one in polynomial approximation, using Chebyshev polynomials

$$A_e(e^{j\omega}) = \sum_{k=0}^L a_k (\cos \omega)^k,$$

$$A_e(e^{j\omega}) = P(x)|_{x=\cos \omega}, \quad P(x) = \sum_{k=0}^L a_k x^k.$$



Parks-McClellan – Alternation Theorem

- The algorithm uses Chebyshev's alternation theorem to recognize the optimal solution.

Define the error $E(x)$ as above, namely

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

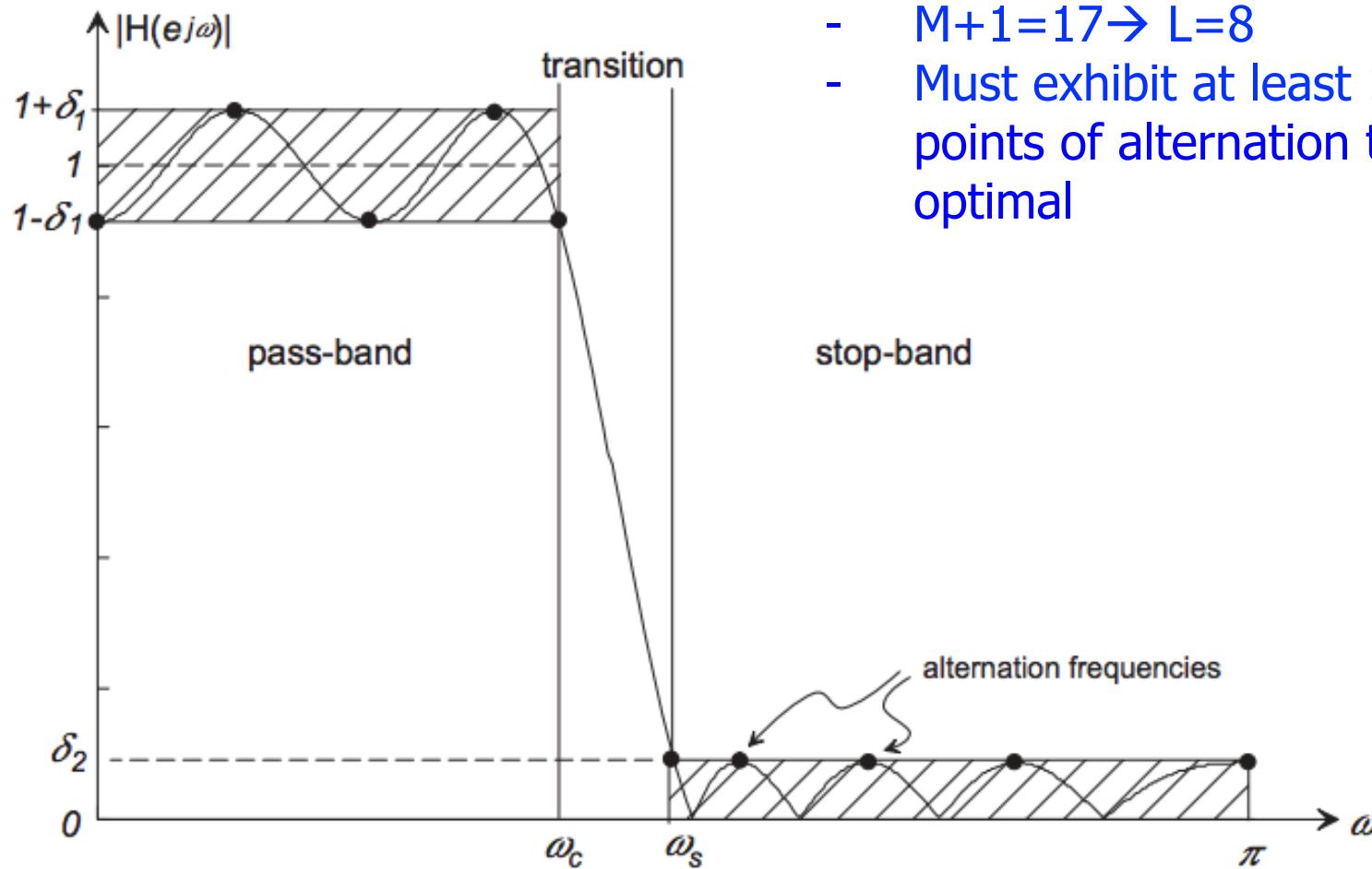
and the maximum error as

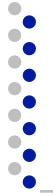
$$\|E(e^{j\omega})\|_\infty = \operatorname{argmax}_{x \in \Omega} |E(e^{j\omega})|$$

