

# EN 2063 Project 200029B

## Report

$$A=0, B=2, C=9$$

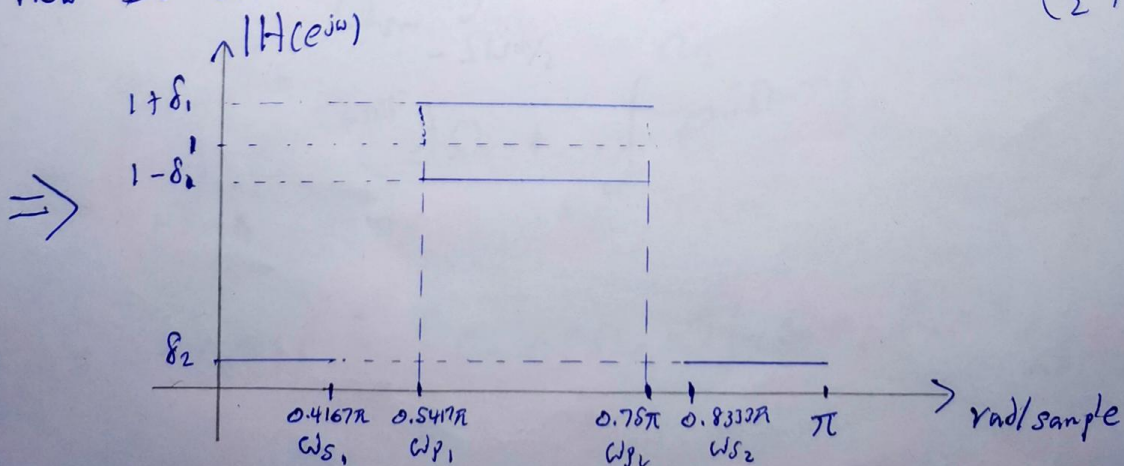
Parameter	Value
Maximum passband ripple, $\tilde{A}_p$	0.1 dB
Minimum stopband attenuation, $\tilde{A}_s$	52 dB
Lower passband edge, $\Omega_{p1}$	1300 rad/s
Upper passband edge, $\Omega_{p2}$	1800 rad/s
Lower stopband edge, $\Omega_{s1}$	1000 rad/s
Upper stopband edge, $\Omega_{s2}$	2000 rad/s
Sampling frequency, $\Omega_{sm}$	4800 rad/s

We can convert the above analog domain frequencies into digital domain frequencies using the relation,

$$\omega = \frac{2\pi\Omega}{\Omega_{sm}} \Rightarrow$$

$\omega$ (rad/sample)	$\Omega$ (rad/sec)
$\omega_{p1} = 0.5417\pi$	$\Omega_{p1} = 1300$
$\omega_{p2} = 0.75\pi$	$\Omega_{p2} = 1800$
$\omega_{s1} = 0.4167\pi$	$\Omega_{s1} = 1000$
$\omega_{s2} = 0.8333\pi$	$\Omega_{s2} = 2000$

\* now It can be see that  $\omega_{p1}, \omega_{p2}, \omega_{s1}, \omega_{s2} < \left(\frac{\omega_{sm}}{2}\right)$  Nyquist rate.



1) Designing the FIR filter using Kaiser window to above BPF specification.

- Cut off frequency calculation.

$$B_t = \min[(\Omega_{p1} - \Omega_{s1}), (\Omega_{s2} - \Omega_{p2})]$$

$$= \min[(1300 - 100), (2000 - 1800)]$$

$$= \min[1200, 200] = 200 \text{ rad/s}$$

$$\therefore \Omega_{c1} = \Omega_{p1} - \frac{B_t}{2}, \quad \Omega_{c2} = \Omega_{p2} + \frac{B_t}{2}$$

$$\Omega_{c1} = 1200 \text{ rad/s}$$

$$\Omega_{c2} = 1900 \text{ rad/s}$$

~~Frequency~~ Frequency Response of the Ideal BPF

$$H(e^{j\Omega T}) = \begin{cases} 1 & \text{for } -\Omega_{c2} \leq \Omega \leq -\Omega_{c1} \\ 1 & \text{for } \Omega_{c1} \leq \Omega \leq \Omega_{c2} \\ 0 & \text{otherwise} \end{cases}$$

where  $T$  sample period.

