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EE125: Digital Signal Processing Matlab Project 6 Alexander Christenson

```
clear all, close all
```

Designing a filter by frequency sampling

```
% Plugging in the given information
fs = 5000; % Sampling Freq, Hz
L = 23; % Filter length
fc = 750;

% Creating the frequency vector
df = fs/(L-1);
M = (L-1)/2;
f = [(0:M)*df (-M:-1)*df];

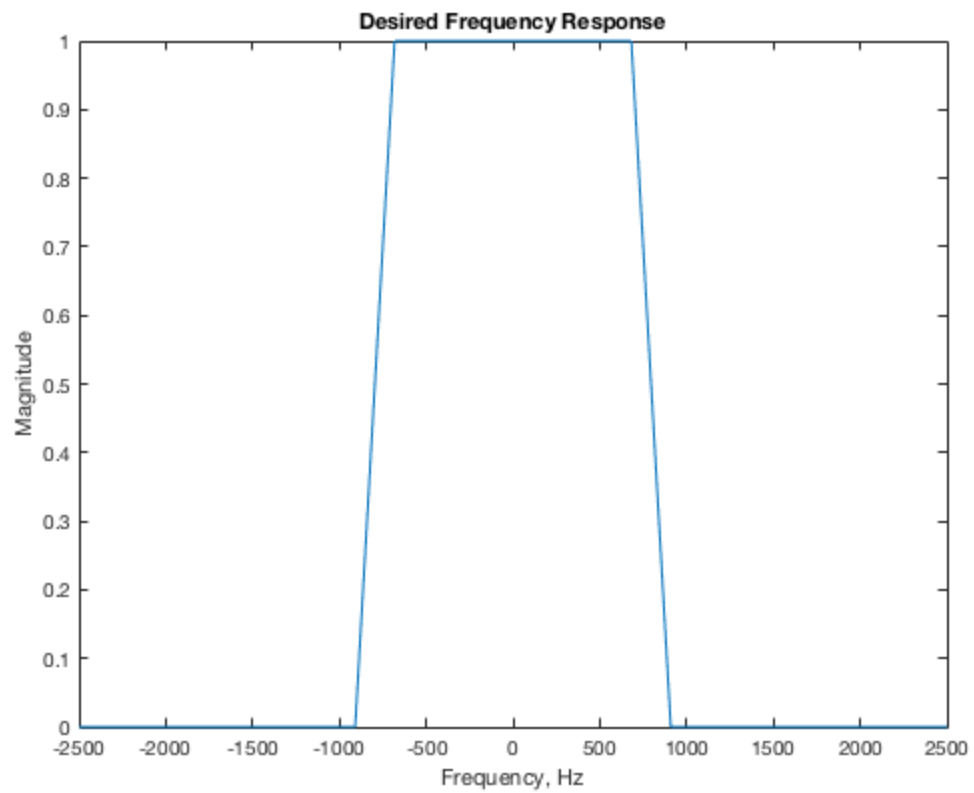
% Creating the magnitude of the filter frequency response
Hmag = double(abs(f)<fc);

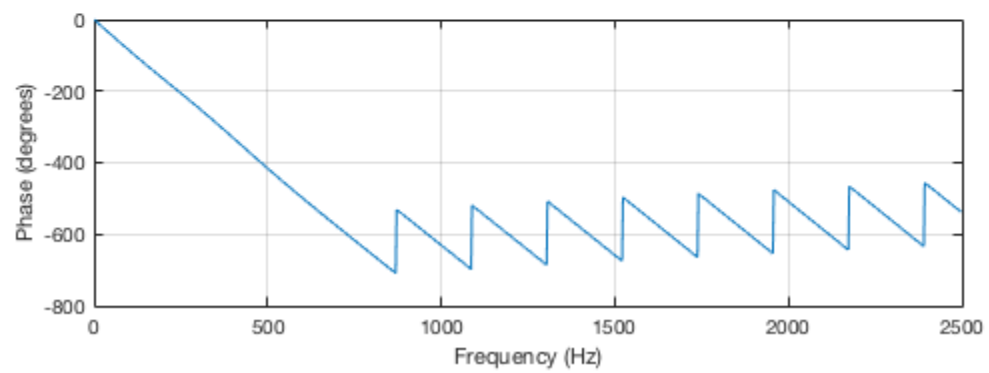
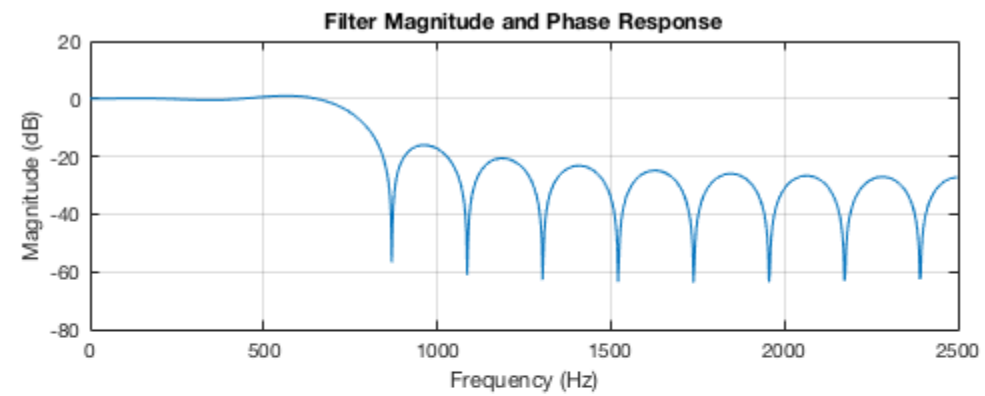
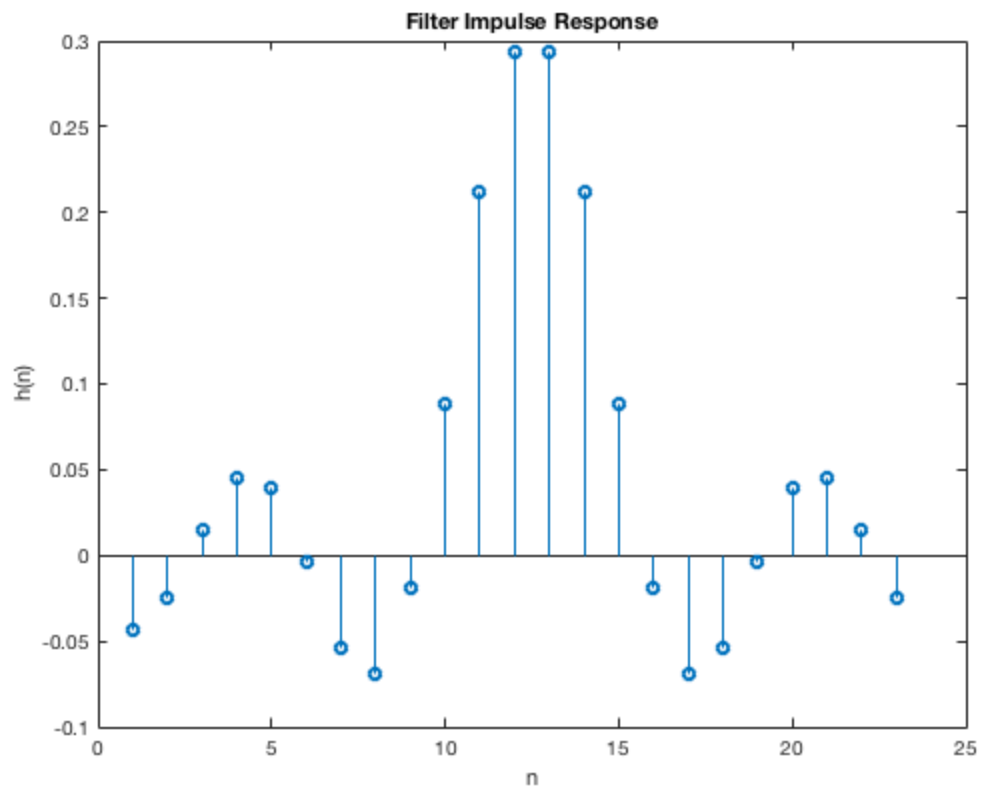
% Plotting the frequency response to check
figure
plot(fftshift(f),fftshift(Hmag));
title('Desired Frequency Response')
xlabel('Frequency, Hz')
ylabel('Magnitude');

% Creating the linear phase
w = f*2*pi/fs;
Hphase = exp(1i*w*M);
H = Hmag.*Hphase;
h = real(ifft(H));
figure
stem(1:L,h)
```

```
title('Filter Impulse Response')
xlabel('n')
ylabel('h(n)')

figure
freqz(h,1,1024,fs);
title('Filter Magnitude and Phase Response')
% Yes, this is the frequency and phase responses I was expecting.
% Since the
% magnitude response is plotted in decibels it makes sense that you
% couldnt
% have that represented in the response because that equates to -inf!!
```