A necessary and sufficient condition that $H(e^{j\omega})$ is the unique L th-order polynomial minimizing $\|E(e^{j\omega})\|_\infty$ is that $E(e^{j\omega})$ exhibit at least $L + 2$ extremal frequencies, or “alternations”, that is there must exist at least $L+2$ values of ω , $\omega_k \in \Omega$, $k = [0, 1, \dots, L + 1]$, such that $\omega_0 < \omega_1 < \dots < \omega_{L+1}$, and such that

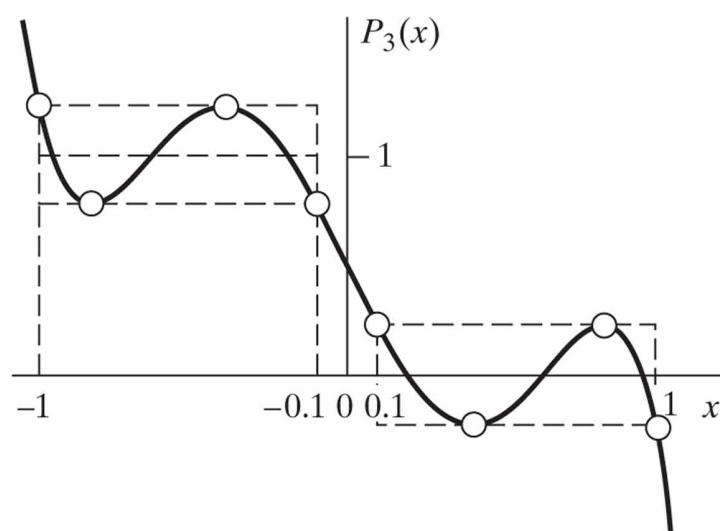
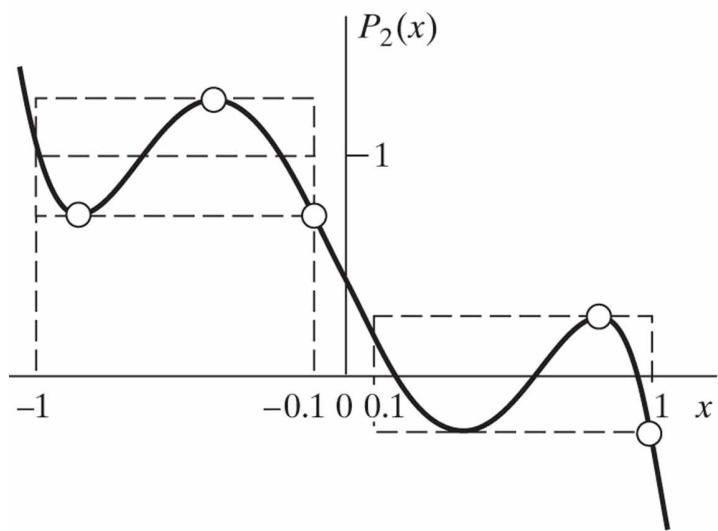
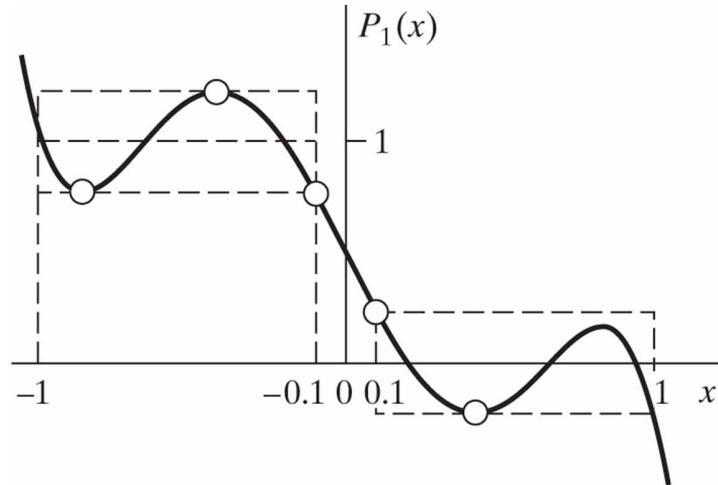
$$E(e^{j\omega_k}) = -E(e^{j\omega_{k+1}}) = \pm (\|E(e^{j\omega})\|_\infty).$$

