

## Today's lecture

- Linear phase filters
- Different filter design techniques

1. Frequency sampling
2. Least-square approximation
3. Chebyshev approximation

## Linear Phase Filters

$h[n]$ : length  $N$  FIR

has linear phase  $\theta(\omega) = k_1 + k_2 \omega$

$$H(\omega) = e^{-j\omega M} \sum_{n=0}^{N-1} h[n] e^{j\omega(M-n)} \quad \text{where } M = \frac{N-1}{2}$$

Rewriting this we can show that

$$H(\omega) = e^{-j\omega M} \left( (h[0] + h[N-1]) \cos \omega M + j(h[0] - h[N-1]) \sin \omega M \right. \\
+ (h[1] + h[N-2]) \cos \omega(M-1) + j(h[1] - h[N-2]) \sin \omega(M-1) \\
+ \vdots \left. \right)$$

Now assume that we have even symmetry, i.e.,  $h[N-n-1] = h[2M-n] = h[n]$

Then all the sine terms will drop away:

$$H(\omega) = e^{-j\omega M} \left( (h[0] + h[N-1]) \cos \omega M + (h[1] + h[N-2]) \cos \omega(M-1) + \dots \right)$$

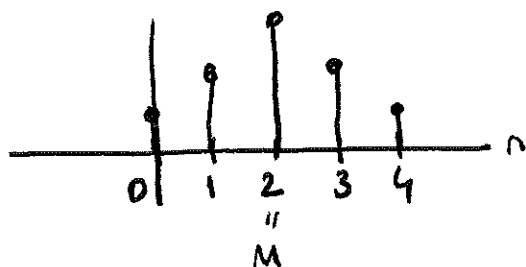
Amplitude response:

$$A(\omega) = \sum_{n=0}^{M-1} 2h[n] \cos(\omega(M-n)) + h[M] \quad \text{if } N \text{ is ODD}$$

$$(n \rightarrow M-n) = \sum_{n=1}^M 2h[M-n] \cos(\omega n) + h[M] \quad \text{'Type I' } N \text{ ODD'}$$

Example

$$N=5, \quad h[n] = h[4-n], \quad M = \frac{N-1}{2} = 2$$



Other Types:

$$A(\omega) = \sum_{n=0}^{N/2-1} 2 h[n] \cos(\omega(M-n))$$

$$(n \rightarrow \frac{N}{2} - n) = \sum_{n=1}^{N/2} 2 h[\frac{N}{2} - n] \cos(\omega(n - \frac{1}{2}))$$

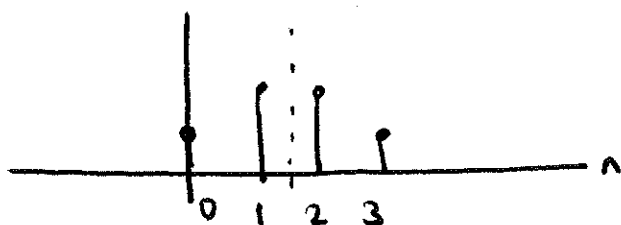
'Type II N EVEN'

change of variables  $n \rightarrow N/2 - n$

Example

$$N=4, \quad h[n] = h[3-n]$$

$$M = \frac{N-1}{2} = \frac{3}{2}$$



Recall that  $\theta(\omega) = k_1 + k_2 \omega$  and let  $k_1 = \frac{\pi}{2}$

$$H(\omega) = A(\omega) e^{j\theta(\omega)} = A(\omega) e^{j\frac{\pi}{2}} e^{jk_2 \omega} = j A(\omega) e^{-jM\omega}$$

$$h[n] = -h[N-n-1] \quad (\text{odd symmetry})$$

$$A(\omega) = \sum_{n=0}^{M-1} 2 h[n] \sin(\omega(M-n))$$

'Type III N ODD'