Ideal Impulse Response

$$h(nT) = \frac{1}{\Omega_{sm}} \int_{-\Omega_{sm}/2}^{\Omega_{sm}/2} H(e^{j\Omega T}) e^{j\Omega nT} d\Omega$$

$$= \frac{1}{\Omega_{sm}} \left[ \int_{-\Omega_{c2}}^{-\Omega_{c1}} e^{j\Omega nT} d\Omega + \int_{\Omega_{c1}}^{\Omega_{c2}} e^{j\Omega nT} d\Omega \right]$$

$$\Rightarrow h(nT) = \frac{1}{n\pi} [\sin c[\Omega_{c2} nT] - \sin c[\Omega_{c1} nT]]$$



• Kaiser window parameters

$$\beta_1 = \frac{10^{0.05(\tilde{A}_p - 1)}}{10^{0.05(\tilde{A}_p)} + 1} = 0.00576$$

$$\beta_2 = 10^{-0.05(\tilde{A}_a)} = 0.002512$$

$$\beta = \min \{ \beta_1, \beta_2 \} = 0.002512$$

∴ parameter alpha ( $\alpha$ ) (or  $\beta$  parameter)

$$\alpha = 0.1102 (\tilde{A}_a - 8.7) = 4.77166$$

parameter D,

$$D = \frac{\tilde{A}_a - 7.95}{14.36} = 3.0675$$

order of the filter

$$M = \left\lceil \frac{\Omega_s D}{B_t} \right\rceil = 74$$

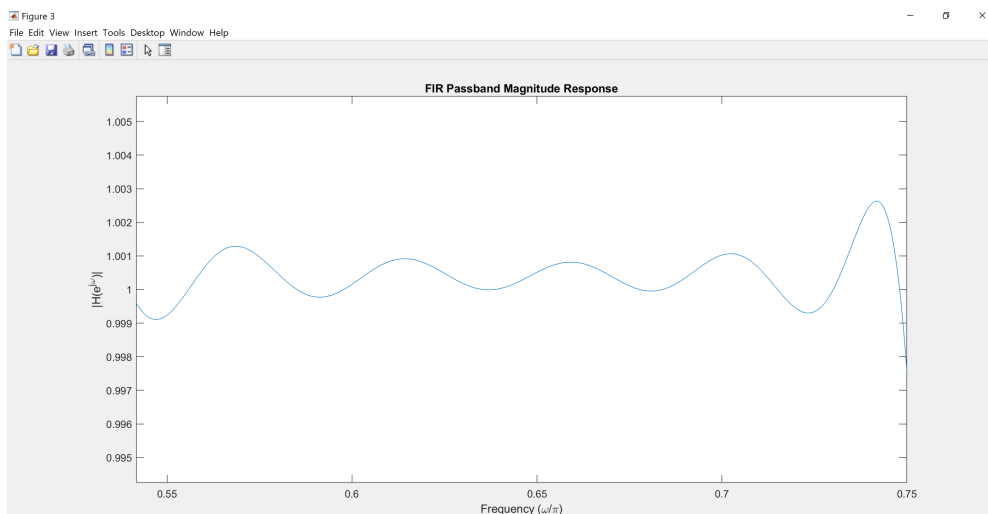
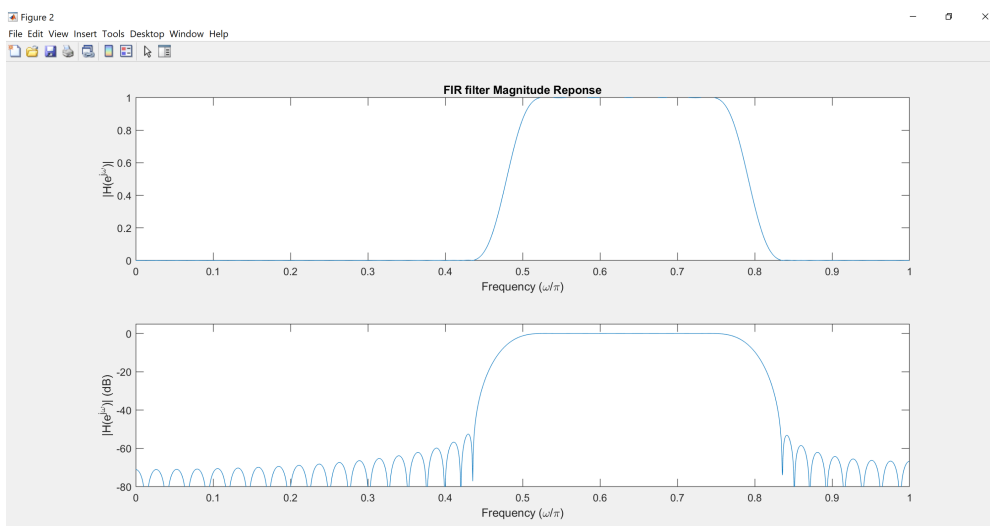
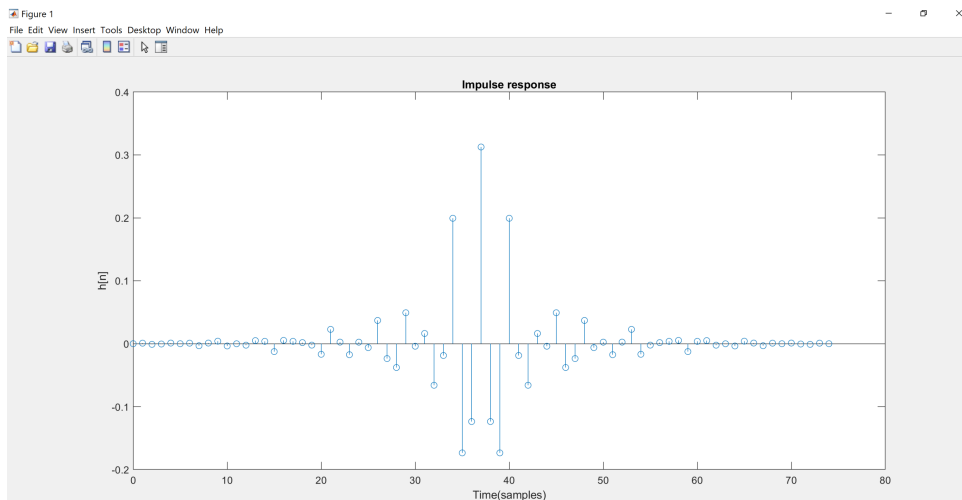
\*\* In matlab order is determined by kaiserord function.