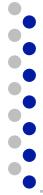
Parks-McClellan – Alternation Theorem



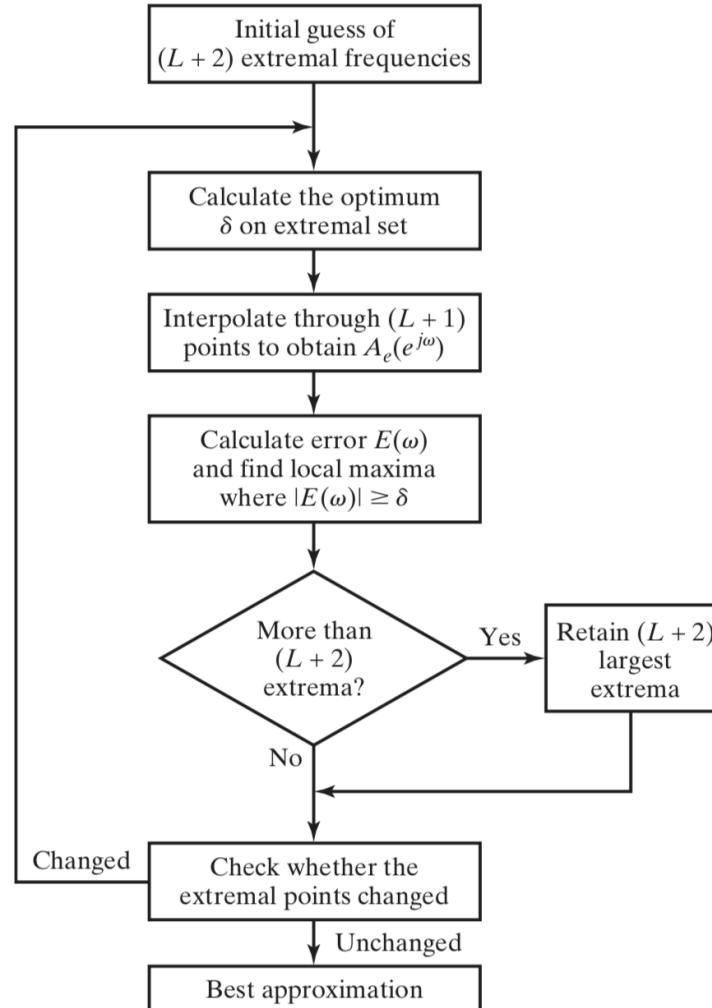


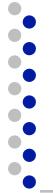
Alternation Theorem Example – 5th order





Parks–McClellan algorithm

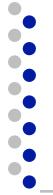




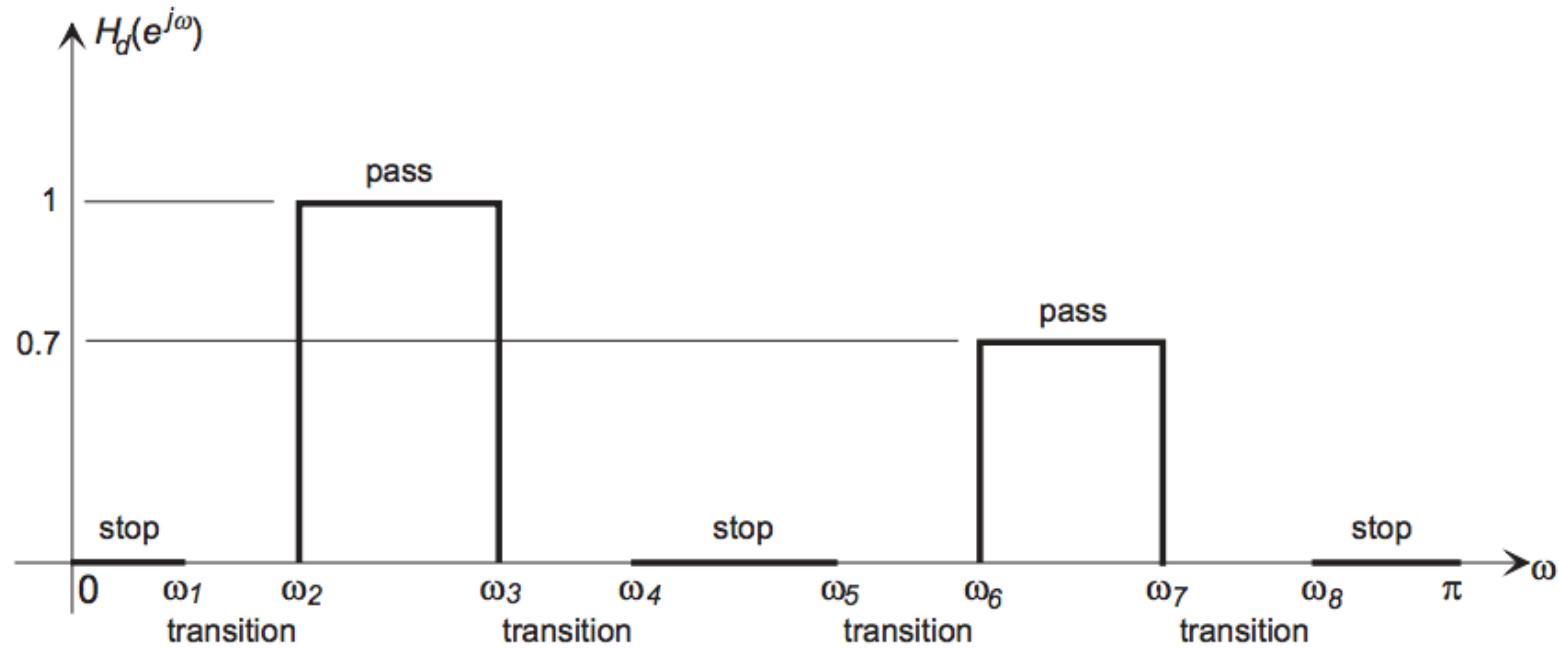
MATLAB Parks-McClellan Function

❑ **b = firpm(M, F, A, W)**

- b is the array of filter coefficients (impulse response)