Effect of the transition band

```
[Herr,ferr] = freqz(h,1,1024,fs);
pb = find(ferr<fc);
Hpb = abs(Herr(pb));
figure
plot(ferr(pb),Hpb,'b')
[maximum,idx] = max(Hpb);
hold on
f0 = find(f==0);
H0 = find(Hmag==0);
plot(f(f0:H0),Hmag(f0:H0),'r')
plot(ferr(idx),Hpb(idx),'vc')
hold off
title('Passband Error')
xlabel('Frequency, hz')
ylabel('Magnitude')
legend('Reconstructed Magnitude','Desired Magnitude',...
       'Maximum Reconstruction','Location','southwest')
txt = sprintf('  (%.4f,%.4f)',ferr(idx),Hpb(idx));
text(ferr(idx),Hpb(idx),txt)

pbErr = Hpb(idx)-1 % This error is the largest, 0.12, at ~564 Hz

sb = find(ferr>=fc);
Hsb = abs(Herr(sb));
fsb = ferr(sb);
figure
plot(fsb,Hsb,'b')
hold on
[peaks,pts] = findpeaks(Hsb);
[maximum,idx] = max(peaks);
idx = pts(idx);
txt = sprintf('  (%.4f,%.4f)',fsb(idx),Hsb(idx));
text(fsb(idx),Hsb(idx),txt)
plot(fsb(idx),Hsb(idx),'vc')
title('Stopband Error')
xlabel('Frequency, hz')
ylabel('Magnitude')
hold off

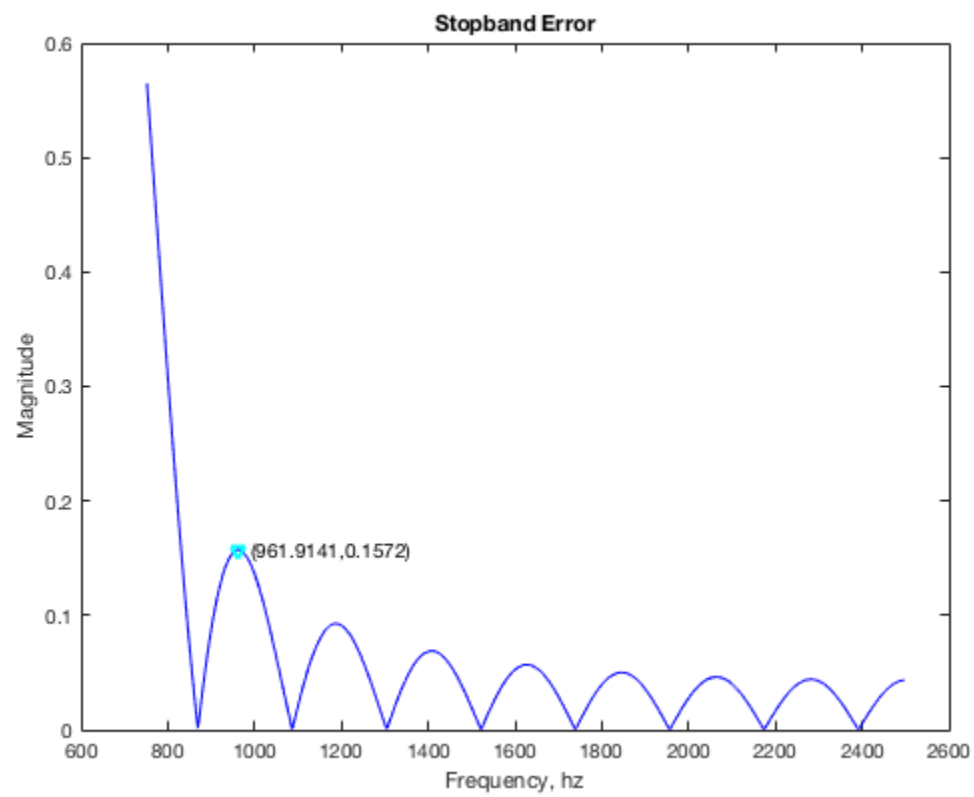
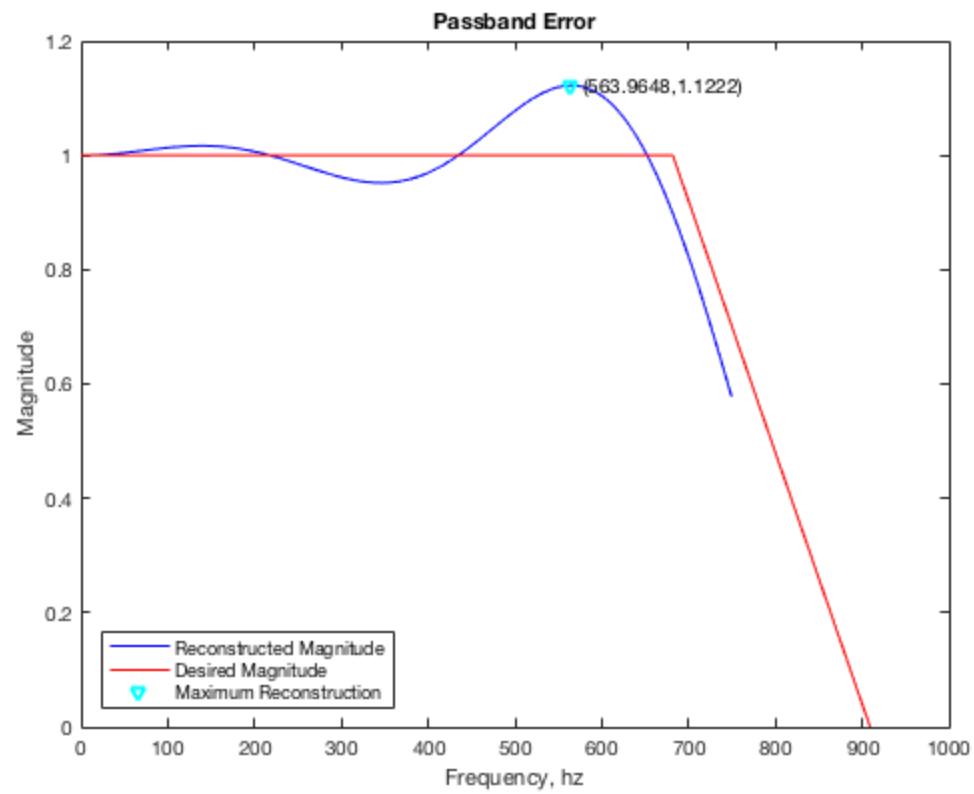
sbErr = Hsb(idx) % This maximum stopband error of 0.157 is at ~962 Hz

pbErr =

    0.1222

sbErr =

    0.1572
```