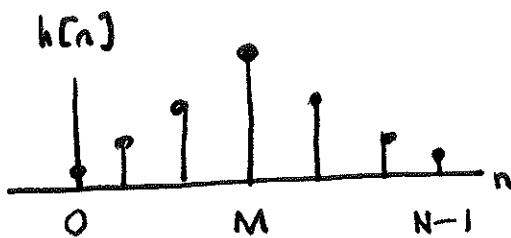
$$A(\omega) = \sum_{n=0}^{N/2-1} 2 h[n] \sin(\omega(M-n))$$

$$= \sum_{n=1}^{N/2} 2 h[\frac{N}{2} - n] \sin(\omega(n - \frac{1}{2}))$$

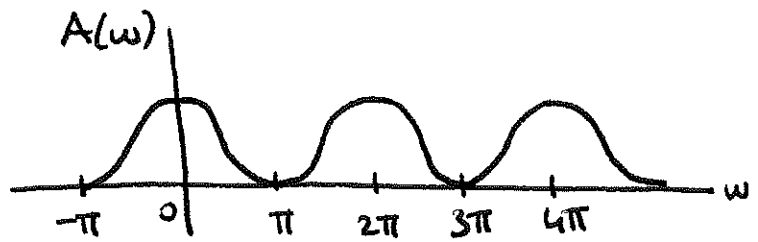
'Type IV N EVEN'

4 possible types that give linear phase.

### Type I N ODD

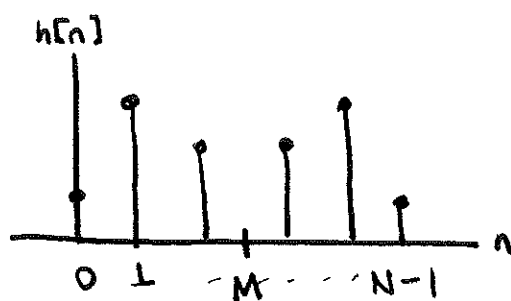


Has odd length  
Even symmetric about  $M$



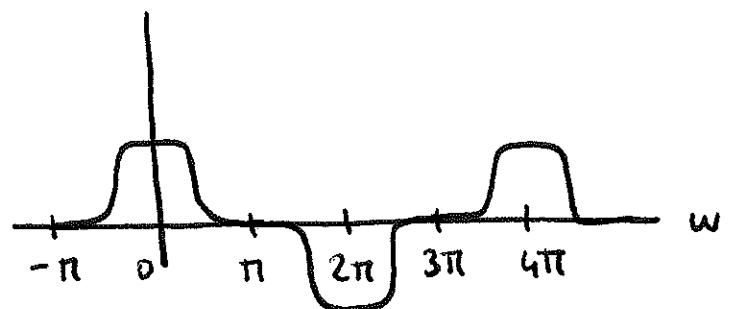
Even about  $w=0, \pi$   
Periodic with  $2\pi$   
Low pass characteristics

### Type II N EVEN



Has even length  
Even symmetric about  $M = \frac{N-1}{2}$

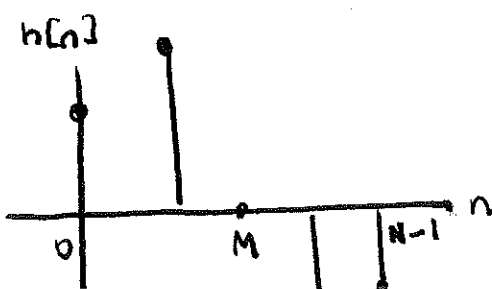
$A(w)$



Even about  $w=0$   
Odd about  $w=\pi$   
Periodic with  $4\pi$

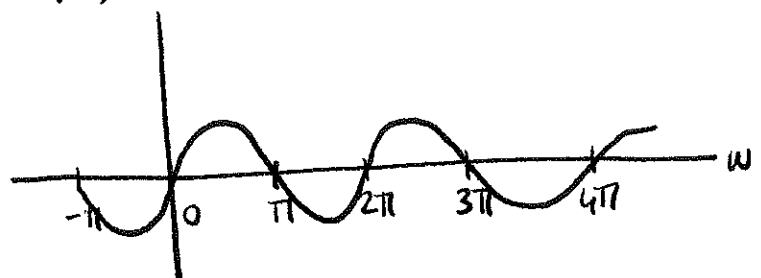
Note: DTFT  $H(w)$  is always  $2\pi$  periodic.  $A(w)$  is a continuous, real valued, smooth function based on both  $H(w)$  and  $\angle H(w)$ :  $A(w) = |H(w)|$