- M is the filter order (M+1 is the length of the filter),
- F is a vector of band edge frequencies in ascending order
- A is a set of filter gains at the band edges
- W is an optional set of relative weights to be applied to each of the bands



MATLAB Parks-McClellan Function



$$F = [0 \ \omega_1 \ \omega_2 \ \omega_3 \ \omega_4 \ \omega_5 \ \omega_6 \ \omega_7 \ \omega_8 \ 1]$$

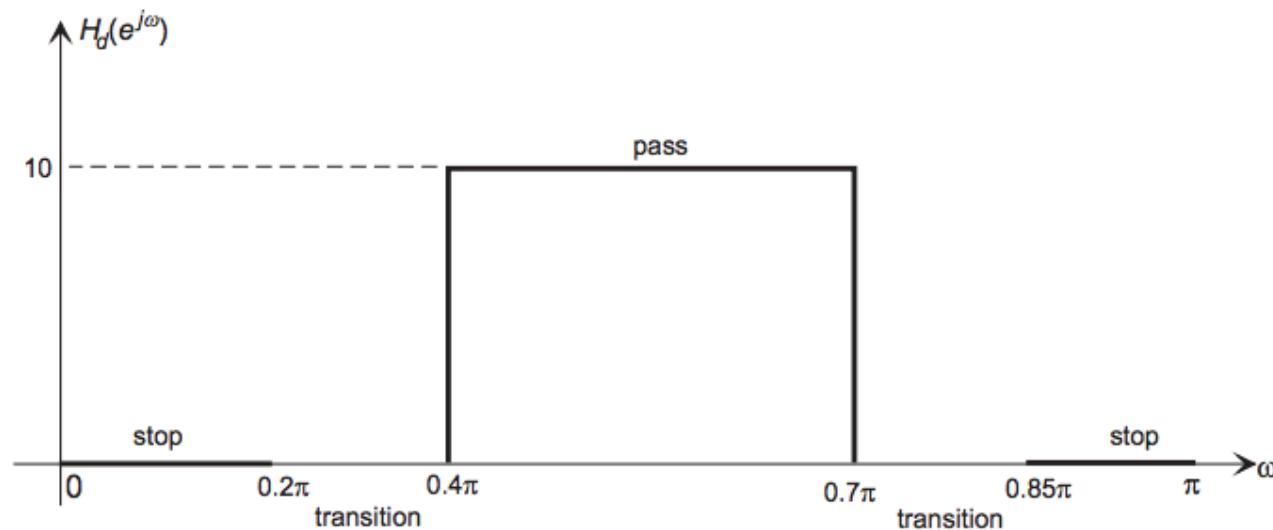
$$A = [0 \ 0 \ 1 \ 1 \ 0 \ 0 \ 0.7 \ 0.7 \ 0 \ 0]$$

