Digital Signal Processing

FIR Filter Design

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FIR Design

- No connection with the analog filter design
- Based on the magnitude response
- Linear Phase
- Described by transfer function polynomial in z⁻¹
- Guaranteed Stability
- Design Method
 - Windowed FS
 - Frequency sampling

FIR Design

$$H_d(e^{\,j\omega}) = \sum_{n=-\infty}^\infty h_d[n] e^{-jwn} \qquad \qquad h_d[n] = \!\int_{-\pi}^\pi \! H_d(e^{\,j\omega}) e^{\,j\omega n} d\omega$$

Objective

Find a finite-duration $\{h_t[n]\}$ of length 2M+1 whose DTFT $H_t(e^{j\omega})$ approximates the desired DTFT $H_d(e^{j\omega})$ in some sense

$$egin{align} \Phi &= rac{1}{2\pi} \! \int_{-\pi}^{\pi} \! |H_d(e^{j\omega}) - H_t(e^{j\omega})|^2 d\omega \ &= \sum_{n=-\infty}^{\infty} |h_d[n] - h_t[n]|^2 \ &= \sum_{n=-M}^{M} |h_d[n] - h_t[n]|^2 + \Phi_M \ \end{gathered}$$

 Φ is minimized when $h_t[n] = h_d[n]$ for $-M \le n \le M$

$$h_t[n] = h_d[n]W_R[n]$$

Truncating operation \rightarrow windowing operation:

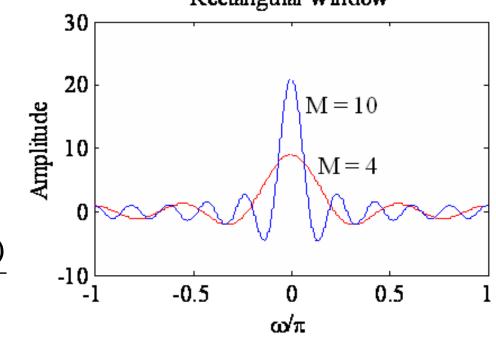
$$h_t[n] = h_d[n]W_R[n]$$

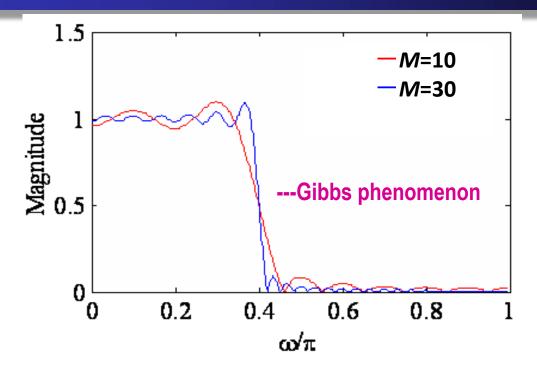
$$H_t(e^{j\omega}) = rac{1}{2\pi}\!\int_{-\pi}^{\pi}\! H_d(e^{j au}) \varPsi_Rig(e^{j(\omega- au)}ig) d au$$

Rectangular window

$$w_R[n] = \begin{cases} 1, & 0 \le |n| \le M \\ 0, & \text{otherwise} \end{cases}$$

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





- Observation: ↑M (length of the lowpass filter) → ↑ number of ripples (both passband and stopband) with corresponding ripple widths
- Height of the largest ripples remain the same independent of length

- Gibbs phenomenon—Oscillatory of the magnitude due to truncation
- Rectangular window—Zero outside -M≤n≤M
- To Reduce Gibbs phenomenon
 - Using a window that tapers smoothly to zero
 - smooth transition from passband to stopband in the magnitude specifications

- Windowing: causes the height of the sidelobes to diminish, with a increased main lobe width
- Hann:

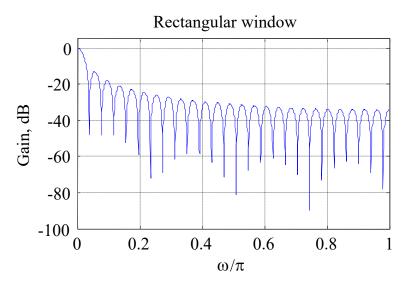
$$w[n] = 0.5 + 0.5\cos[2\pi n/(2M+1)], -M \le n \le M$$

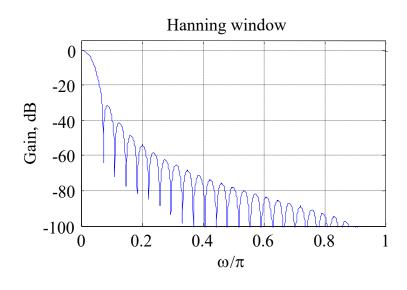
Hamming:

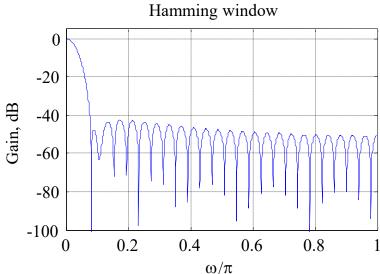
$$w[n] = 0.54 + 0.46\cos[2\pi n/(2M+1)], -M \le n \le M$$

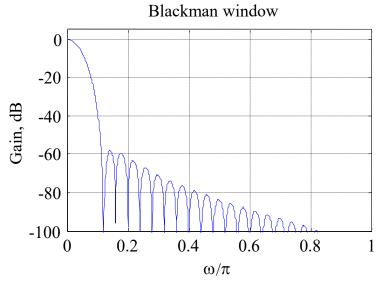
Blackman:

$$w[n] = 0.42 + 0.5\cos[2\pi n/(2M+1)] + 0.08\cos[4\pi n/(2M+1)]$$









Type of windows	Main lobe width Δ _{ML}	Relative sidelobe level A _{sl}	Minimum stopband attenuation	Transition bandwidth Δω
Rectangular	4π/(2M+1)	13.3dB	20.9dB	0.92π/Μ
Hann	8π/(2M+1)	31.5dB	43.9dB	3.11π/M
Hamming	8π/(2M+1)	42.7dB	54.5dB	3.32π/M
Blackman	12π/(2 <i>M</i> +1)	75.3dB	75.3dB	5.56π/M

Adjustable window Function: Kaiser window.

$$w[n] = \frac{I_0(\beta \sqrt{1 - \left[\frac{n}{M}\right]^2})}{I_0(\beta)}, -M \le n \le M$$

$$N \cong \frac{-20 \log_{10}(\sqrt{\delta_P \delta_S}) - 13}{14.6(\omega_S - \omega_P)/(2\pi)}$$

- (i) All above formulas are used in case $\omega_S > \omega_{P}$.
- (ii) Interchange δ_S , δ_P , the order of FIR are same.
- (iii) Ripples δ_S , δ_P are decreased, the order of FIR is increased.
- (iv) Transition bandwidth $\omega_s \omega_P$ is decreased, the order of FIR is increased.

Design Steps for Windowed Low Pass FIR Filters

(1) Choose a pass band edge frequency in Hz:

$$f_c = (f_p + f_s)/2$$

(2) Calculate $\omega_c = 2\pi f_c/f_T$, the infinite impulse response for an ideal low pass filter:

$$h_d[n] = \sin(n\omega_c)/n\pi$$

- (3) Choose a window based on the specifications
- (4) Calculate FIR from $h_t[n] = h_d[n]w[n], \text{ notice it is noncausal.}$
- (5) Shift $h_t[n]$ to the right by M to make the filter causal $h[n] = h_t[n M]$

FIR Filter Design Example 1st

The specifications of a low pass filter:

Pass band edge f_p 2kHz

Stop band edge f_s 3kHz

Stop band attenuation 40dB

Sampling frequency f_{τ} 10kHz

Design

(1) Transition width
$$\Delta f = 3 - 2 = 1 \text{ kHz}$$

$$f_c = (2000+3000)/2 = 2500 \text{Hz}$$

$$\omega_c = 2\pi f_1/f_T = 2\pi 2500/10000 = 0.5\pi$$

- (2) $h_d[n] = \sin(n\omega_c)/n\pi = \sin(0.5\pi n)/n\pi$
- (3) Choose Hanning window (Table 10.2, 40dB required)

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Transition Width \Delta \omega = 2\pi \Delta f / f_T = 0.2\pi

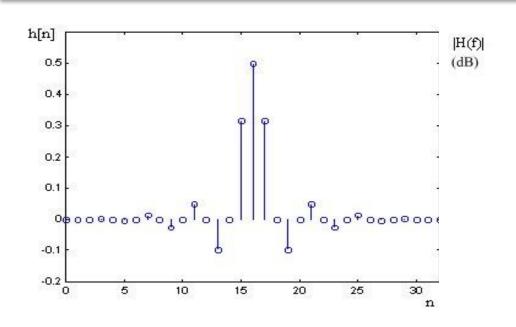
M = 3.11\pi/\Delta \omega = 15.55, the least M = 16, choose order N = 2M = 32, length N+1 = 33

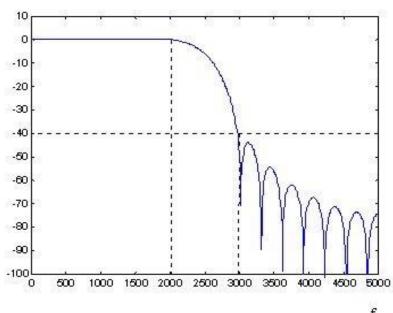
w[n] = 0.5 + 0.5\cos(2\pi n/33)

(4) h_t[n] = h_d[n]w[n] = [\sin(0.5\pi n)/n] w[n],

(5) h[n] = h_t[n-M] = h_t[n-16]
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Type I linear-phase FIR LPF





frequency Response

Impulse Response

- $G(j\omega_p) = -0.06$ dB at $\omega_p = 2.013$ kHz
- $G(j\omega_s) = -40$ dB at $\omega_s = 2.976$ kHz

From Frequency Specification samples to h[n]

$$H(k) = H\left(e^{jrac{2\pi}{N+1}k}
ight)$$

The impulse response h[n] can be derived from IDFT of H(k).

H(k) must be linear phase FIR.

The Design Steps:

- (1)From ideal filter, the values of H(k) are computed (linear phase).
- (2) From the IDFT of H(k), h[n] is derived.

Design a linear phase FIR BPF, with the passband edge frequencies at 500Hz and 700Hz.

The sampling frequency is $f_s = 3.3$ KHz, and the order N=32(length 33).

Design

(1) Determination of H(k). $N = 32, f_s = 3.3$ KHz,

$$\Delta f = \frac{fs}{N+1} = 0.1 \text{KHz}$$

we choose Type 1 FIR.

$$H(\omega) = H(2\pi - \omega)$$

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we choose Type 1 FIR.

$$H(\omega) = H(2\pi - \omega)$$

$$heta(\omega) = -M\omega = -16\omega$$
 $\omega_k = 2\pi/33$

$$H(k) = Higg(e^{jrac{2\pi}{N+1}k}igg) = Higg(e^{jrac{2\pi\Delta f}{f_s}k}igg)$$

$$H(k) = \begin{cases} 1 \cdot e^{j\frac{32\pi}{33}k} & k = 5, 67, 26, 27, 28 \\ 0 & \text{otherwise} \end{cases}$$

$$h[n] = \frac{1}{N+1} \sum_{k=0}^{N} H(k) W_{N+1}^{-kn} \stackrel{0}{\underset{0}{|}} 1 2 3 4 5 6 7 8 16 26 27 28 32 k$$

$$< N+1-k >_{N+1} = < 33-k >_{33}$$