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## Exercise 7.1

```
%a)

% i)  $h[n] = \delta[n] - 0.8\delta[n-1]$ 
%  $H[z] = 1 - 0.8z^{-1}$ 
%ROC:  $C - \{ \# \}$ 
%Poles:0
%zeros:0.8
% stable
figure;
zplane(0.8,0);
title('i pole-zero plot');

%The filter matches the analytic equation, the pole corresponds to the
%decay towards zero

impulse = [ zeros(1,25) 1 zeros(1,25)];
a = [1 -0.8];
b = [1];
y = filter(b,a,impulse);

figure;
stem(y);
title('i filter plot')

% ii)  $h[n] = \delta[n] + 0.8\delta[n-1]$ 
%  $H[z] = 1 + 0.8z^{-1}$ 
%ROC:  $C - \{ \# \}$ 
%zeros:-0.8
%poles:0
%stable
figure;
zplane(-0.8,0);
title('ii pole-zero plot');

a = [1 +0.8];
b = [1];
y = filter(b,a,impulse);

figure;
stem(y);
title('ii filter plot')

%The filter matches the analytic equation, the pole corresponds to the
%decay and growth towards zero

% iii)  $h[n] = \delta[n + 10] - 0.8\delta[n + 9]$ 
% $H[z] = z^{10} - 0.8z^9$ 
%ROC:  $C - \{0\}$ 
%zeros:0.8, 0 mult. 9
%poles: none
```

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%stable
figure;
zplane([-0.8;0;0;0;0;0;0;0;0;0]);
title('iii pole-zero plot');

a = [1 +0.8];
b = [0 0 0 0 0 0 0 0 0 0 1];
y = filter(b,a,impulse);

figure;
stem(y);
title('iii filter plot')

%The filter matches the analytic equation, the pole corresponds to the
%decay and growth towards zero

% iv)  $h[n] = (1/2)\delta[n] - \delta[n-1]$ 
%H[z] =  $1/2 - z^{-1}$ 
%ROC:  $C - \{#\}$ 
%zeros: 2
%poles: 0
%stable
figure;
zplane(2,0);
title('iv pole-zero plot');

a = [1 -2];
b = [2];
y = filter(b,a,impulse);

figure;
stem(y);
title('iv filter plot')

%The filter matches the analytic equation, the pole corresponds to the
%growth

%b)

%i)  $h[n] = \delta[n] - \cos(0.2\pi)\delta[n-1] - 0.25\delta[n-2]$ 
%freq resp:  $H(e^{j\omega}) = 1 - \cos(0.2\pi)e^{-j\omega} - 0.25e^{-2j\omega}$ 
%  $H[z] = 1 - \cos(0.2\pi)z^{-1} - 0.25z^{-2}$ 
%ROC:  $C - \{#\}$ 
%stable
zer = [-0.23863; 1.04765];
poles = [0;0];

figure;
zplane(zer,poles);
title('ib pole-zero plot');

a = [1 -cos(0.2*pi) -0.25];

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b = [1];
y = filter(b,a,impulse);

figure;
subplot(2,1,1)
stem(y);
title('ib filter plot')

[b,a] = zp2tf(zer,poles,1);
fvtool(b,a);
title ( 'Magnitude Response ib')

% The poles at zero correspond with the magnitude response

%ii)  $h[n] = \delta[n] - 1.8\cos(0.2\pi)\delta[n-1] - 0.81\delta[n-2]$ 
%  $H[z] = 1 - 1.8\cos(0.2\pi)z^{-1} - 0.81z^{-2}$ 
%freq resp:  $H(e^{j\omega}) = 1 - 1.8\cos(0.2\pi)e^{-j\omega} - 0.81e^{-2j\omega}$ 
%ROC:  $C = \{ \# \}$ 
%stable
zer = [-0.429534; 1.88576];
poles = [0;0];

figure;
zplane(zer,poles);
title('iib pole-zero plot');

a = [1 -1.8*cos(0.2*pi) -0.81];
b = [1];
y = filter(b,a,impulse);

figure;
subplot(2,1,1)
stem(y);
title('iib filter plot')

[b,a] = zp2tf(zer,poles,1);
fvtool(b,a);
title ( 'Magnitude Response iib')

% The poles at zero correspond with the magnitude response

%iii)  $h[n] = \delta[n] - 1.98\cos(0.2\pi)\delta[n-1] - 0.9801\delta[n-2]$ 
%  $H[z] = 1 - 1.98\cos(0.2\pi)z^{-1} - 0.9801z^{-2}$ 
%freq resp:  $H(e^{j\omega}) = 1 - 1.98\cos(0.2\pi)e^{-j\omega} - 0.9801e^{-2j\omega}$ 
%ROC:  $C = \{ \# \}$ 
%stable
zer = [-0.42487; 2.07434];
poles = [0;0];

figure;
zplane(zer,poles);