### Type III N ODD



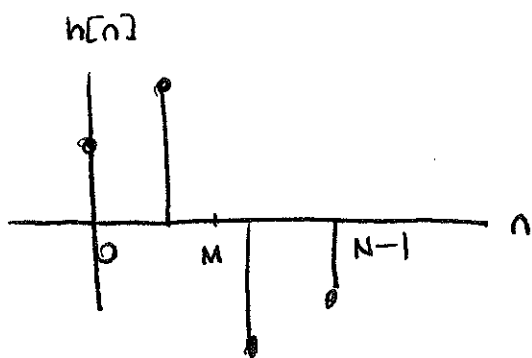
Odd length  
Odd symmetric about  $M$

$A(w)$



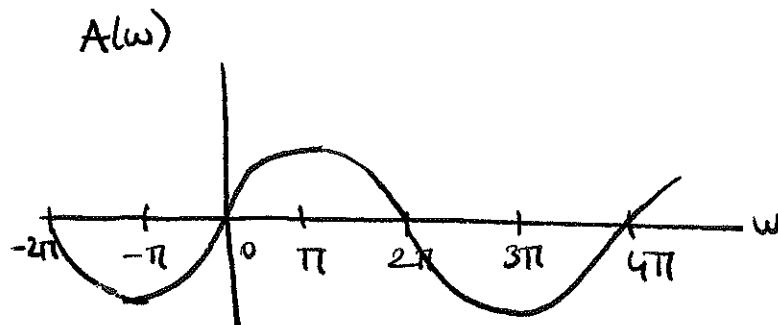
Odd about  $w=0, \pi$   
Periodic with  $2\pi$

## Type IV N EVEN



Even length

Odd symmetric about  $M = \frac{N-1}{2}$



Odd about  $w=0$

Even about  $w=\pi$

Periodic with  $4\pi$

- Notes
- Types III, IV have  $A(0)=0 \Rightarrow$  bad for LPF
  - Types II, III have  $A(\pi)=0 \Rightarrow$  bad for HPF
  - Type III is good for BPF.

Zero Locations for Linear Phase FIR Filters

$$H(z) = \sum_{n=0}^{N-1} h[n] z^{-n}$$

$$= z^{-(N-1)} (h[0] z^{N-1} + h[1] z^{N-2} + \dots + h[N-1])$$

1. If  $h[n]$  is real, then if  $H(z)=0 \Rightarrow H(z^*)=0$

Why?  $H(z^*) = (z^*)^{-(N-1)} (h[0](z^*)^{N-1} + h[1](z^*)^{N-2} + \dots + h[N-1])$

( $h[n]$  real)  $= (z^*)^{-(N-1)} (h^*[0](z^*)^{N-1} + h^*[1](z^*)^{N-2} + \dots + h^*[N-1])$

$$= (H(z))^* = 0$$

2. If  $h[n] = \pm h[N-1-n]$  where  $+$  for Types I-II ,  
 $-$  for Types III-IV

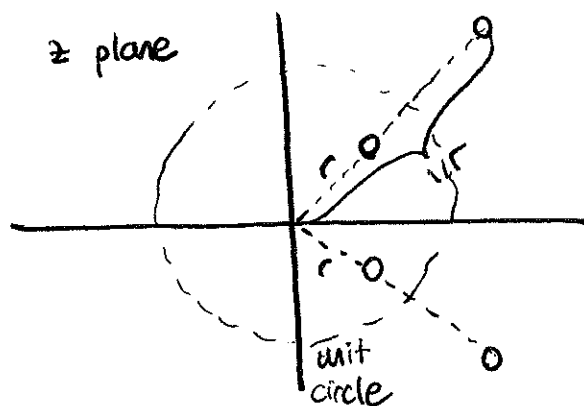
then if  $H(z)=0 \Rightarrow H(\frac{1}{z})=0$

