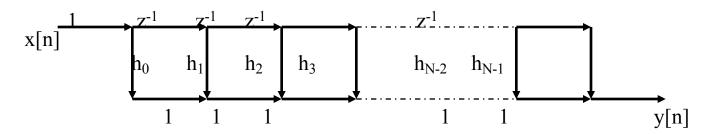
Finite Impulse Response (FIR) Digital Filters (II) <u>Ideal Impulse Response Design</u> <u>Examples</u>

Yogananda Isukapalli

• FIR Filter Design Problem

Given H(z) or H(e^{jw}), find filter coefficients { b_0 , b_1 , b_2 ,, b_{N-1} } which are equal to { h_0 , h_1 , h_2 ,, h_{N-1} } in the case of FIR filters.



Consider a general (infinite impulse response) definition:

$$H(z) = \sum_{n=-\infty}^{\infty} h[n]_{Z}^{-n}$$

From complex variable theory, the inverse transform is:

$$h[n] = \frac{1}{2\pi j} \oint_C H(z) z^{n-1} dz$$

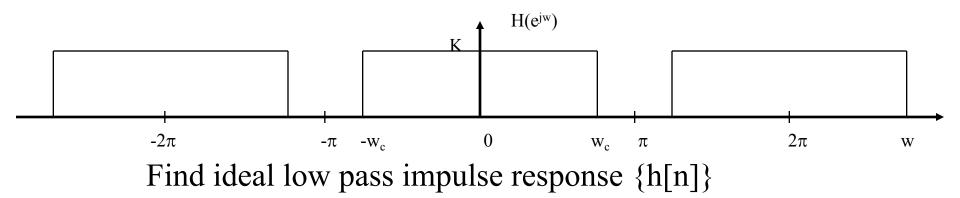
Where C is a counterclockwise closed contour in the region of convergence of H(z) and encircling the origin of the z-plane

• Evaluating H(z) on the unit circle ($z = e^{jw}$):

$$H(e^{jw}) = \sum_{n=-\infty}^{\infty} h[n]e^{-jnw}$$

$$h[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} H(e^{jw}) e^{jnw} dw \quad \text{where } dz = j e^{jw} dw$$

• Design of an ideal low pass FIR digital filter



$$h_{LP}[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} H(e^{jw}) e^{jnw} dw$$

$$=\frac{1}{2\pi}\int_{-w_c}^{w_c} Ke^{jnw}dw$$

Hence $h_{LP}[n] = \frac{K}{n\pi} \sin(nw_c)$

$$n = 0, \pm 1, \pm 2, \ldots \pm \infty$$

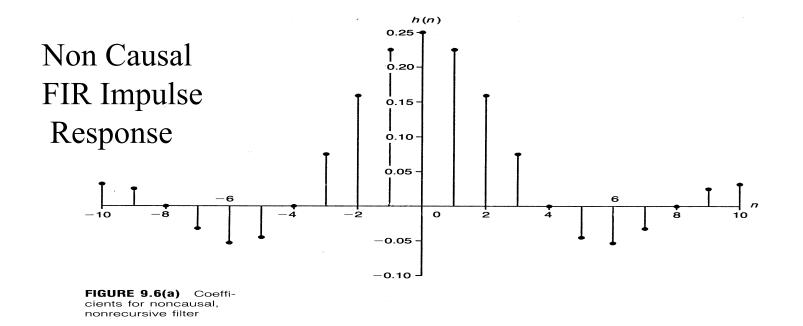
Let
$$K = 1$$
, $w_c = \pi/4$, $n = 0, \pm 1, ..., \pm 10$

The impulse response coefficients are

n = 0,	h[n] = 0.25	$n = \pm 4,$	h[n] = 0
$=\pm1,$	= 0.225	$=\pm5,$	= -0.043
$=\pm2,$	= 0.159	$=\pm 6,$	= -0.053
$=\pm3,$	= 0.075	$=\pm7,$	= -0.032