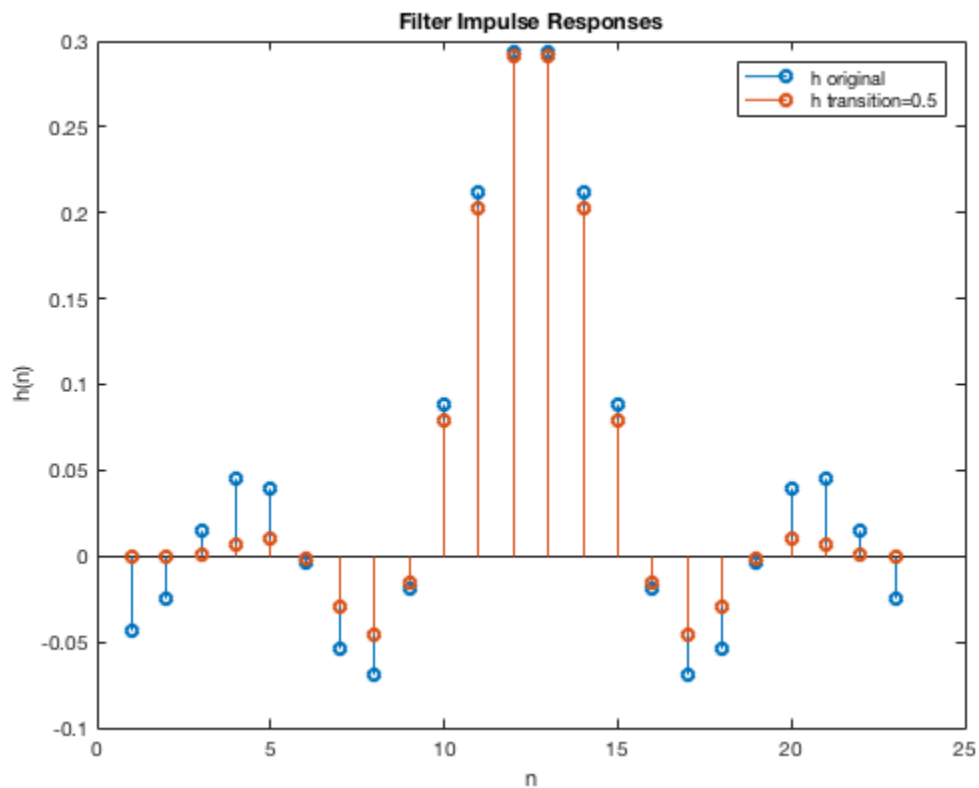
Adding the transition band

```
[minimum, trans] = min(abs(f-fc));
Hmag1 = Hmag;
Hmag1(trans) = 0.5; % Set trans value
Hmag1(L+1-trans) = 0.5; % Do it for the negative side too

% Using the same phase, invert the Fourier transform
H1 = Hmag1.*Hphase;
h1 = real(ifft(H1));

% Get an plot filter coefficients
figure
stem(1:L,h)
hold on
stem(1:L,h1)
title('Filter Impulse Responses')
legend('h original', 'h transition=0.5')
xlabel('n')
ylabel('h(n)')
hold off

% This looks like the original impulse response, however, it is a
% little
% more vertically squished, but it does still look sinc-like
```



Repeating previous steps w trans band

```
[Herr,ferr] = freqz(h1,1,1024,fs);
pb = find(ferr<fc);
Hpb = abs(Herr(pb));
figure
plot(ferr(pb),Hpb,'b')
[maximum,idx] = max(Hpb);
hold on
f0 = find(f==0);
H0 = find(Hmag==0);
plot(f(f0:H0),Hmag(f0:H0),'r')
plot(ferr(idx),Hpb(idx),'vc')
hold off
title('Passband Error')
xlabel('Frequency, hz')
ylabel('Magnitude')
legend('Reconstructed Magnitude','Desired Magnitude',...
       'Maximum Reconstruction','Location','southwest')
txt = sprintf(' (%.4f,%.4f)',ferr(idx),Hpb(idx));
text(ferr(idx),Hpb(idx),txt)
```

pbErrTrans = Hpb(idx)-1 % Max err of 0.0099 at ~361 Hz

```
sb = find(ferr>=fc);
Hsb = abs(Herr(sb));
fsb = ferr(sb);
figure
plot(fsb,Hsb,'b')
hold on
[peaks,pts] = findpeaks(Hsb);
[maximum,idx] = max(peaks);
idx = pts(idx);
txt = sprintf(' (%.4f,%.4f)',fsb(idx),Hsb(idx));
text(fsb(idx),Hsb(idx),txt)
plot(fsb(idx),Hsb(idx),'vc')
title('Stopband Error')
xlabel('Frequency, hz')
ylabel('Magnitude')
hold off
```

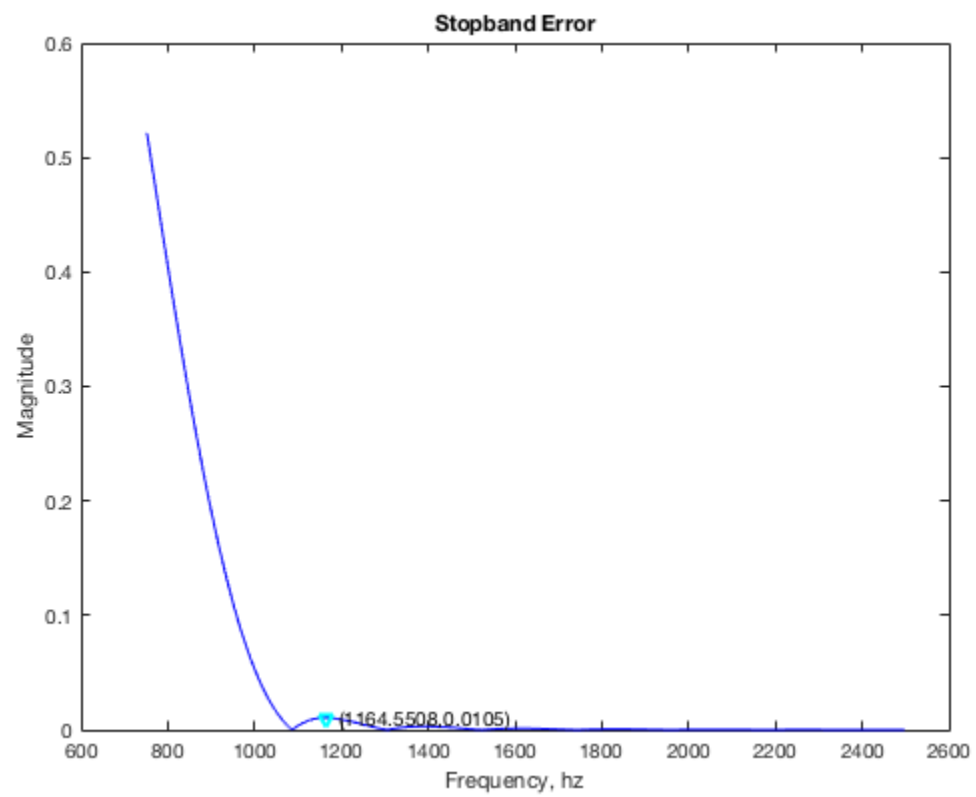
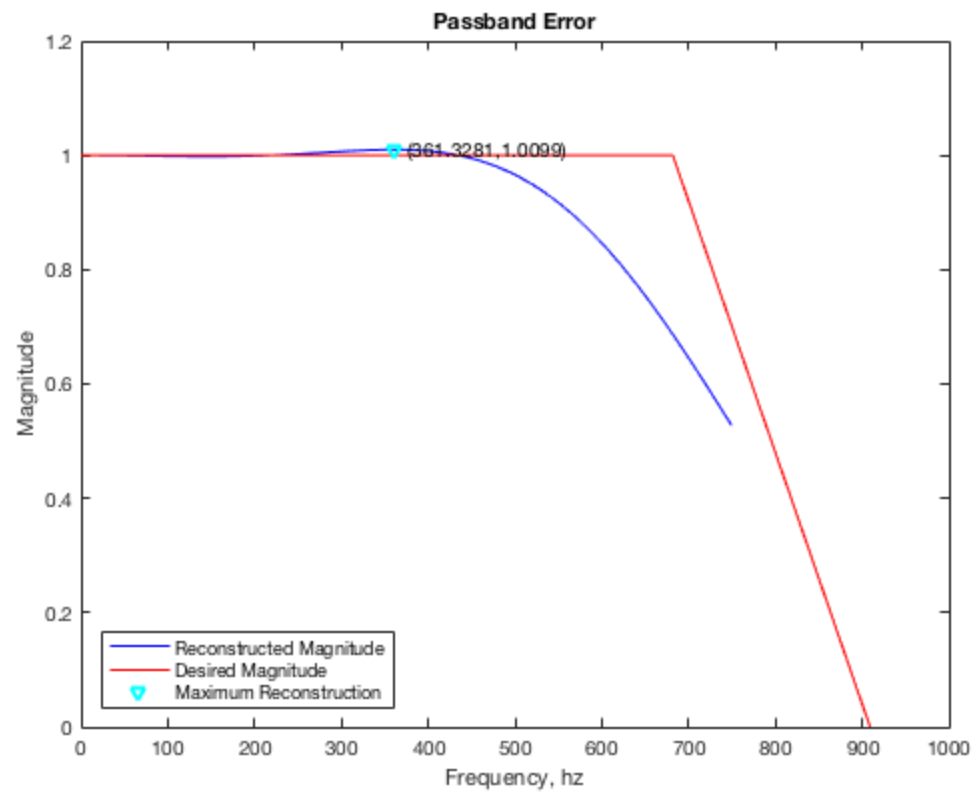
sbErrTrans = Hsb(idx) % Max err of 0.0105 at ~1165 Hz

