



درس ۲۰

طراحی فیلترهای گسسته-زمان با ينجرهزني

**Discrete-Time Filter Design by Windowing** 

http://courses.fouladi.ir/dsp

## **Discrete-Time Filter Design by Windowing**

Digital Signal Processing

## Filter Design by Windowing

- Simplest way of designing FIR filters
- Method is all discrete-time no continuous-time involved
- Start with ideal frequency response

$$H_{d}\left(e^{j\omega}\right) = \sum_{n=-\infty}^{\infty} h_{d}\left[n\right]e^{-j\omega n} \qquad h_{d}\left[n\right] = \frac{1}{2\pi} \int_{-\pi}^{\pi} H_{d}\left(e^{j\omega}\right)e^{j\omega n}d\omega$$

- Choose ideal frequency response as desired response
  - Most ideal impulse responses are of infinite length
- The easiest way to obtain a causal FIR filter from ideal is

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

• More generally

$$h[n] = h_d[n]w[n]$$
 where  $w[n] = \begin{cases} 1 & 0 \le n \le M \\ 0 & \text{otherwise} \end{cases}$ 

## Windowing in Frequency Domain

• Windowed frequency response (Periodic Convolution)

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

• The windowed version is smeared version of desired response





• If w[n] = 1 for all *n*, then  $W(e^{j\omega})$  is impulse train with  $2\pi$  period (ideal case)

## **Properties of Windows**

- Prefer windows that concentrate around DC in frequency
  - (More similar to impulse  $\Rightarrow$ ) Less smearing, closer approximation
- Prefer window that has minimal span in time
  - Less coefficient in designed filter, computationally efficient

- So we want concentration in time and in frequency
  - Contradictory requirements!

### **Example: Rectangular window**

• Example: Rectangular window

### **Commonly Used Windows**



## **Rectangular Window**

#### • Narrowest main lob

- $4\pi/(M+1)$
- Sharpest transitions at discontinuities in frequency response H<sub>d</sub>(e<sup>jw</sup>)
- Large side lobs
  - -13 dB
  - Large oscillation around discontinuities
- Simplest possible window

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



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## **Bartlett (Triangular) Window**

• Medium main lob

 $- 8\pi/M$ 

- Side lobs
  - − −25 dB

• Simple equation:

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



# Hanning Window (Hann)

• Medium main lob

 $- 8\pi/M$ 

- Side lobs - - -31 dB
- Hamming window performs better
- Same complexity as Hamming

$$w[n] = \begin{cases} \frac{1}{2} \left[ 1 - \cos\left(\frac{2\pi n}{M}\right) \right] & 0 \le n \le M \\ 0 & \text{otherwise} \end{cases}$$



## **Hamming Window**

• Medium main lob

 $- 8\pi/M$ 

• Good side lobs

- −41 dB

• Simpler than Blackman

 $w[n] = \begin{cases} 0.54 - 0.46 \cos\left(\frac{2\pi n}{M}\right) \\ 0 \end{cases}$ 



## **Blackman Window**

Rectangular w[n]Large main lob 1.0 $12\pi/M$ Blackman 0.8 0.6 Very good side lobs 0.4 -57 dB0.2 0  $\frac{M}{2}$ М Complex equation  $w[n] = \begin{cases} 0.42 - 0.5 \cos\left(\frac{2\pi n}{M}\right) + 0.08 \cos\left(\frac{4\pi n}{M}\right) \\ 0 \end{cases}$  $0 \le n \le M$ otherwise M = 50-20 $20 \log_{10} |W(e^{j\omega})|$ -40-60 -80-100  $0.2\pi$  $0.4\pi$  $0.8\pi$  $0.6\pi$  $\pi$ Radian frequency  $(\omega)$ 

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#### **Incorporation of Generalized Linear Phase**

- Windows are designed with linear phase in mind
  - Symmetric around M/2

$$w[n] = \begin{cases} w[M-n] & 0 \le n \le M \\ 0 & \text{otherwise} \end{cases}$$

• So their Fourier transform are of the form

 $W(e^{j\omega}) = W_e(e^{j\omega})e^{-j\omega M/2}$  where  $W_e(e^{j\omega})$  is a real and even