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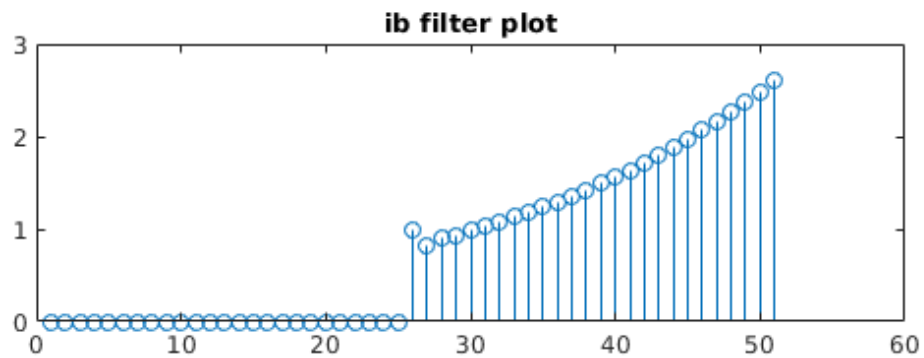
```
title('iiib pole-zero plot');

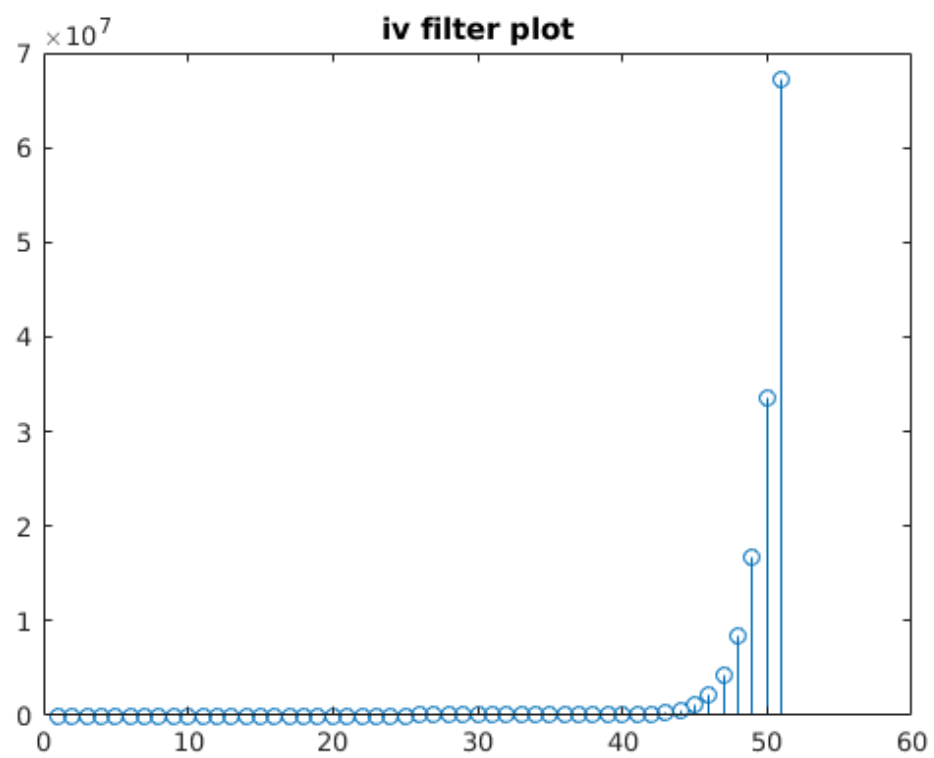