Why? 
$$H(\frac{1}{z}) = z^{N-1} (h[0] z^{-(N-1)} + h[1] z^{-(N-2)} + \dots + h[N-1])$$

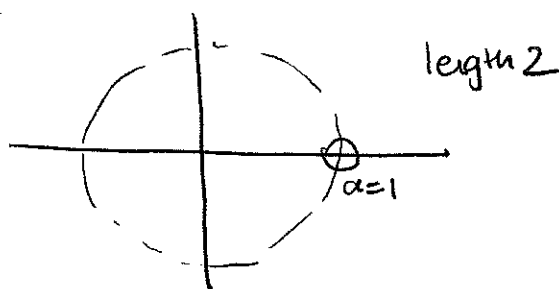
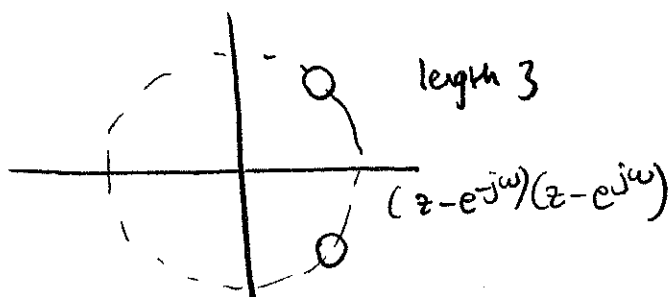
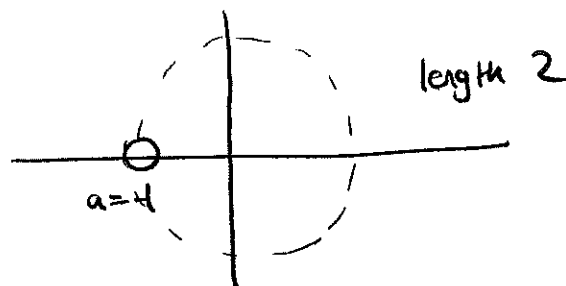
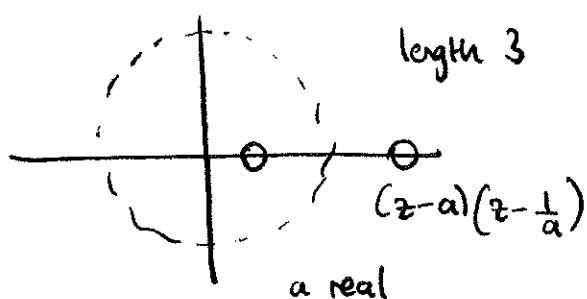
$$= \pm z^{N-1} (h[N-1] z^{-(N-1)} + h[N-2] z^{-(N-2)} + \dots + h[0])$$

$$= \pm z^{N-1} H(z) = 0$$

The zeros of linear phase, real FIR filters look like:

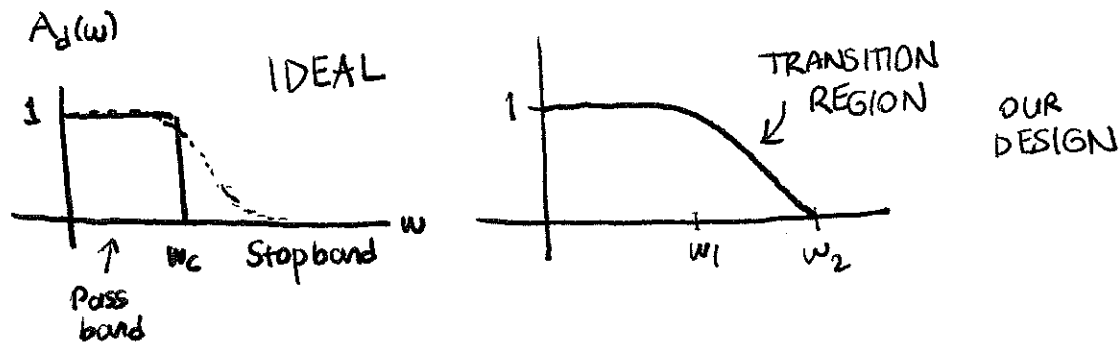


length 5 filter,



\* We could decompose or cascade these decompositions.