length of the Kaiser window

$$N = M + 1 = 75$$

now using the Kaiser window function, the equation of Kaiser window  $W_K(nT)$  can be generated.

this  $w_n(nT)$  is symmetric around 0 and have  $N$  points.



now by shifting  $\left(\frac{N-1}{2}\right)$  both of the  $h(nT)$  and  $w_n(nT)$  we have causal impulse response for the BPF which  $N$  points.

2) Designing the IIR filter using Bilinear transformation method to have specified BPF.

$D=1 \Rightarrow$  we have to realize the BPF using Chebyshev filter approximation.

first we have to have digital frequencies of the frequencies Analog Domain :-  $\omega_{s1}, \omega_{p1}, \omega_{p2}, \omega_{s2}$

then we do mapping

$$\Omega = \frac{2}{T} \tan\left(\frac{\omega}{2}\right)$$

this equation is used to pre-warb the digital frequencies and have s-domain frequencies.

since  $T$  is arbitrary we take  $T=1$  for clarity. Then transformation become

$$\Omega = 2 \tan\left(\frac{\omega}{2}\right)$$



now we have s domain prewarped ~~and~~  
BPF filter frequencies

$$\Omega'_s, \Omega'_p, \Omega'_r, \Omega'_s$$

now using the cheb1ord function  
we get the order of the  
chebyshev filter and using ~~cheb1~~

~~cheb1ord~~ ~~function~~ cheb1 we set  
Numerator, and Denominator coefficients  
of the H(z)

now using

$$S = \frac{2}{T} \left( \frac{z-1}{z+1} \right) \quad \swarrow \text{Bilinear transformation.} \quad \text{let } T=1$$

we get the H(z) of filter.  
this is done by bilinear() function which  
gives the coefficient of Numerator and  
Denominator of H(z).

~~using~~ ~~impz()~~ ~~function~~ ~~we~~ ~~can~~ ~~see~~ ~~that~~

using impz() function we can see that  
h(z) impulse response of the filter is  
Causal have ~~247~~ 347 points.

03) length of impulse response of  
FIR filter,  $h_{FIR(nT)} = 75$

length of impulse response of  
IIR filter,  $h_{IIR(nT)} = 347$

$\Rightarrow$  ~~less~~ more computational power need  
for IIR filter

2) a)

	Numerator Coefficients	Denominator Coefficients
$Z^{12}$	0.0002	1
$Z^{11}$	0	4.8254
$Z^{10}$	-0.009	14.1501
$Z^9$	0	28.6998
$Z^8$	0.0023	45.0606
$Z^7$	0	55.9344
$Z^6$	-0.003	56.6232
$Z^5$	0	46.4895
$Z^4$	0.0023	31.1138
$Z^3$	0	16.4405
$Z^2$	-0.0009	6.72
$Z^1$	0	1.8949
$Z^0$	0.0002	0.3273