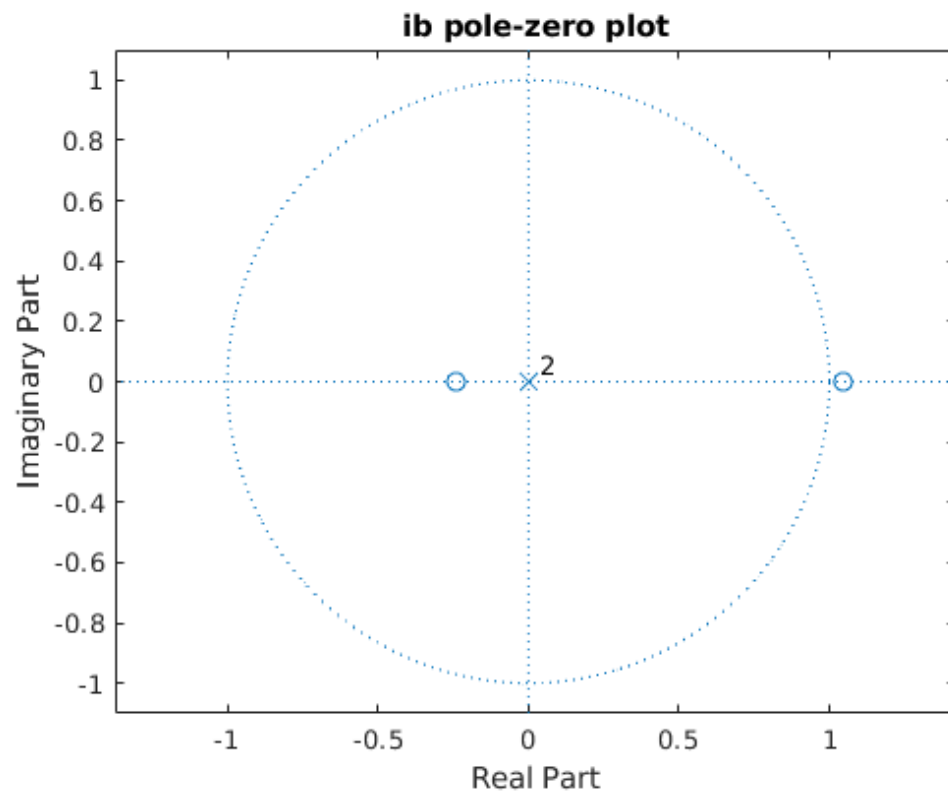
a = [1 -1.98*cos(0.2*pi) -0.9801];
b = [1];
y = filter(b,a,impulse);

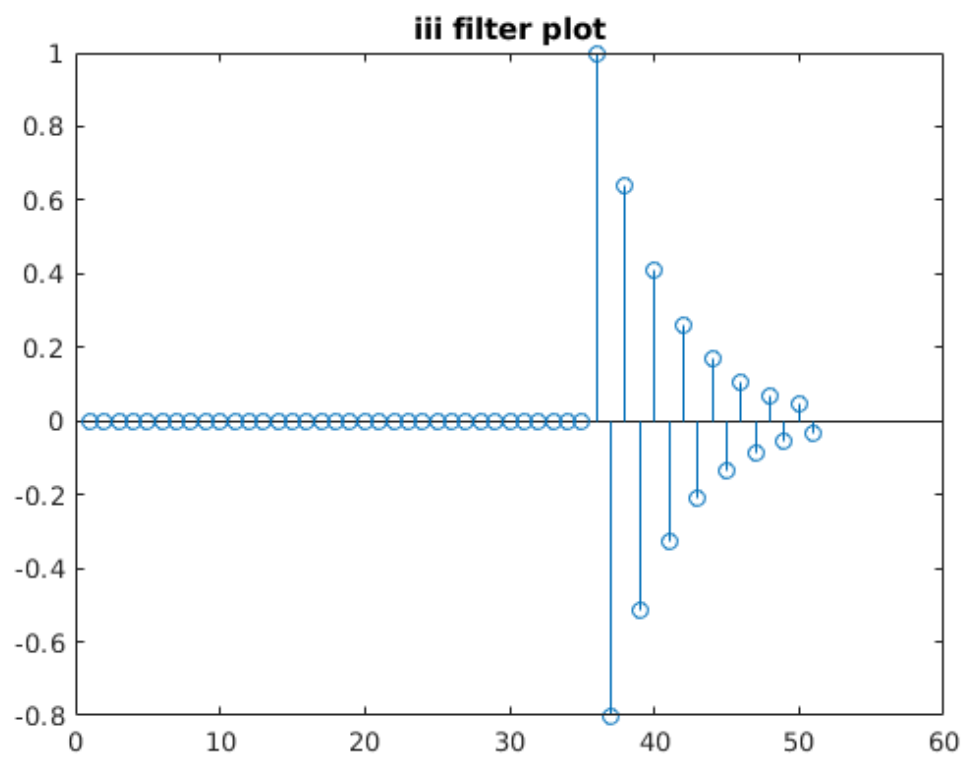
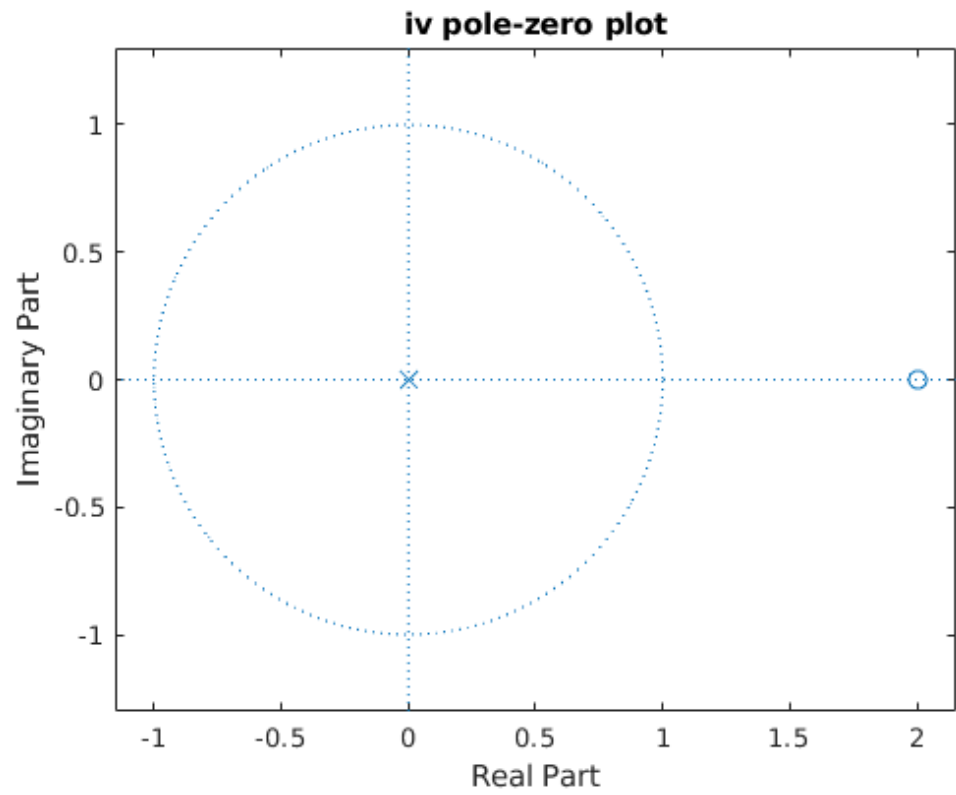