pbErrTrans =

0.0099

sbErrTrans =

0.0105



Test and tabulate various transition value errors

```
pbErrTab = zeros(1,9);
sbErrTab = zeros(1,9);
for t = 0.1:0.1:0.9
    Hmag1(trans) = t;
    Hmag1(L+1-trans) = t;
    H1 = Hmag1.*Hphase;
    h1 = real(ifft(H1));
    HH = freqz(h1,1,1024,fs);
    pbErrTab(round(t*10)) = max(abs(HH(pb)))-1;
    sbErrTab(round(t*10)) = max(findpeaks(abs(HH(sb)))));
end
Passband_Error = pbErrTab';
Stopband_Error = sbErrTab';
Transition_Values =
    {'0.1';'0.2';'0.3';'0.4';'0.5';'0.6';'0.7';'0.8';'0.9'};
TGap1 = table(Passband_Error,
    Stopband_Error, 'RowNames',Transition_Values)

% From the table there is a clear tradeoff between the transition
% value and
% the pass/stop band error. As the transition value is shifted right
% the
% passband error decreases and the stopband error increases. So for
% good
% stopband performance choose a low transition value and for good
% passband
% performance choose a high transition value.
```

TGap1 =

9x2 table

| | <i>Passband_Error</i> | <i>Stopband_Error</i> |
|------------|-----------------------|-----------------------|
| | <hr/> | <hr/> |
| <i>0.1</i> | <i>0.025604</i> | <i>0.0089399</i> |
| <i>0.2</i> | <i>0.021676</i> | <i>0.0026562</i> |
| <i>0.3</i> | <i>0.01775</i> | <i>0.0021327</i> |
| <i>0.4</i> | <i>0.013829</i> | <i>0.0062425</i> |
| <i>0.5</i> | <i>0.0099102</i> | <i>0.010484</i> |
| <i>0.6</i> | <i>0.0060083</i> | <i>0.01475</i> |
| <i>0.7</i> | <i>0.0021896</i> | <i>0.019025</i> |
| <i>0.8</i> | <i>0.0009212</i> | <i>0.023303</i> |
| <i>0.9</i> | <i>0.0086479</i> | <i>0.027581</i> |