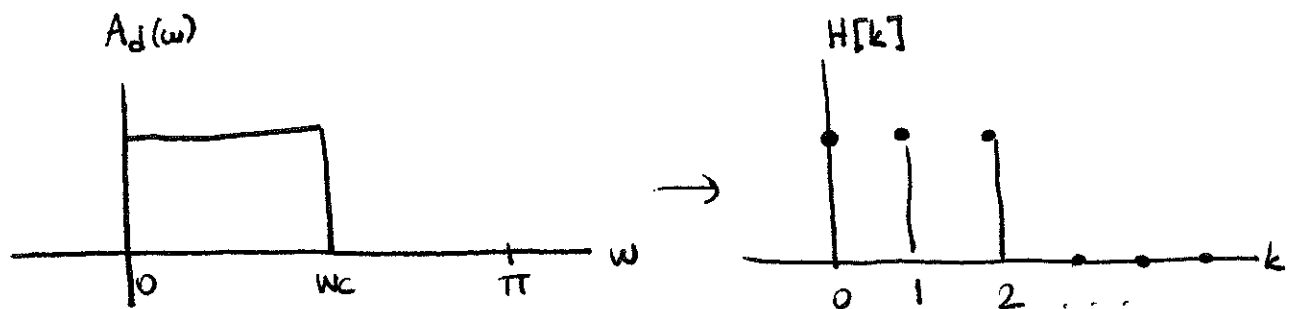
## Different FIR Design Techniques



Different approximations

- Least-squares (LS): to minimize the average or squared error in frequency domain.
- Chebyshev approximation: to minimize the maximum error over certain regions of frequency response.
- Butterworth: Taylor series approximation to desired response.

### 1. Frequency Sampling Design



Assume that we take  $N$  equally spaced samples in the frequency domain at

$$\frac{2\pi k}{N}, \quad k = 0, \dots, N-1$$

The inverse DFT

$$h[n] = \frac{1}{N} \sum_{k=0}^{N-1} H[k] e^{j\frac{2\pi}{N}kn}, \quad n=0, \dots, N-1$$

We have  $N$  equations and  $N$  unknowns  $\Rightarrow$  We can solve this.

$N \log N$  operations to implement DFT. (using FFT algorithm)

Now assume that  $h[n]$  is real, linear phase.

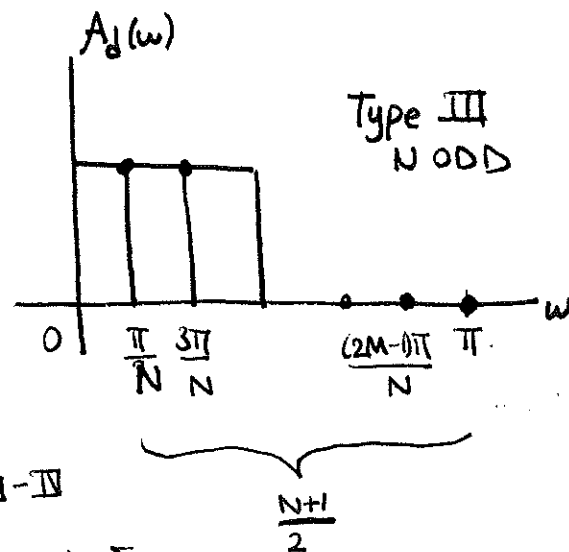
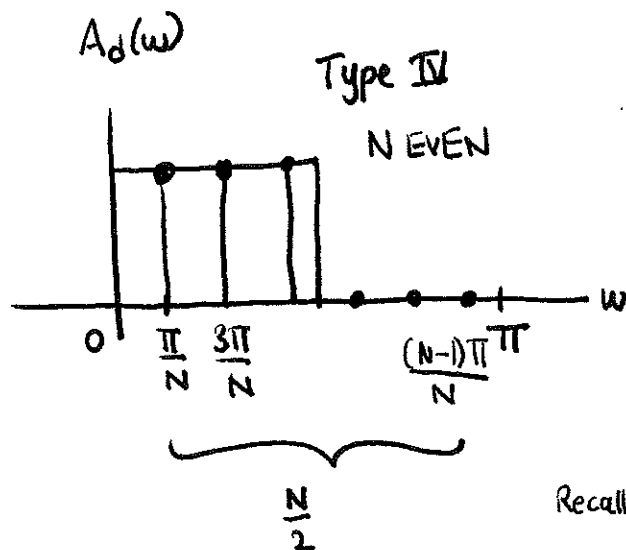
$\Rightarrow$  We can get a gain of 4. Why? \*