figure;
subplot(2,1,1)
stem(y);
title('iiib filter plot')

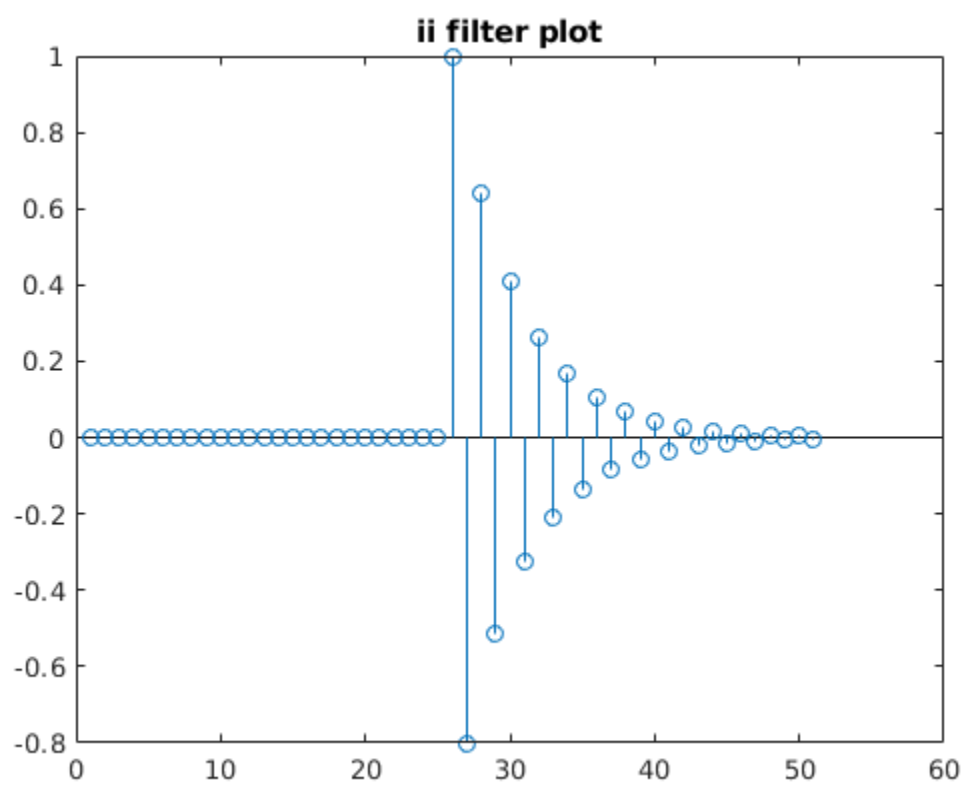
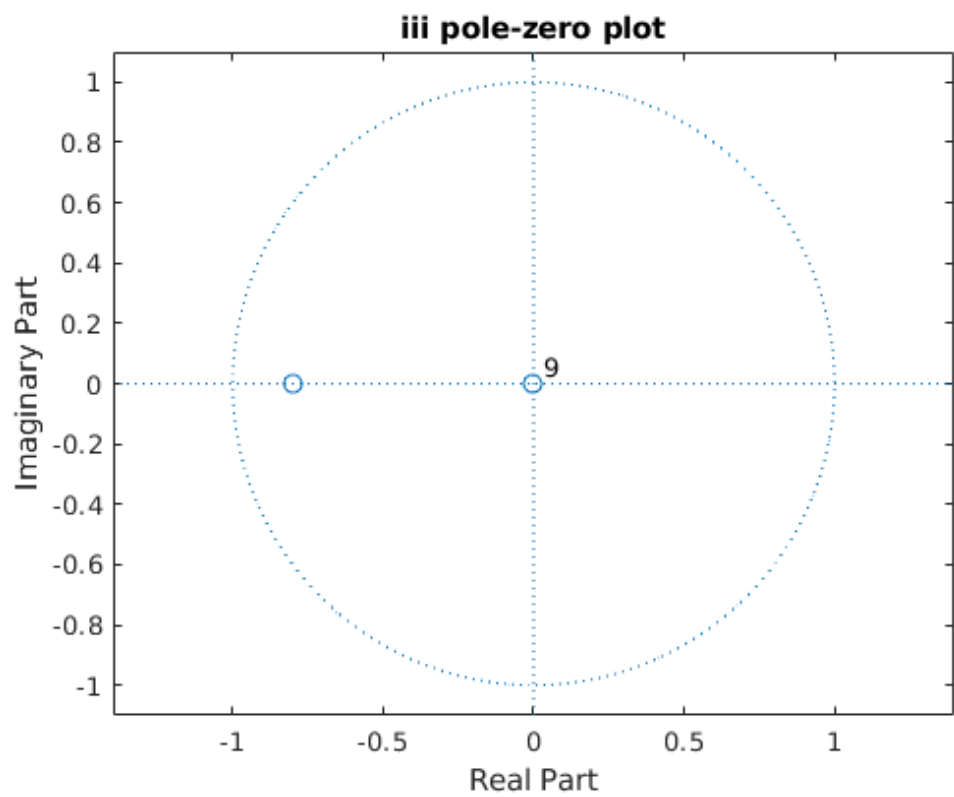
[b,a] = zp2tf(zer,poles,1);
fvtool(b,a);