Repeat for a 2 sample transition band

```
Hmag2 = Hmag;
Hmag2(trans:trans+1) = [2 1]/3;
Hmag2(L-trans:L+1-trans) = [1 2]/3;
H2 = Hmag2.*Hphase;

% Using the same phase, invert the Fourier transform
h2 = real(ifft(H2));

% Get an plot filter coefficients
figure
stem(1:L,h)
hold on
stem(1:L,h1)
stem(1:L,h2)
title('Filter Impulse Responses')
legend('h original','h transition, 1 sample','h transition, 2 sample')
xlabel('n')
ylabel('h(n)')
hold off

% Repeating previous steps w trans band
[Herr,ferr] = freqz(h2,1,1024,fs);
pb = find(ferr<fc);
Hpb = abs(Herr(pb));
figure
plot(ferr(pb),Hpb,'b')
[maximum,idx] = max(Hpb);
hold on
f0 = find(f==0);
H0 = find(Hmag==0);
plot(f(f0:H0),Hmag(f0:H0),'r')
plot(ferr(idx),Hpb(idx),'vc')
hold off
title('Passband Error')
xlabel('Frequency, hz')
ylabel('Magnitude')
legend('Reconstructed Magnitude','Desired Magnitude',...
       'Maximum Reconstruction','Location','southwest')
txt = sprintf('  (%.4f,%.4f)',ferr(idx),Hpb(idx));
text(ferr(idx),Hpb(idx),txt)

pbErrTrans2 = Hpb(idx)-1

sb = find(ferr>=fc);
Hsb = abs(Herr(sb));
fsb = ferr(sb);
figure
plot(fsb,Hsb,'b')
hold on
[peaks,pts] = findpeaks(Hsb);
[maximum,idx] = max(peaks);
```

```
idx = pts(idx);
txt = sprintf('  (%.4f,%.4f)',fsb(idx),Hsb(idx));
text(fsb(idx),Hsb(idx),txt)
plot(fsb(idx),Hsb(idx),'vc')
title('Stopband Error')
xlabel('Frequency, hz')
ylabel('Magnitude')
hold off

sbErrTrans2 = Hsb(idx)

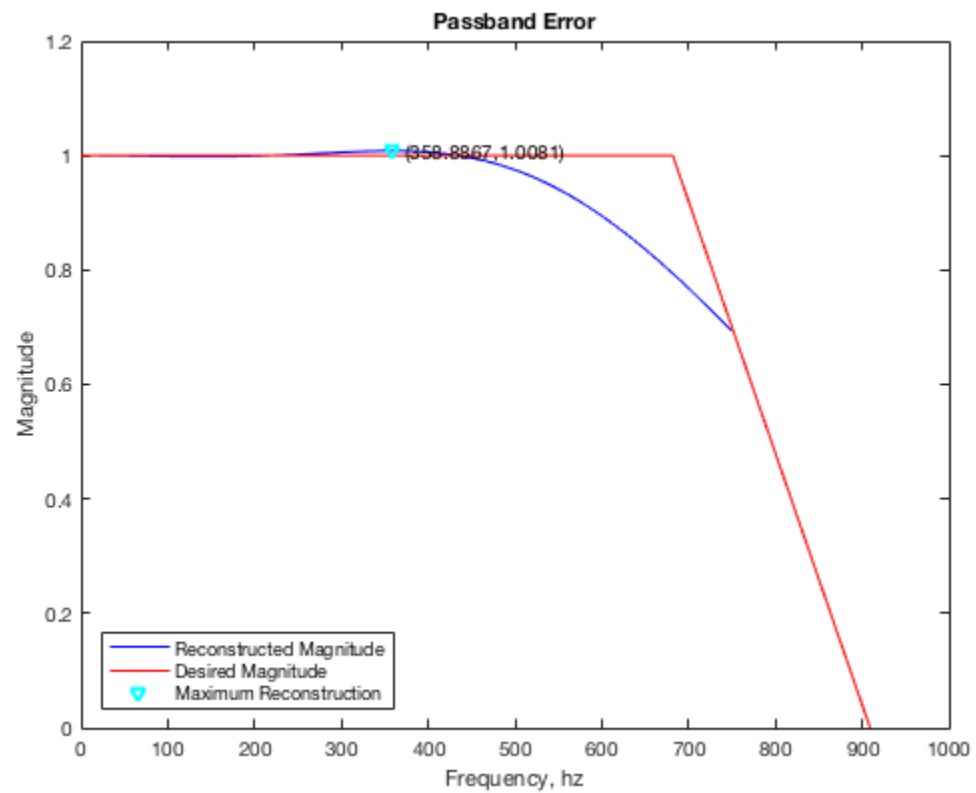
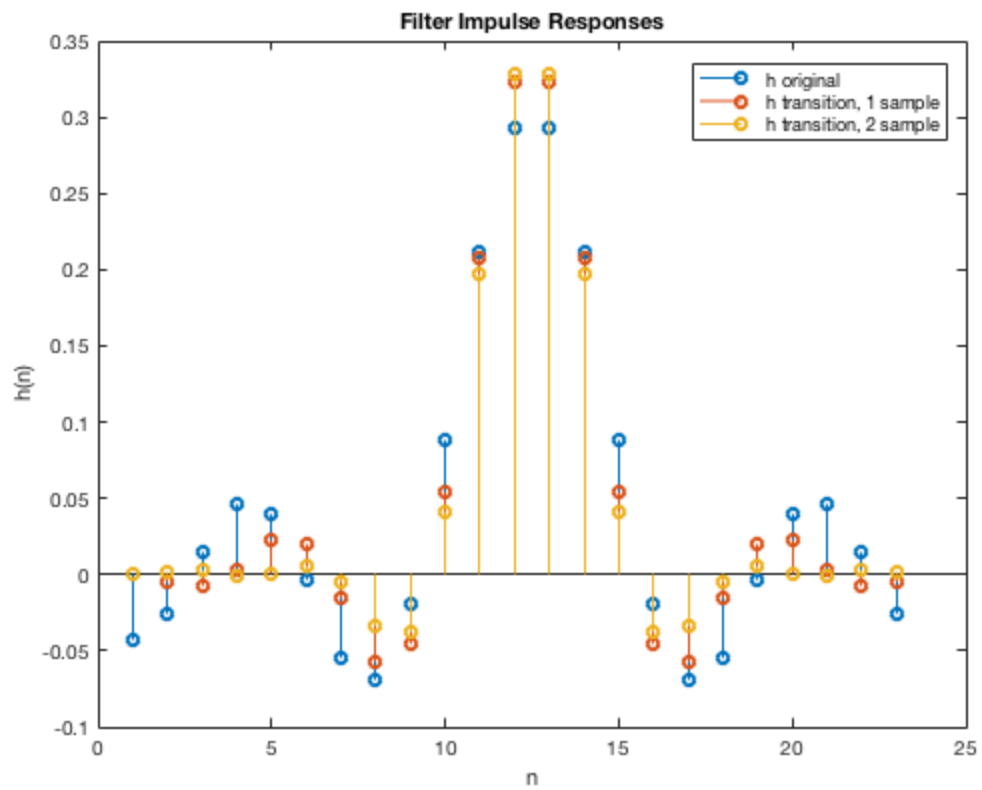
% Notice that using the linear phase 2 sample transition band causes
% the
% measured error in both the stop and pass band to be smaller than any
% of
% the 1 sample transition band errors from the table. (compare table
% with
% pbErrTrans2 and sbErrTrans2 for the passband and stopband errors
% with 2
% samples)

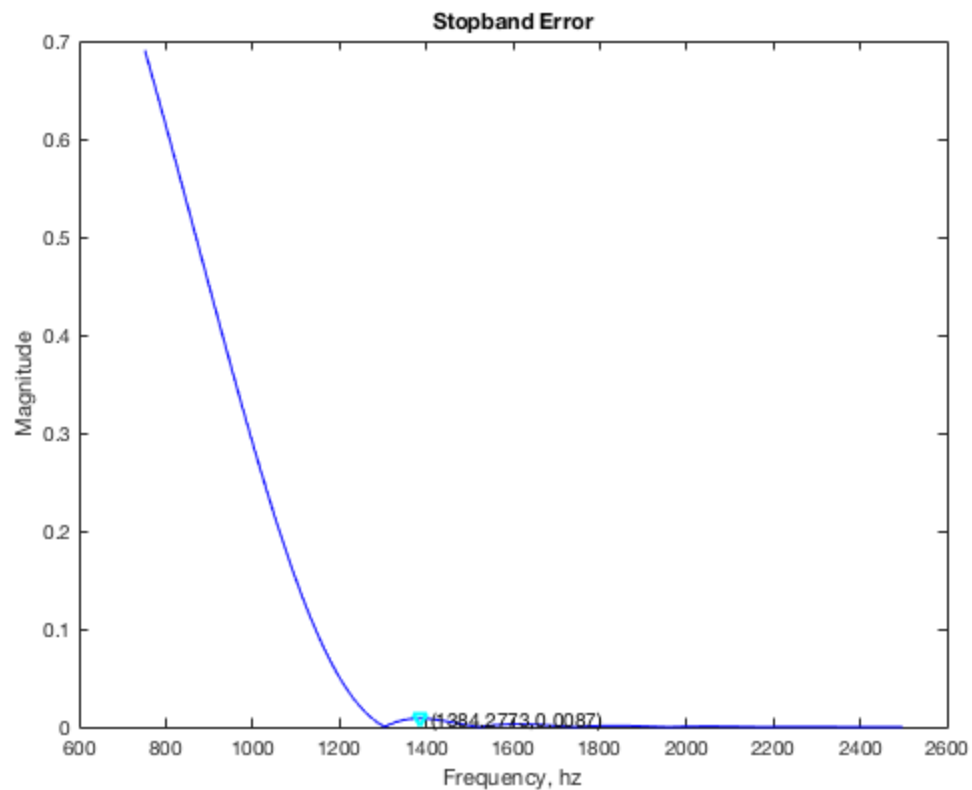
pbErrTrans2 =

    0.0081

sbErrTrans2 =

    0.0087
```