$$h[n] = \frac{1}{N} \left[ A[0] + \sum_{k=1}^M 2A[k] \cos \frac{2\pi(n-M)k}{N} \right] \quad \text{Type I}$$

where  $M = \frac{N-1}{2}$  and  $A[0], \dots, A[M]$  are samples of amplitude response

\* because  $M \approx \frac{N}{2}$  and we only have the "cos" terms.

We have equivalent forms for other types (Types II, III, IV)



Recall for Types III-IV

$$\sum_n h[n] = 0 \Rightarrow H(0) = \sum h[n] = 0$$

Note: We could also use non-uniform sampling. However, in that case we can no longer use IDFT.

## 2. Least-Squares (LS) error in frequency domain.

a)  $L \geq N$  discrete frequency samples where  $L$  is the length of the filter

$$E = \sum_{k=0}^{L-1} |A(\omega_k) - A_d(\omega_k)|^2 \quad \text{mean-squared error}$$

$$\omega_k = \frac{2\pi k}{L}, \quad k=0, \dots, L-1, \quad \text{with amplitude samples } A_d(\omega_k).$$

From Parseval's theorem,

$$E = \sum_{n=-\frac{L-1}{2}}^{\frac{L-1}{2}} |h[n] - h_d[n]|^2$$

where  $h_d[n]$  is a length- $L$  FIR filter.

Using  $M = \frac{N-1}{2}$  we rewrite  $E$ :

$$E = \sum_{n=-M}^M |h[n] - h_d[n]|^2 + \underbrace{\sum_{n=M+1}^{(L-1)/2} 2|h_d[n]|^2}_{\text{residual error,}}$$

$\downarrow$                        $\downarrow$   
 length  $N$       length  $L$

To minimize  $E$ , take  $h[n]$  to agree with  $h_d[n]$  on first  $N$  filter taps, i.e., truncate  $h_d[n]$ .

To adjust the 'residual error', choose  $N$  appropriately.



## Solving a LS problem

Recall that the amplitude response of a linear phase filter is

$$A(\omega) = \sum_{n=0}^{M-1} 2h[n] \cos(\omega(M-n)) + h[M] \quad (\text{Type I } N \text{ ODD})$$

$$a = F h$$

$\downarrow$   
 $L \times 1$   
 vector

$\downarrow$   
 $L \times (M+1)$   
 matrix

$\downarrow$   
 $(M+1) \times 1$   
 vector

where  $L > M+1$ , i.e.,  $F$  is a tall matrix.

$$F = \begin{bmatrix} \end{bmatrix}$$

$$F^T a = \overbrace{F^T F}^{\text{invertible}} h \quad \text{where } F^T \text{ is the transpose.}$$

$$\hat{h} = \underbrace{(F^T F)^{-1} F^T}_{(M+1) \times (M+1) \text{ matrix}} \alpha \quad \text{is the solution that minimizes}$$

$$(Fh - a_d)^T (Fh - a_d)$$