title ( 'Magnitude Response iiib')

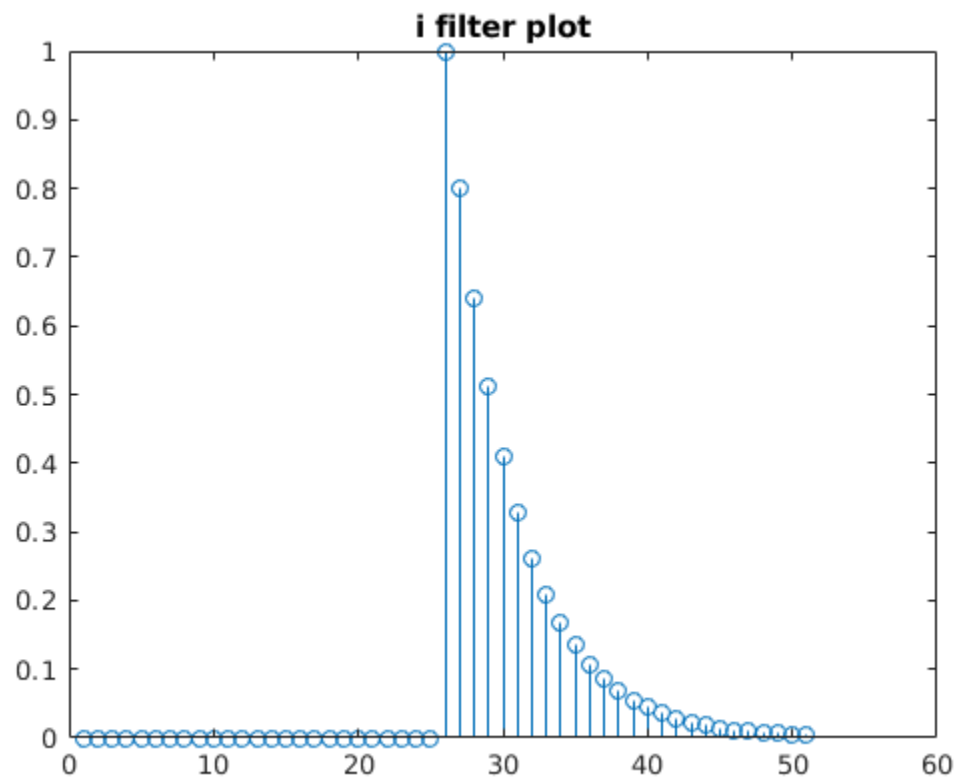
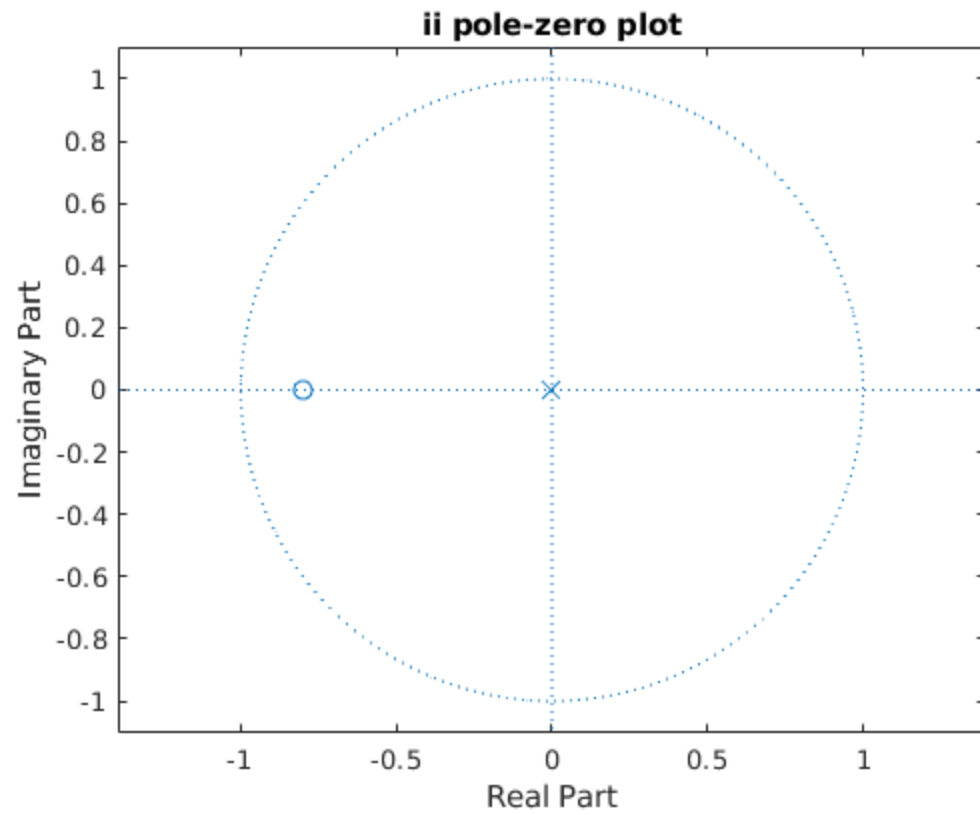
% The poles at zero correspond with the magnitude response
```