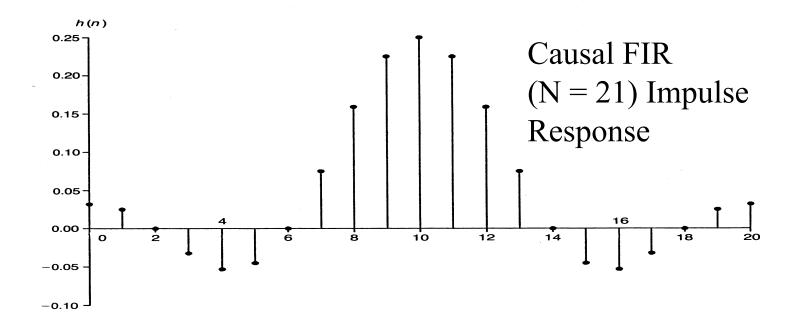
$$n = \pm 8, \quad h[n] = 0$$

= $\pm 9, = 0.025$
= $\pm 10, = 0.032$



We can make it causal if we shift $h_{LP}[n]$ by 10 units to the right:

$$h_{LP}[n] = \frac{K}{(n-10)\pi} \sin((n-10)w_c)$$
 $n = 0, 1, 2, ..., 20$

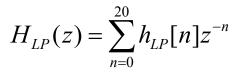


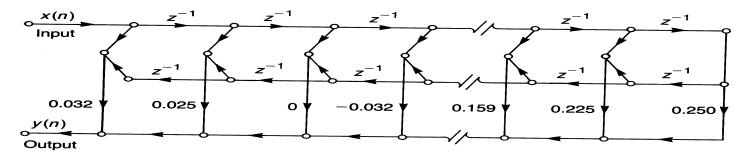
Notice the symmetry: h[n] = h[N-1-n] which satisfies the linear phase condition.

Filter Transfer Function

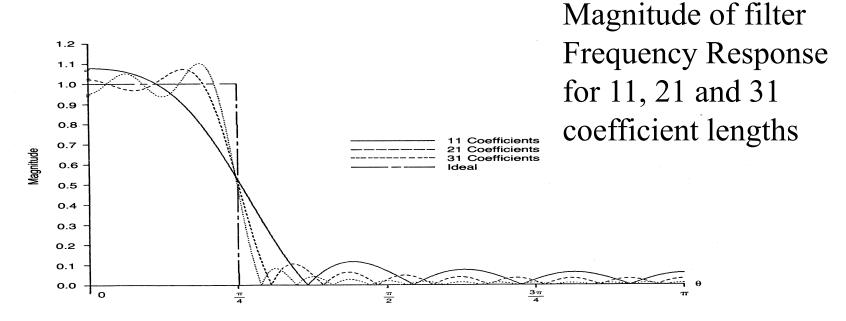
$$H_{LP}(e^{jw}) = \sum_{n=0}^{20} h_{LP}[n]e^{-jnw}$$
 since $z = e^{jw}$

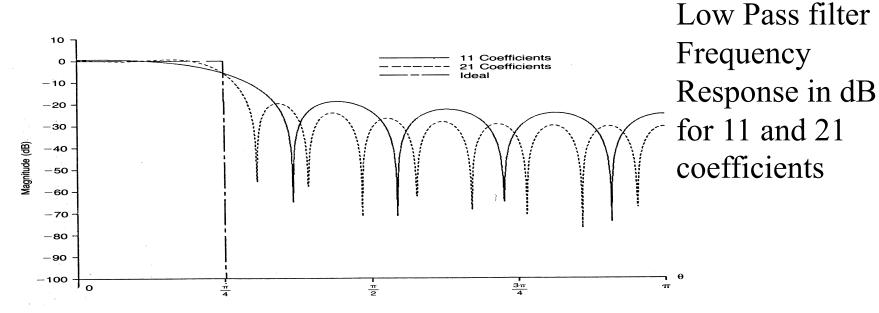
Implies

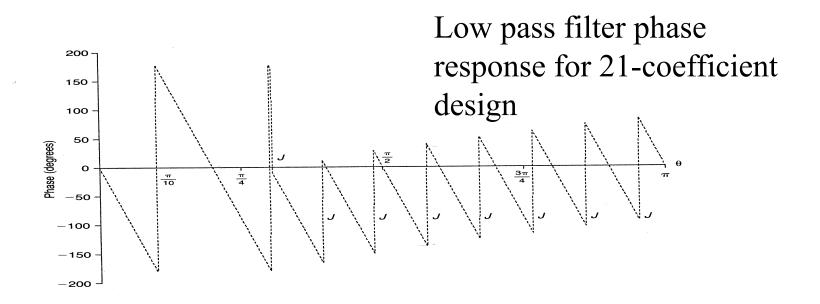


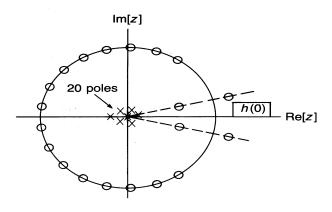


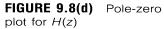
Filter Implementation for 21 coefficients