$$W = [10 \ 1 \ 10 \ 1 \ 10]$$



MATLAB Example

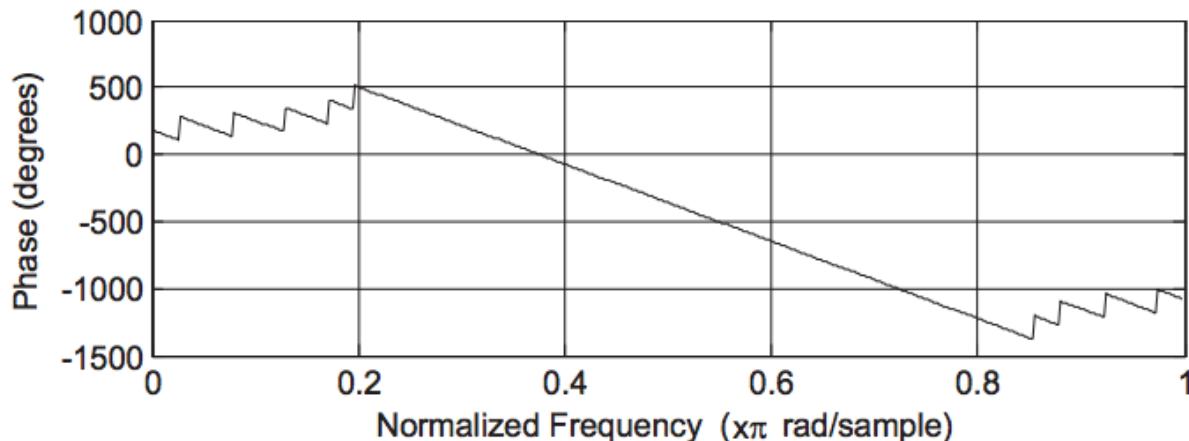
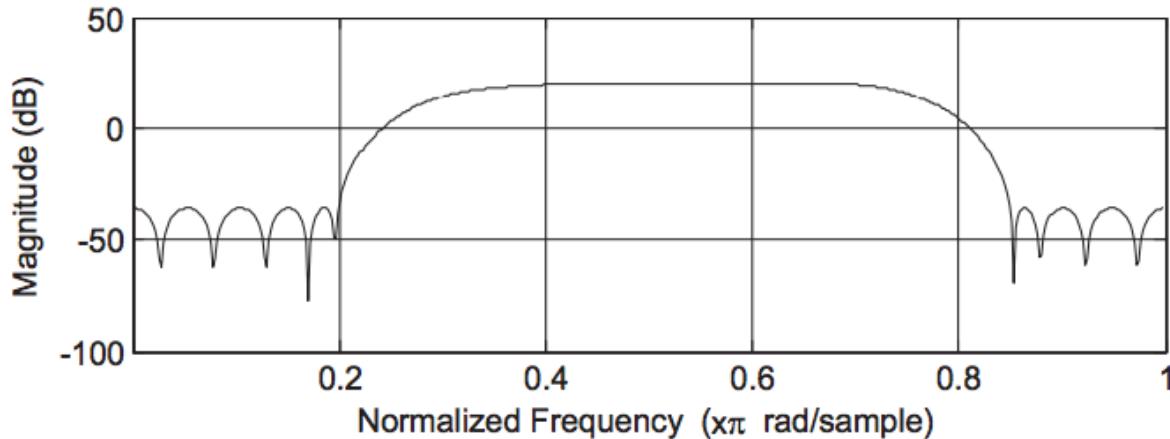
- ❑ Design a 33 length PM band-pass filter and weight the stop-band ripple 10x more than the pass-band ripple



```
h=firpm(32,[0 0.2 0.4 0.7 0.85 1],[0 0 10 10 0 0],[10 1 10])  
freqz(h,1)
```