- Will keep symmetry properties of the desired impulse response
- Assume <u>symmetric</u> desired response:

$$H_d(e^{j\omega}) = H_e(e^{j\omega})e^{-j\omega M/2}$$

• With <u>symmetric</u> window

$$H(e^{j\omega}) = A_e(e^{j\omega})e^{-j\omega M/2} \qquad A_e(e^{j\omega}) = \frac{1}{2\pi} \int_{-\pi}^{\pi} H_e(e^{j\theta}) W_e(e^{j(\omega-\theta)}) d\theta$$

- Periodic convolution of real functions

## **Linear-Phase Lowpass filter**

• Desired frequency response (with generalized linear phase):

$$H_{lp}\left(e^{j\omega}\right) = \begin{cases} e^{-j\omega M/2} & \left|\omega\right| < \omega_{c} \\ 0 & \omega_{c} < \left|\omega\right| \le \pi \end{cases}$$

• Corresponding impulse response (is also symmetric):

$$h_{lp}[n] = \frac{\sin[\omega_c(n-M/2)]}{\pi(n-M/2)}$$

• Desired response is **even symmetric**, use symmetric window

$$h[n] = \frac{\sin[\omega_c(n-M/2)]}{\pi(n-M/2)} w[n]$$



## Kaiser Window Filter Design Method

- Parameterized equation forming a set of windows
  - Has parameter to change main-lob width and side-lob area trade-off

$$w[n] = \begin{cases} I_0 \left( \beta \sqrt{1 - \left(\frac{n - M/2}{M/2}\right)^2} \right) & 0 \le n \le 1 \\ \hline I_0(\beta) & 0 & \text{otherwith} \end{cases}$$

- I<sub>0</sub>(.) represents zeroth-order modified
Bessel function of 1<sup>st</sup> kind



### **Determining Kaiser Window Parameters**

- Given filter specifications Kaiser developed empirical equations
  - Given the peak approximation error  $\delta$  or in dB as  $A = -20\log_{10}\delta$
  - and transition band width  $\Delta \omega = \omega_s \omega_p$
- The shape parameter  $\beta$  should be

$$\beta = \begin{cases} 0.1102(A-8.7) & A > 50\\ 0.5842(A-21)^{0.4} + 0.07886(A-21) & 21 \le A \le 50\\ 0 & A < 21 \end{cases}$$

• The filter order *M* is determined approximately by

$$M = \frac{A - 8}{2.285 \Delta \omega}$$

### **Example: Kaiser Window Design of a Lowpass Filter**

- Specifications  $\omega_p = 0.4\pi, \omega_p = 0.6\pi, \delta_1 = 0.01, \delta_2 = 0.001$
- Window design methods assume  $\delta_1 = \delta_2 = 0.001$
- Determine cut-off frequency

- Due to the symmetry we can choose it to be  $\omega_c = 0.5\pi$ 

• Compute

$$\Delta \omega = \omega_s - \omega_p = 0.2\pi \qquad A = -20 \log_{10} \delta = 60$$

• And Kaiser window parameters

$$\beta = 5.653$$
  $M = 37$  Odd (Type II FIR with Lin. Phase)

• Then the impulse response is given as

$$h[n] = \begin{cases} \frac{\sin[0.5\pi(n-18.5)]}{\pi(n-18.5)} \frac{I_0 \left[ 5.653\sqrt{1 - \left(\frac{n-18.5}{18.5}\right)^2} \right]}{I_0 (5.653)} & 0 \le n \le M\\ 0 & \text{otherwise} \end{cases}$$



### **Example Cont'd**



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#### **General Frequency Selective Filters**

• A general <u>multiband impulse response</u> can be written as

$$h_{mb}[n] = \sum_{k=1}^{N_{mb}} (G_k - G_{k+1}) \frac{\sin \omega_k (n - M/2)}{\pi (n - M/2)}$$

$$G_{N_{mb}+1}=0$$

- Window methods can be applied to multiband filters
- Example multiband frequency response
  - Special cases of
    - Bandpass
    - Highpass
    - Bandstop



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