Parks Mclellan

```
% From firpm doc
rp = 10*log10(1.02); % Passband ripple
rs = 40; % Stopband ripple
fs = 1000; % Sampling frequency
f = [100 175]; % Cutoff frequencies
a = [1 0]; % Desired amplitudes

dev = [(10^(rp/20)-1)/(10^(rp/20)+1) 10^(-rs/20)];
[n,fo,ao,w] = firpmord(f,a,dev,fs);
b = firpm(n,fo,ao,w);
freqz(b,1,1024,fs)
title('Parks Mclellan Filter')

[Herr,ferr] = freqz(b,1,1024,fs);
pb = find(ferr<f(1));
Hpb = abs(Herr(pb));
figure
plot(ferr(pb),Hpb,'b')
[maximum,idx] = max(Hpb);
hold on
plot(ferr(idx),Hpb(idx),'vc')
hold off
title('Passband Error')
```

```

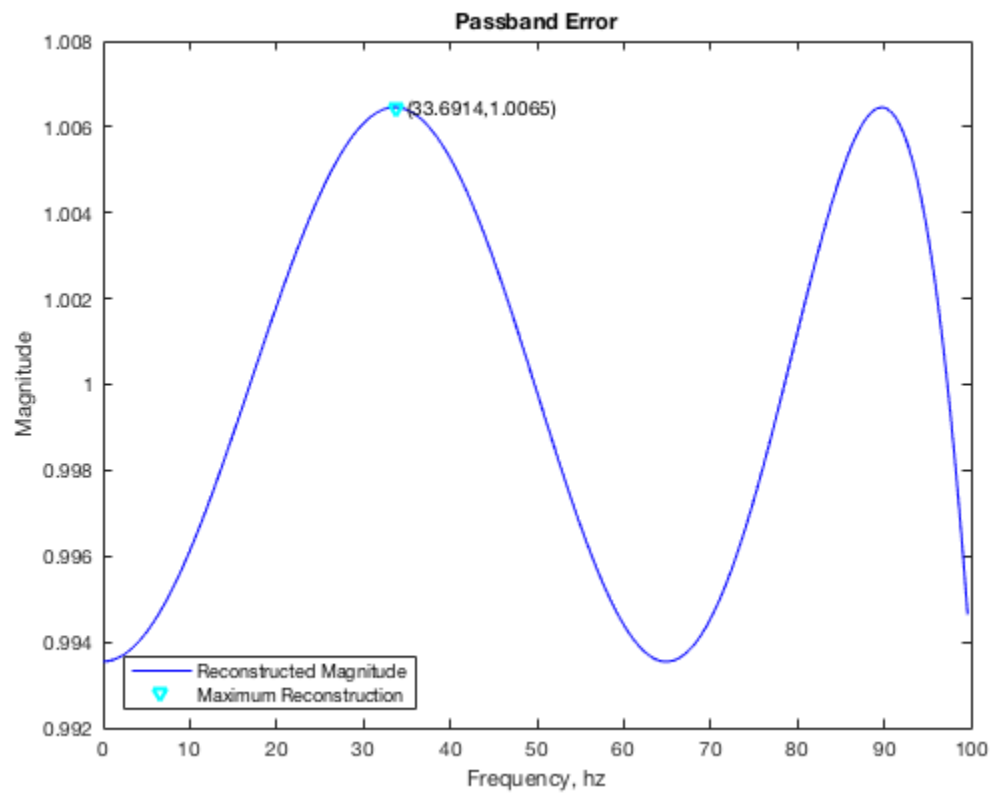
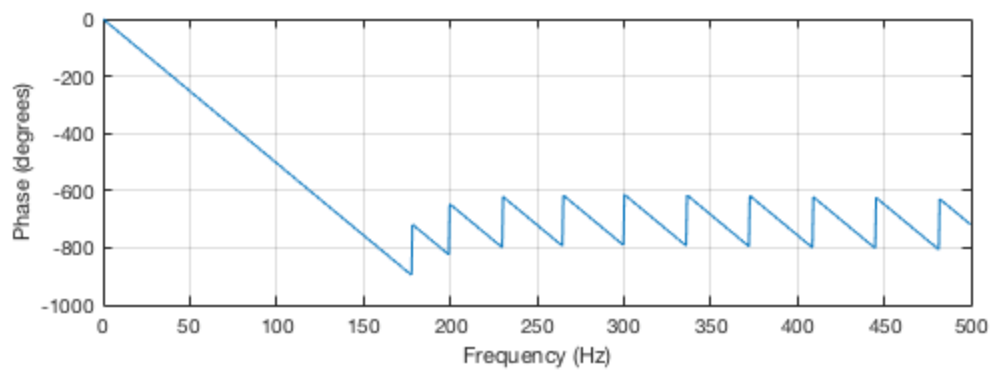
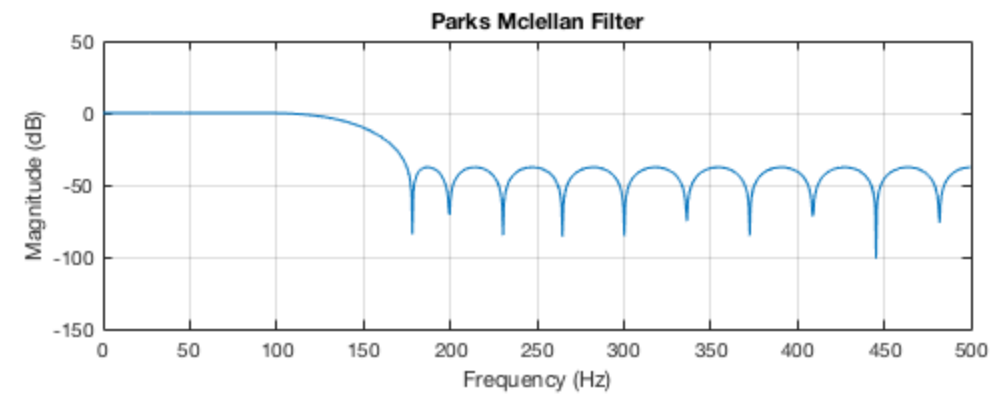
xlabel('Frequency, hz')
ylabel('Magnitude')
legend('Reconstructed Magnitude',...
       'Maximum Reconstruction','Location','southwest')
txt = sprintf('  (%.4f,%.4f)',ferr(idx),Hpb(idx));
text(ferr(idx),Hpb(idx),txt)

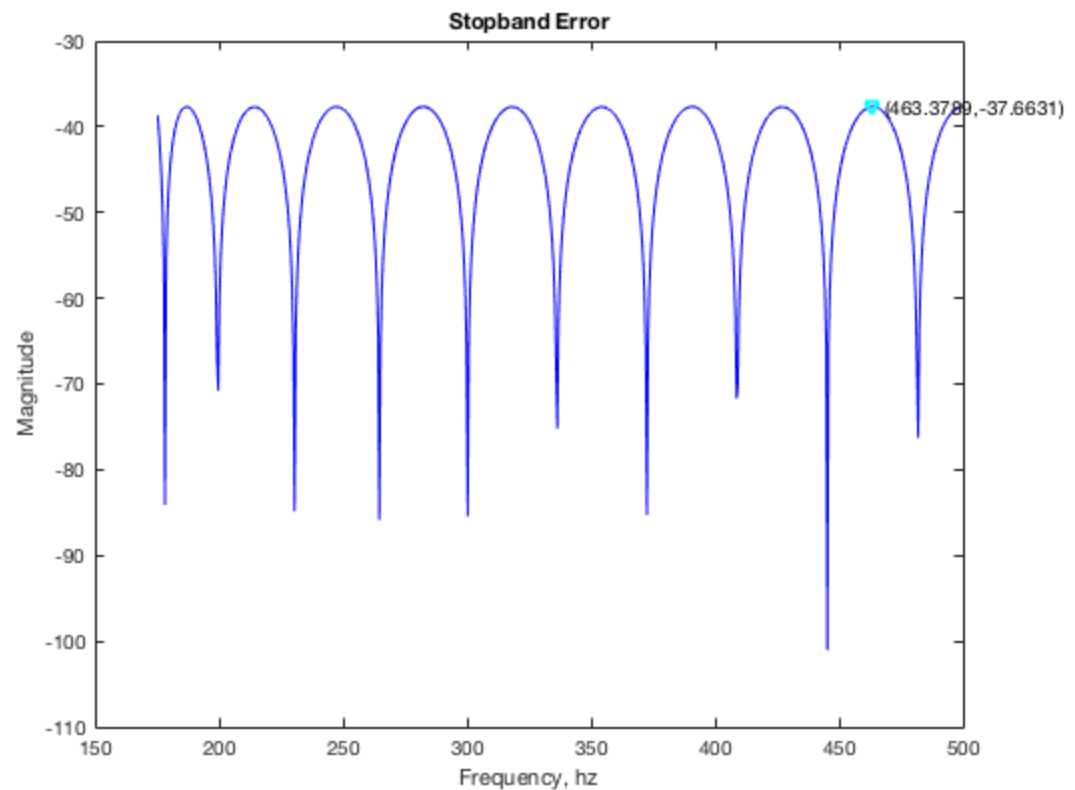
pbErrPM = Hpb(idx) - 1;

sb = find(ferr>f(2));
Hsb = mag2db(abs(Herr(sb)));
fsb = ferr(sb);
figure
plot(fsb,Hsb,'b')
hold on
[peaks,pts] = findpeaks(Hsb);
[maximum,idx] = max(peaks);
idx = pts(idx);
txt = sprintf('  (%.4f,%.4f)',fsb(idx),Hsb(idx));
text(fsb(idx),Hsb(idx),txt)
plot(fsb(idx),Hsb(idx),'vc')
title('Stopband Error')
xlabel('Frequency, hz')
ylabel('Magnitude')
hold off

sbErPM = Hsb(idx) + rs;

% No adjustment necessary
```