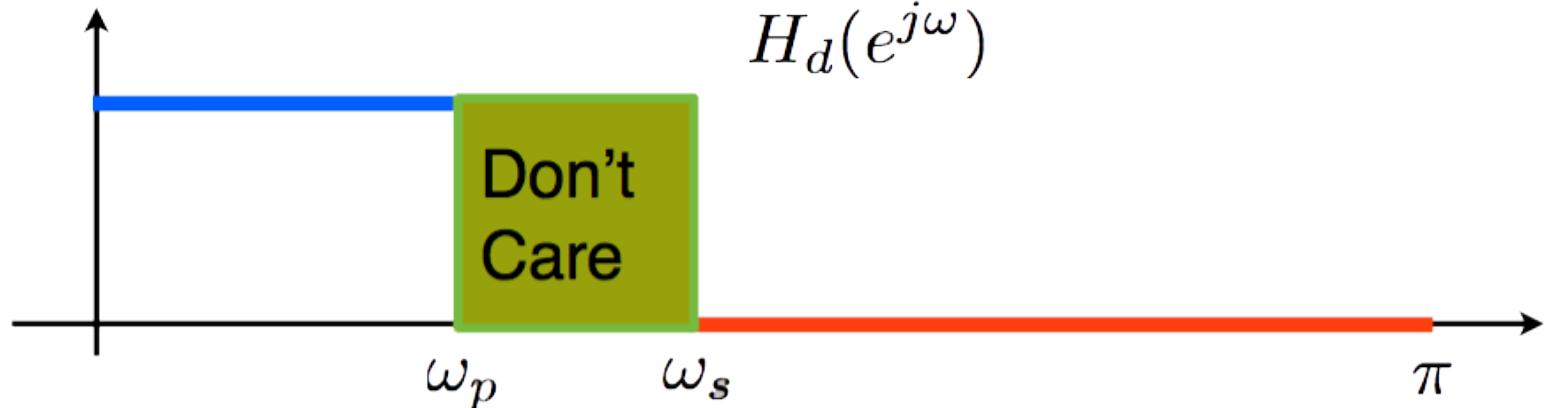
MATLAB Example

- Design a bandpass filter



```
h=firpm(10,[0.1 0.9], [1 10])  
freqz(h)
```

Optimality – Least Squares



- Least Squares:

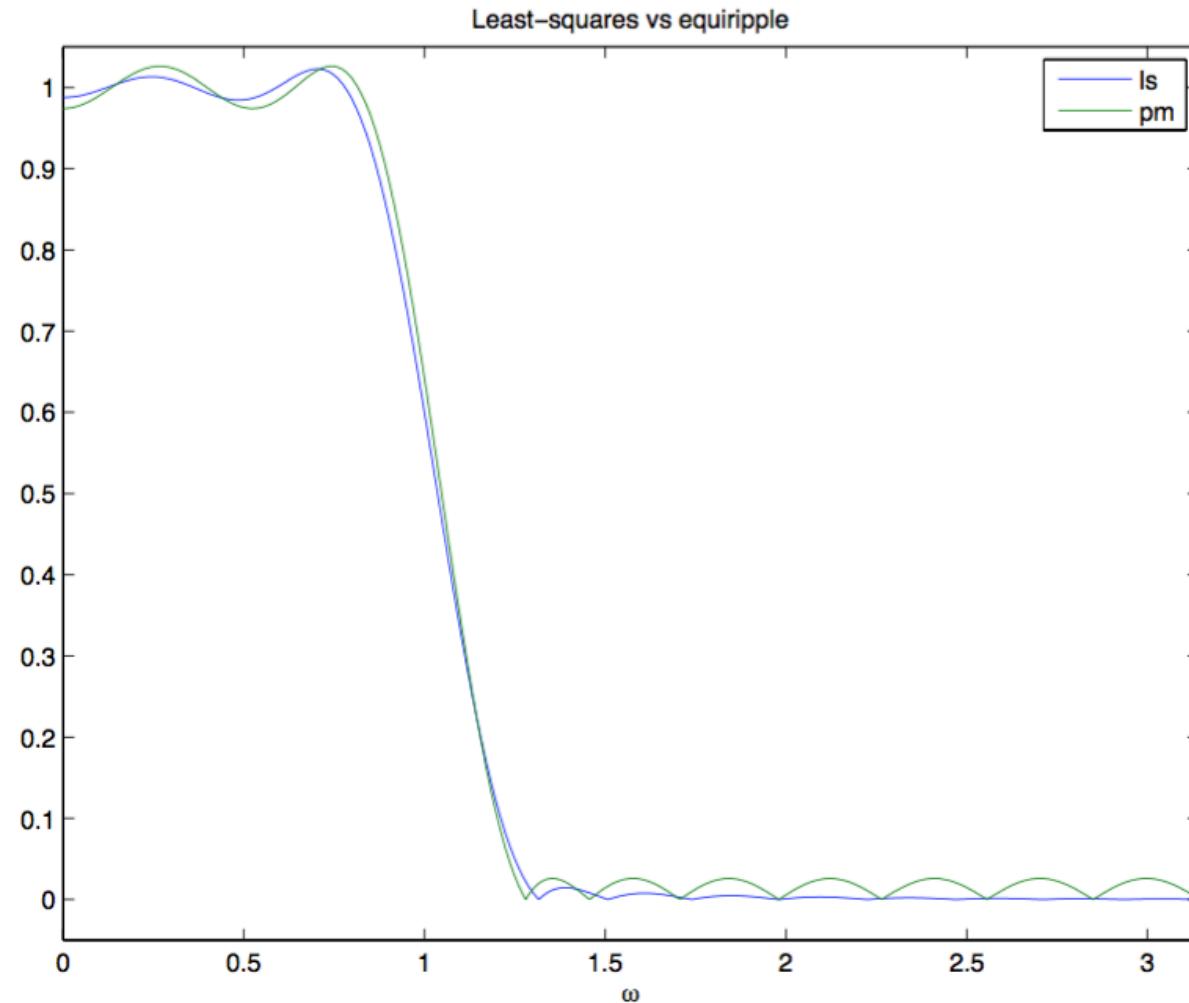
$$\text{minimize} \quad \int_{\omega \in \text{care}} |H(e^{j\omega}) - H_d(e^{j\omega})|^2 d\omega$$