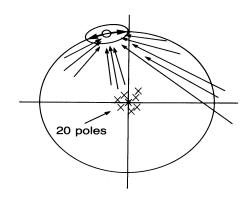
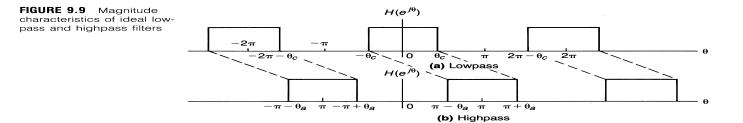


FIGURE 9.8(e) Vectors for frequency response near $\theta = 1.813$



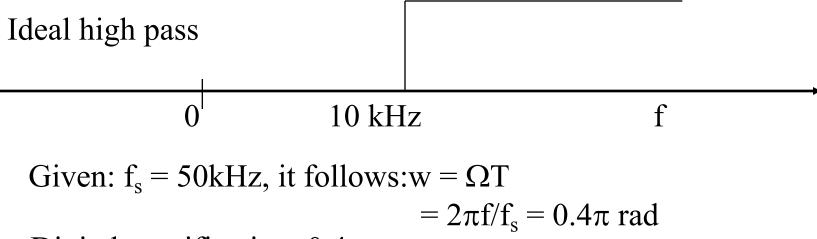
The above figure implies

$$H_{HP}(e^{jw}) = H_{LP}(e^{j(w-\pi)})$$
$$h_{HP}[n] = h_{LP}[n](-1)^n$$

Now:

$$H_{HP}(e^{jw}) = \sum_{n=-\infty}^{\infty} h_{LP}[n]e^{-jn(w-\pi)} = \sum_{n=-\infty}^{\infty} h_{LP}[n]e^{-jnw}(-1)^{n}$$
$$= \underbrace{\sum_{n=-\infty}^{\infty} h_{LP}[n](-1)^{n}e^{-jnw}}_{h_{HP}[n]}$$

Example: Analog high pass filter specification



Digital specification: $0.4\pi = \pi - w_a$

:.
$$w_c (LPF) = \pi - w_a = \pi - 0.4\pi = 0.6\pi$$

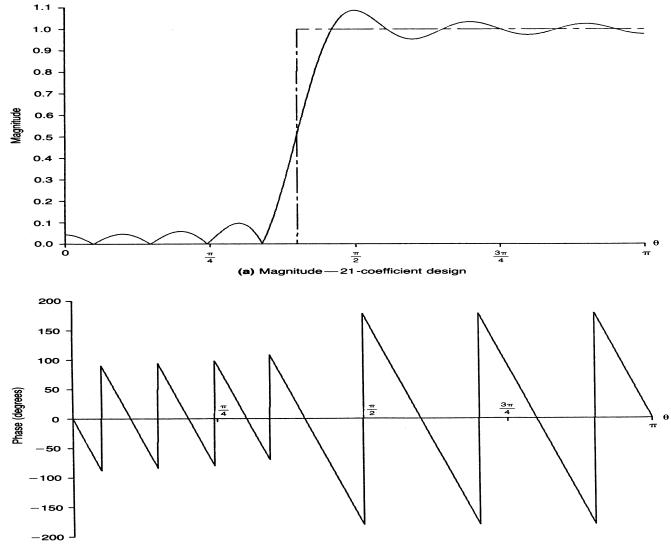
First design ideal LPF with $w_c = 0.6\pi$ Answer: $h_{LP}[n] = \frac{1}{n\pi} \sin(0.6\pi n)$ $n = 0, \pm 1, \pm 2, \dots \pm \infty$

Follows now:

$$h_{HP}[n] = h_{LP}[n](-1)^n$$
$$h_{HP}[n] = \frac{(-1)^{n-I}}{\pi(n-I)}\sin(0.6\pi(n-I))$$