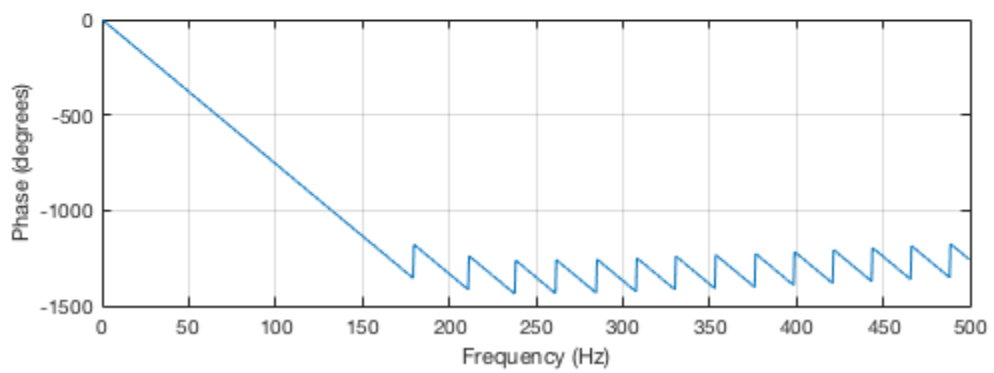
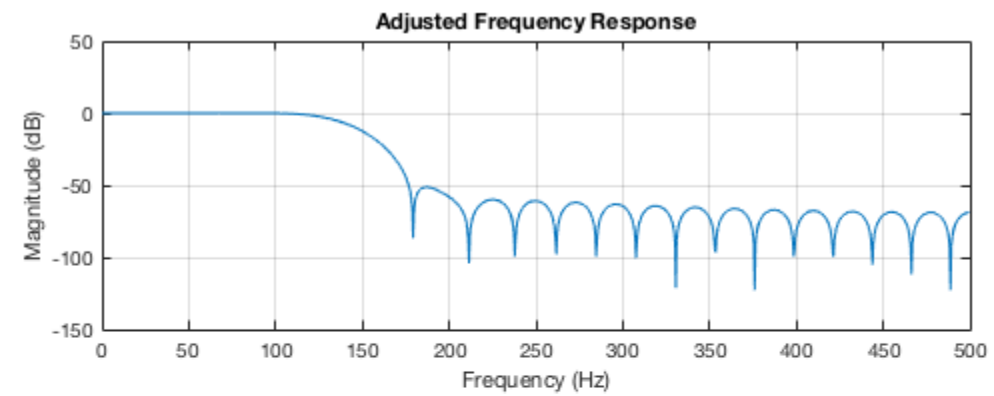
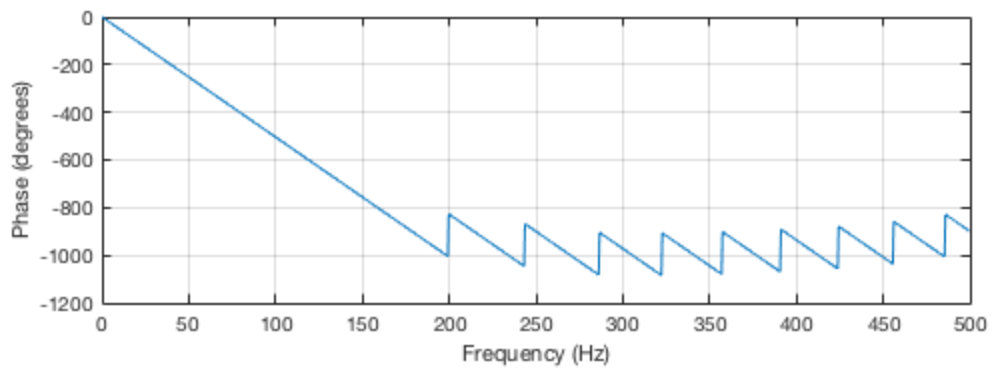
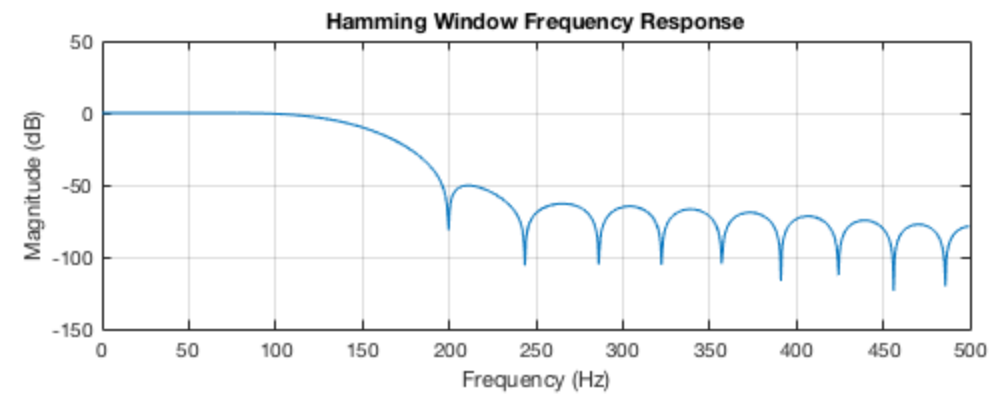


Filter Design By Windowing

1. Hamming window cutoff $\rightarrow 2f_c/f_s$ - middle of trans band

```
hamm = fir1(n, (f(1)+f(2))/fs, 'low');  
figure  
freqz(hamm, 1, 1024, fs)  
title('Hamming Window Frequency Response')  
% Doesn't meet specs
```

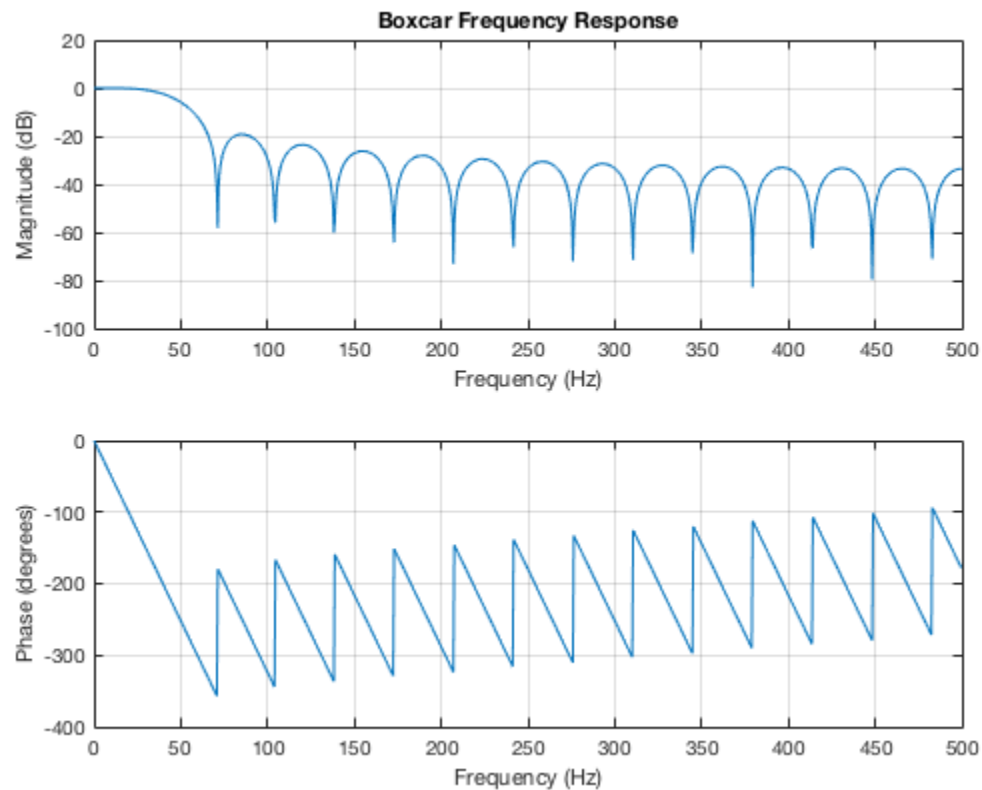
```
hamm = fir1(1.5*n, (f(1)+f(2))/fs, 'low');  
figure  
freqz(hamm, 1, 1024, fs)  
title('Adjusted Frequency Response')  
% Looks like it meets the specs
```