- Parks-McClellan

$$\min_{\{h_e[n]: 0 \leq n \leq L\}} \left(\max_{\omega \in F} |E(\omega)| \right),$$

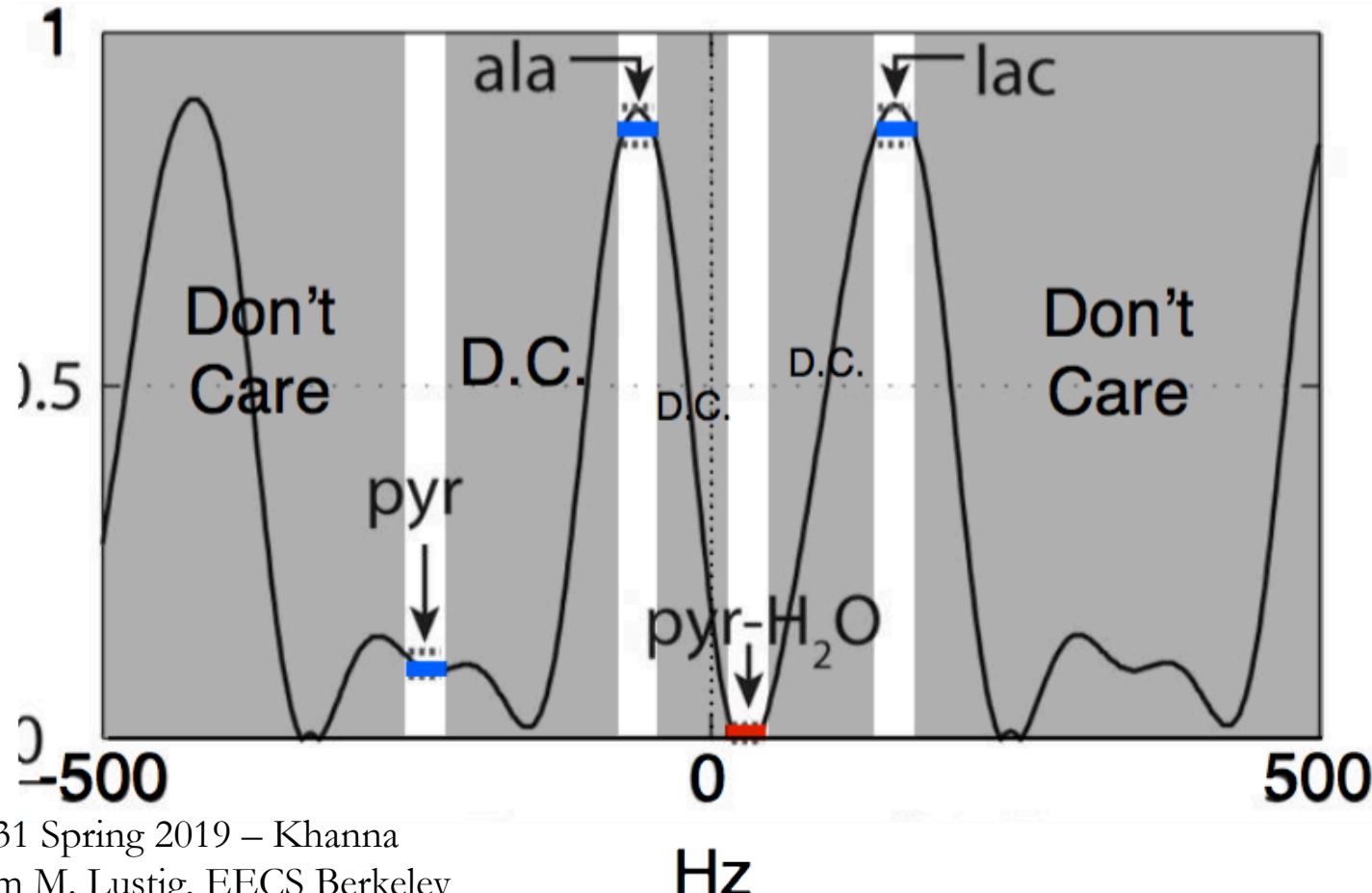


Least-Squares vs. Min-Max



Example of Complex Filter

- ❑ Larson et. al, “Multiband Excitation Pulses for Hyperpolarized ^{13}C Dynamic Chemical Shift Imaging” JMR 2008;194(1):121-127
- ❑ Need to design length 11 filter with following frequency response:





Convex Optimization

- Many tools and Solvers
- Tools:
 - CVX (Matlab) <http://cvxr.com/cvx/>
 - CVXOPT, CVXMOD (Python)
- Engines:
 - Sedumi (Free)
 - MOSEK (commercial)