The I indicates the length of the filter

Hence the above filter is a causal FIR high pass digital filter



(b) Phase - 21-coefficient design

FIGURE 9.10 Highpass filter frequency response

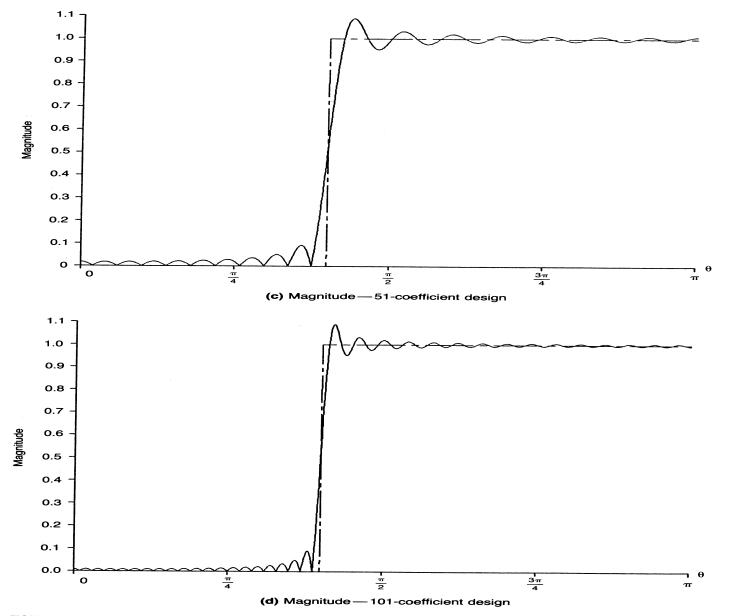
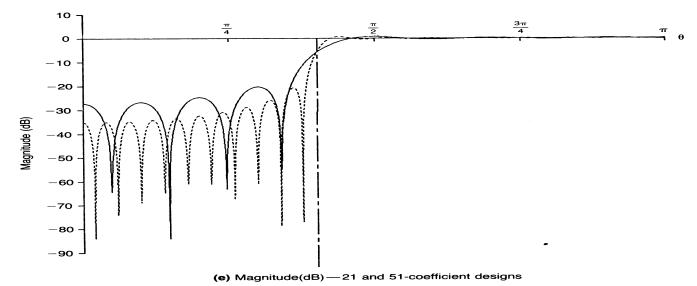


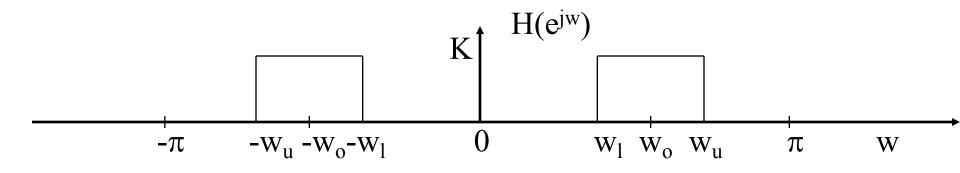
FIGURE 9.10 Highpass filter frequency response



F 9.10 Highpass fil-

FIGURE 9.10 Highpass filter frequency response

Bandpass FIR filter design:



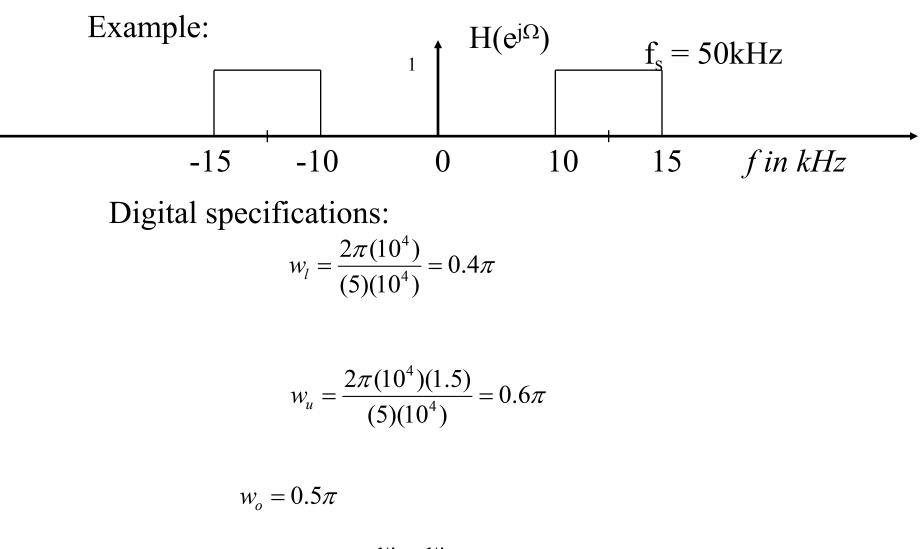
Shift to right: H(e^{jw}) e^{jwo} Shift to left: H(e^{jw}) e^{-jwo} Follows:

$$h_{BP}[n] = [2\cos nw_o]h_{LP}[n]$$

 $n = 0, \pm 1, \pm 2, \dots \pm I$

where

$$w_u - w_l = 2w_c(lowpass)$$
$$w_o = \frac{w_u + w_l}{2}$$



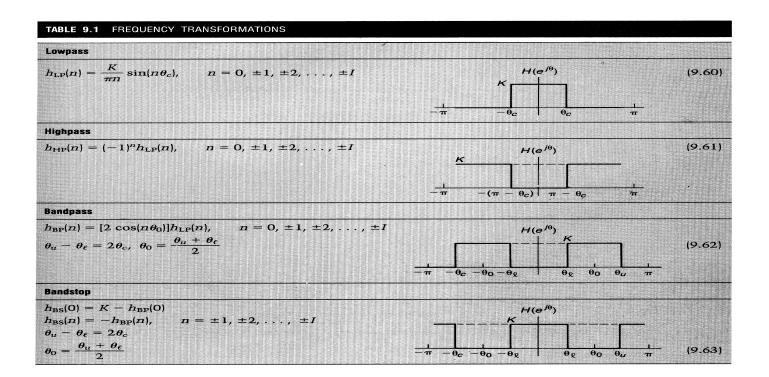
$$w_c = \frac{w_u - w_l}{2} = 0.1\pi$$

First design LPF:

$$h_{LP}[n] = \frac{1}{n\pi} \sin(0.1\pi n)$$
 $n = 0, \pm 1, \pm 2, \dots$

Follows:

$$h_{BP}[n] = [2\cos(0.5\pi n)h_{LP}[n] \quad n = 0, \pm 1, \pm 2, \dots$$



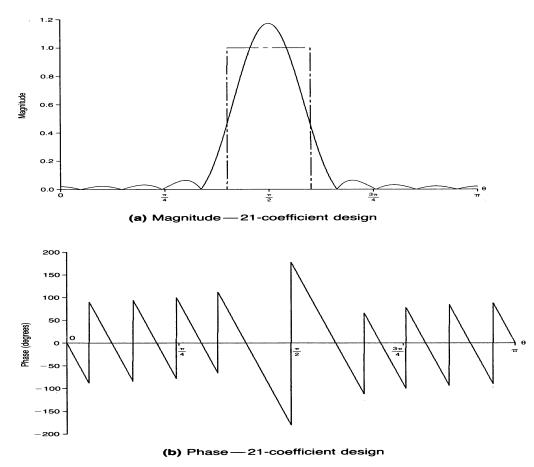


FIGURE 9.12 Bandpass filter frequency response

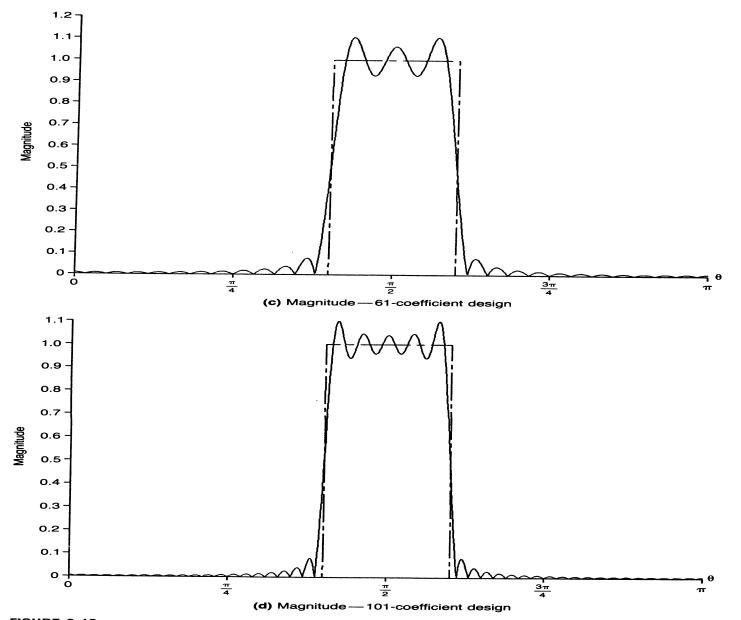
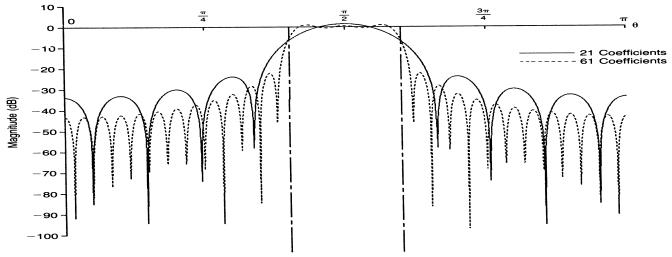