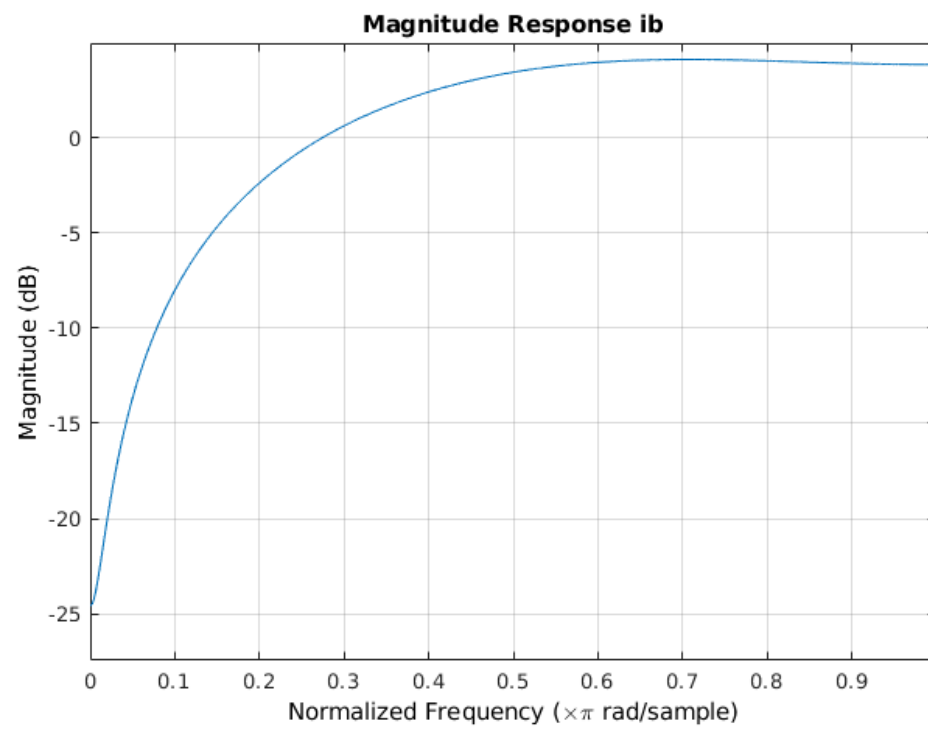
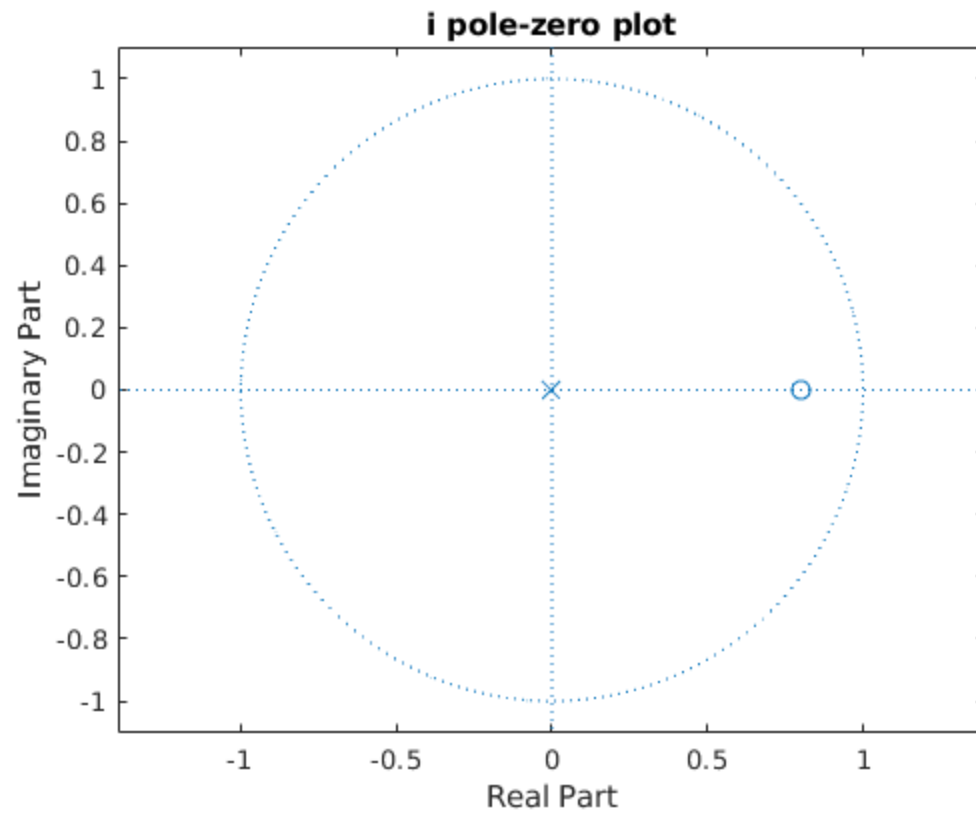