Using CVX (in Matlab)

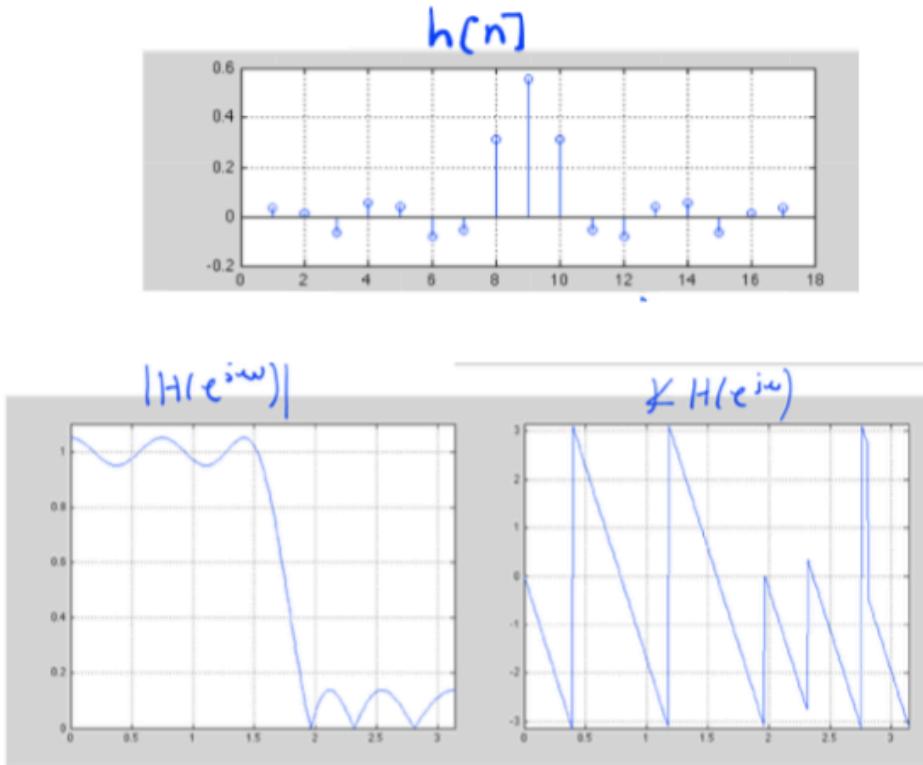
```
M = 16;
wp = 0.5*pi;
ws = 0.6*pi;
MM = 15*M;
w = linspace(0,pi,MM);

idxp = find(w <=wp);
idxs = find(w >=ws);

Ap = [ones(length(idxp),1) 2*cos(kron(w(idxp)',[1:M/2]))];
As = [ones(length(idxs),1) 2*cos(kron(w(idxs)',[1:M/2]))];

% optimization
cvx_begin
    variable hh(M/2+1,1);
    variable d(1,1);

    minimize(d)
    subject to
        Ap*hh <=1+d;
        Ap*hh >=1-d;
        As*hh < d;
        As*hh > -d;
        d>0;
cvx_end
h = [hh(end:-1:1) ; hh(2:end)];
```





Admin

- ❑ HW 6
 - Due Sunday 3/29