FIGURE 9.12 Bandpass filter frequency response



(e) Magnitude (dB)-21- and 61-coefficient designs

Bandpass filter frequency response

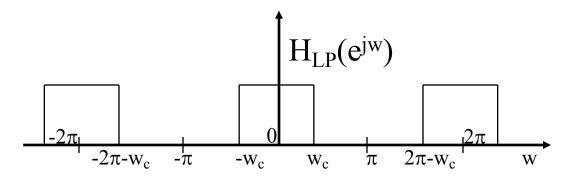
Homework # 6

- 1. Design a length-5 FIR bandpass filter with an antisymmetric impulse response h[n], i.e. h[n] = -h[4-n], $0 \le n \le 4$, satisfying the following magnitude response values : $|H(e^{\frac{j\pi}{4}})|=0.5 \text{ and} |H(e^{\frac{j\pi}{2}})|=1$. Determine the exact expression for the frequency response of the filter designed.
- 2. An FIR filter of length 5 is defined by a symmetric impulse response i.e. $h[n]=h[4-n], 0 \le n \le 4$,. Let the input to this filter be a sum of 3 cosine sequences of angular frequencies: 0.2 rad/samples, 0.5 rad/samples, and 0.8 rad/samples, respectively. Determine the impulse response coefficients so that the filter passes only the midfrequency component of the input.
- 3. The frequency response $H(e^{j\omega})$ of a length-4 FIR filter with a real and antisymmetric impulse response has the following specific values: $H(e^{j\pi})=8$, and $H(e^{j\pi/2})=-2+j2$. Determine H(z).

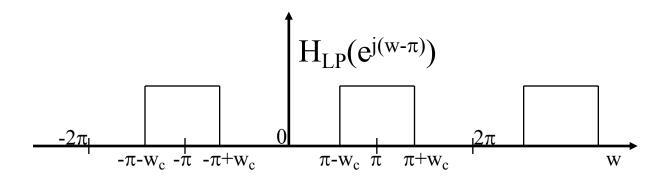
Frequency transformations of ideal filters.

High pass Filter

The ideal low filter frequency response is shown below. $h_{LP}[n]$ represents the non-causal impulse response of $H_{LP}(e^{jw})$



Shifting the low pass filter frequency by π , we get



The above figure is the frequency response of an ideal high pass filter with a cutoff frequency of π -w_c (H_{HP}(e^{jw})).Hence,

$$H_{HP}(e^{jw}) = H_{LP}(e^{j(w-\pi)})$$

From the shifting property of the Fourier transform, we get

 $h_{HP}[n] = h_{LP}[n]e^{j\pi n}$

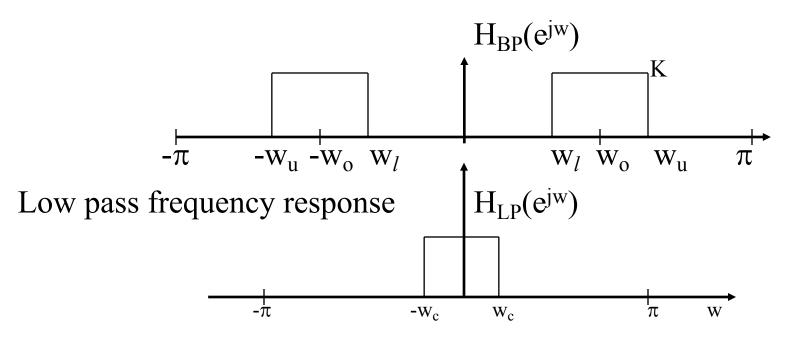
where
$$h_{HP}[n] = (-1)^n h_{LP}[n]$$

 $h_{LP}[n] = \frac{1}{\pi n} \sin(nw_c)$

Hence, to design an ideal high pass filter with cutoff frequency of w_a , first design a ideal low pass filter with a cutoff frequency of $(\pi - w_a)$. Then use the above transformation to obtain the impulse response of the ideal high pass filter.

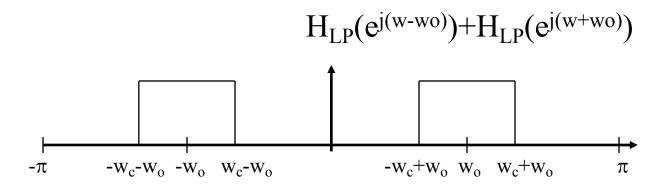
Band pass Filter

The frequency response for an ideal band pass filter is shown below. The center frequency is w_0 .



Looking at the frequency responses of a band pass filter and a low pass filter, we can observe that a band pass filter is obtained by shifting the low pass filter to the left and to the right by w_0 and adding the two shifted responses.