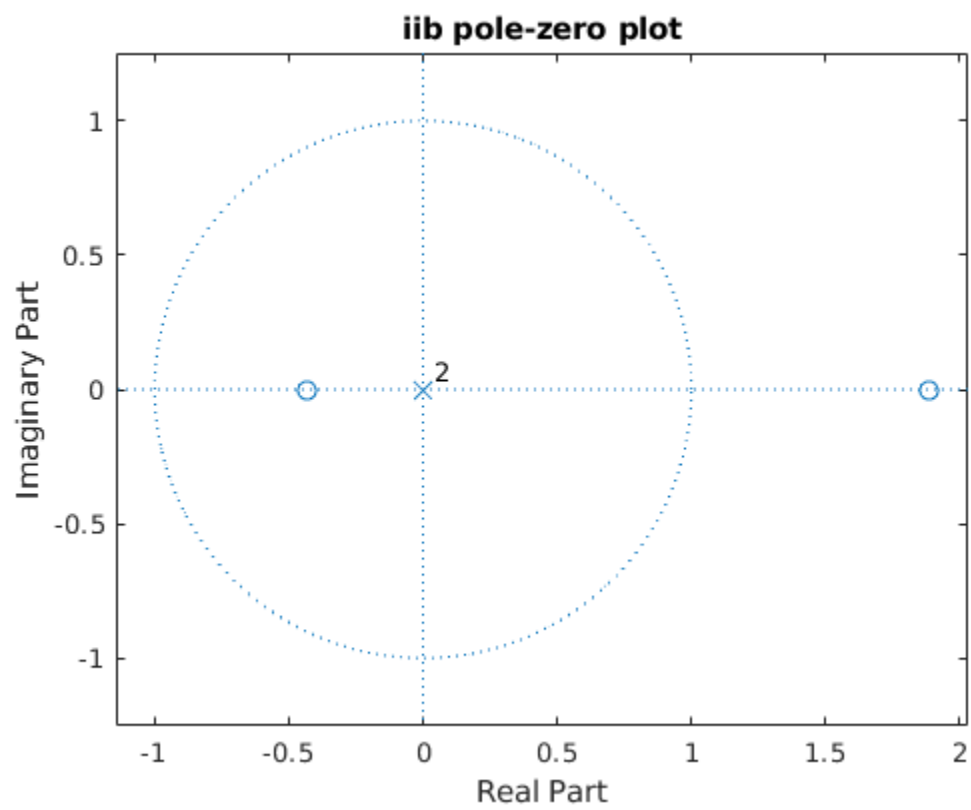
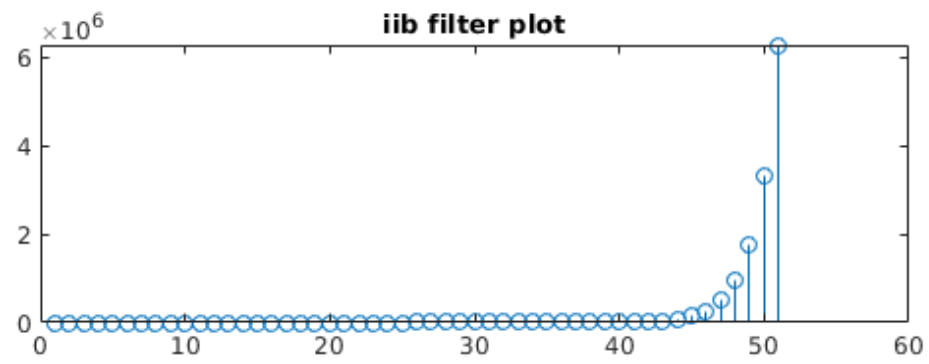