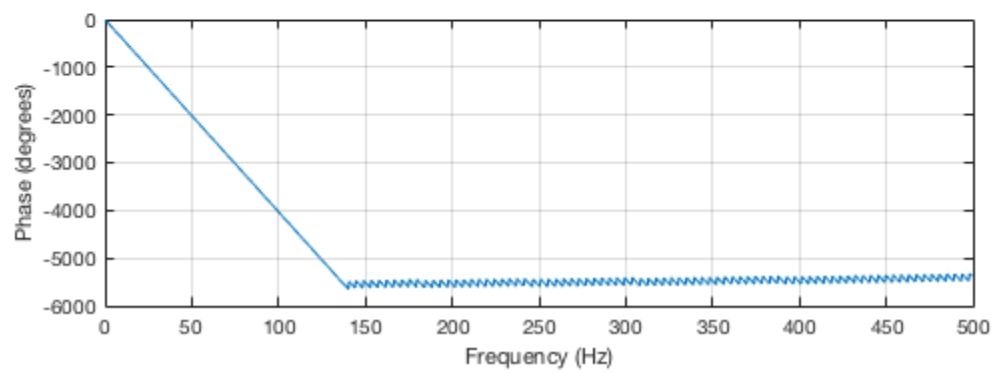
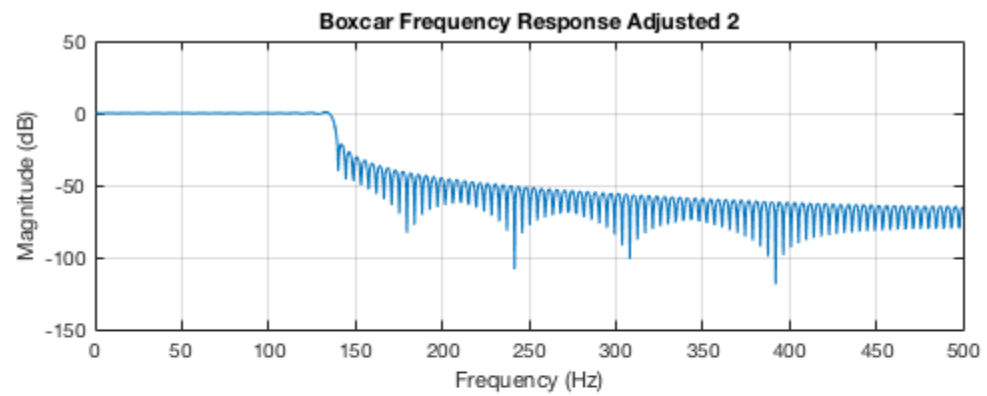
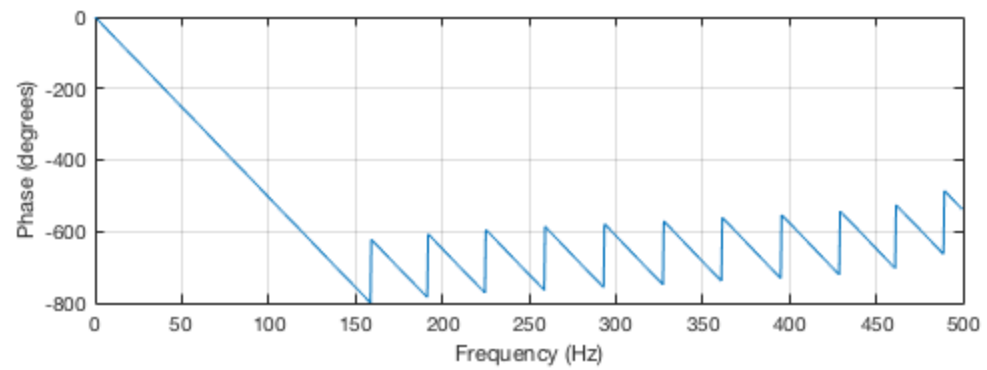
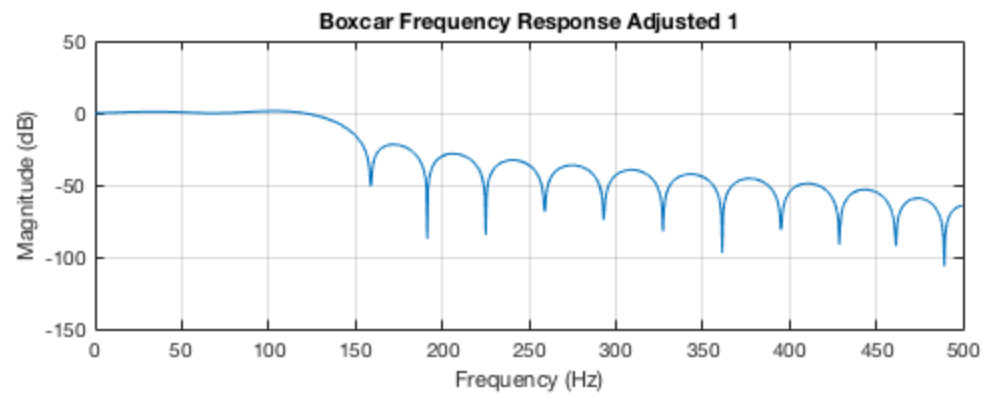



2 Boxcar window

Try making the boxcar with the start at the low cutoff freq

```
bc = fir1(n,f(1)/fs,rectwin(n+1));
figure
freqz(bc,1,1024,fs)
title('Boxcar Frequency Response')
% Doesn't meet specs - cutoff too quick
% Try using the transition point
bc = fir1(n,(f(1)+f(2))/fs,rectwin(n+1));
figure
freqz(bc,1,1024,fs)
title('Boxcar Frequency Response Adjusted 1')
% This is a better frequency to change the window length from
% Improving the response by changing the window length to 8x
bc = fir1(8*n,(f(1)+f(2))/fs,rectwin(8*n+1));
figure
freqz(bc,1,1024,fs)
title('Boxcar Frequency Response Adjusted 2')
% Yay this meets the specs :)
```





3

FFT has complexity of $n \log n$

```
% Calculating complexity for equiripple
EC = n*log(n) % = 93.3017
HC = 2*n*log(2*n) % = 225.4197 - ~2.4x slower
BC = 8*n*log(8*n) % = 1212.2 - ~13x slower

% Calculate percent savings
HE = (HC-EC)/HC * 100 % ~58% Savings
BE = (BC-EC)/BC * 100 % ~92% Savings wow thats a lot of savings

% Shoutout to Richard Preston for guidance in some places
```

EC =

93.3017

HC =

225.4197

BC =

1.2122e+03

HE =

58.6098

BE =

92.3032

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