Shifting the low pass filter by wo to the left and to the right and adding, we get



Comparing the ideal band pass filter and the above figure, we can observe that

$$w_{l} = -w_{c} + w_{o}$$
$$w_{u} = w_{c} + w_{o}$$
$$w_{c} = \frac{w_{u} - w_{l}}{2}$$
$$w_{o} = \frac{w_{u} + w_{l}}{2}$$

The frequency response of the band pass filter $H_{BP}(e^{jw})$ is

$$H_{BP}(e^{jw}) = H_{LP}(e^{j(w-w_o)}) + H_{LP}(e^{j(w+w_o)})$$

From the shifting property of the Fourier transform, we have

 $h_{BP}[n] = h_{LP}[n]e^{jw_o n} + h_{LP}[n]e^{-jw_o n}$

 $h_{BP}[n] = h_{LP}[n] \{ e^{jw_o n} + e^{-jw_o n} \}$

 $h_{BP}[n] = h_{LP}[n] \{ 2\cos(nw_o) \}$

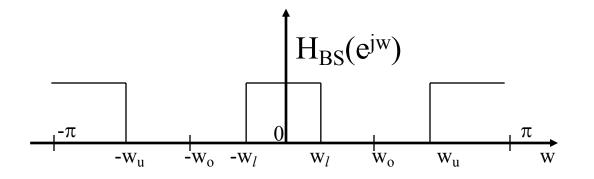
To design a band pass filter, first design a low pass filter with cutoff frequency w_c given by

$$w_c = \frac{w_u - w_l}{2}$$

And use the above transformation to get the impulse response of the band pass filter . 27

Band stop Filter

The frequency response of an ideal band stop filter $(H_{BS}(e^{jw}))$ is



Looking at the frequency responses of band stop and band pass filters, we observe that the band stop filter is obtained by subtracting the band pass from 1.

$$H_{BS}(e^{jw}) = 1 - H_{BP}(e^{jw})$$

$$\sum_{n=-\infty}^{\infty} h_{BS}[n]e^{-jnw} = 1 - \sum_{n=-\infty}^{\infty} h_{BP}[n]e^{-jnw}$$

If we work out the first few terms, we get

$$\dots + h_{BS}[-2]e^{j^{2w}} + h_{BS}[-1]e^{j^{w}} + h_{BS}[0] + h_{BS}[1]e^{-j^{w}} + h_{BS}[2]e^{-j^{2w}}$$
$$= \dots + h_{BP}[-2]e^{j^{2w}} + h_{BP}[-1]e^{j^{w}} + 1 - h_{BP}[0] + h_{BP}[1]e^{-j^{w}} + h_{BP}[2]e^{-j^{2w}}$$

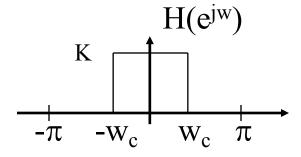
Hence, the impulse response coefficients for a band stop filter are obtained as

$$h_{BS}[0] = 1 - h_{BP}[0] = 1 - 2h_{LP}[0]$$

$$h_{BS}[n] = -h_{BP}[n] = -h_{LP}[n] \{2\cos(nw_o)\}$$
 $n \neq 0$

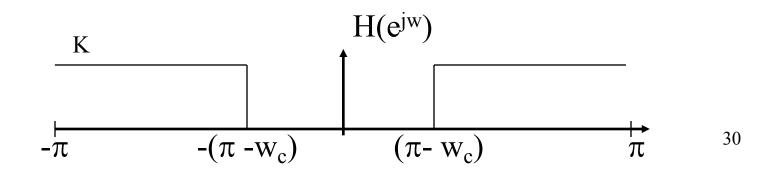
Low pass

$$h_{LP}[n] = \frac{K}{n\pi} \sin(nw_c)$$
 $n = 0, \pm 1, \pm 2, ..., \pm I$

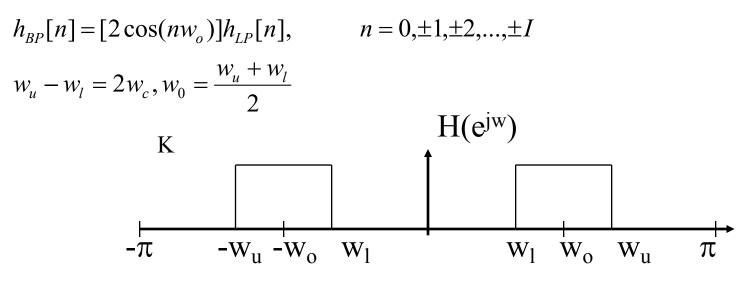


High pass

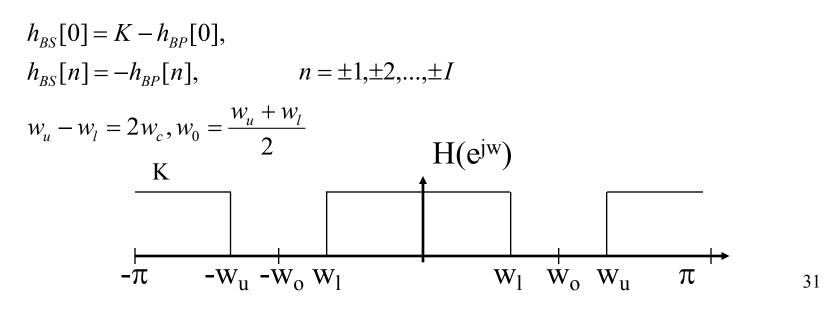
 $h_{HP}[n] = (-1)^n h_{LP}[n], \qquad n = 0, \pm 1, \pm 2, \dots, \pm I$



Band pass



Band stop



Examples

1. An FIR filter is defined by a symmetric impulse response, i.e. h[0] = h[2]. Input to the filter is a sum of two cosine sequences of angular frequencies 0.2 rad/s and 0.5 rad/s

Determine the impulse response coefficients so that it passes only the high frequency component of the input

Solution:

Since h[0] = h[2]
$$\longrightarrow$$
 $H(e^{j\omega}) = h[0](1 + e^{-j2\omega}) + h[1]e^{-j\omega}$
= h[0] $e^{-j\omega}(e^{j\omega} + e^{-j\omega}) + h[1]e^{-j\omega}$
= $e^{-j\omega}(2h[0]\cos(\omega_0) + h[1])$

Since only high frequency component can pass through the filter, *when*,

	$\omega_0 = 0.2$	\Rightarrow	$\mathrm{H}(\mathrm{e}^{\mathrm{j}\omega_0})=0$	and
	$\omega_0 = 0.5$	\Rightarrow	$H(e^{j\omega_0})=1$	
i.e.	$H(e^{j0.2}) =$	2h[$[0]\cos(0.2) + h[$	1] = 0
	$H(e^{j0.5}) =$	2h[$0]\cos(0.5) + h[$	1] = 1

Solving these two equations,

$$h[0] = h[2] = -4.8788$$
 and $h[1] = 9.5631$ 32

2. Design a length – 4 FIR bandpass filter with an anti symmetric impulse response i.e. h[n] = h[-n-4], for 0 n 4 satisfying the following magnitude response Values: $|H(e^{j\pi/4})| = 1$ and $|H(e^{j\pi/2})| = 0.5$

$$H(e^{j\omega}) = h_{o} + h_{1}e^{-jw} + h_{2}e^{-j^{2}w} + h_{3}e^{-j^{3}w}$$

$$= h_{o} + h_{1}e^{-jw} - h_{1}e^{-j^{2}w} - h_{0}e^{-j^{3}w}$$

$$= h_{0}(1 - e^{-j^{3}w}) + h_{1}e^{-jw}(1 - e^{-j^{3}w})$$

$$= h_{0}e^{-j^{3}w/2}(e^{j^{3}w/2} - e^{-j^{3}w/2}) + h_{1}e^{-jw}(e^{-j^{3}w/2}(e^{j^{3}w/2} - e^{-j^{3}w/2}))$$

$$= je^{-j^{3}w/2}(2h_{0}\sin(3w/2) + 2h_{1}\sin(w/2))$$

Since we have $| H(e^{j\pi/4}) |= 1$ and $| H(e^{j\pi/2}) |= 0.5$

$$H(e^{j\pi/4}) = 2h_0 \sin(3\pi/8) + 2h_1 \sin(\pi/8) = 1$$

and
$$H(e^{j\pi/2}) = 2h_0 \sin(3\pi/4) + 2h_1 \sin(\pi/4) = 0.5$$

Solving equations, we get

$$h_0 = -0.0381$$
 and $h_1 = 0.7452$

3. The frequency response of a length-4 FIR filter has values:

$$|H(e^{j0})|=2$$
 $|H(e^{j\pi/2})|=7-j3$ and $|H(e^{j\pi})|=0$

Determine H(z)

Solution

Using the symmetry property of DTFT of a real sequence, we observe that

$$|H(e^{j3\pi/2})| = |H^*(e^{j\pi/2})| = 7 + j3$$

Thus the 4 point DFT of the sequence is given by:

$$H(k) = [2, 7 - j3, 0, 7 + j3]$$

The inverse DFT of the H(k) gives h(n) as:

h(n) = (4 , 2 , -3 , -1) (using **ifft** in MATLAB)

Therefore H(z) is:

$$H(z) = 4 + 2z^{-1} - 3z^{-2} - z^{-3}$$