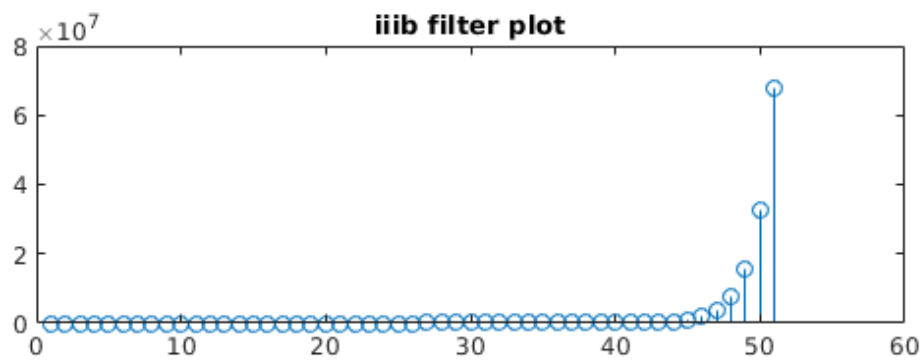
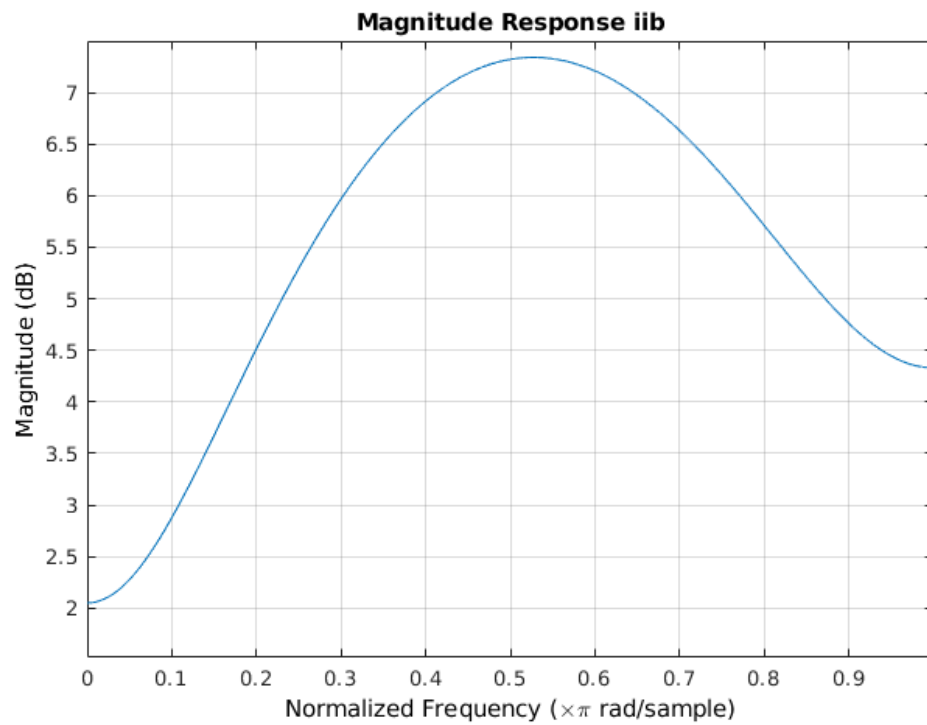


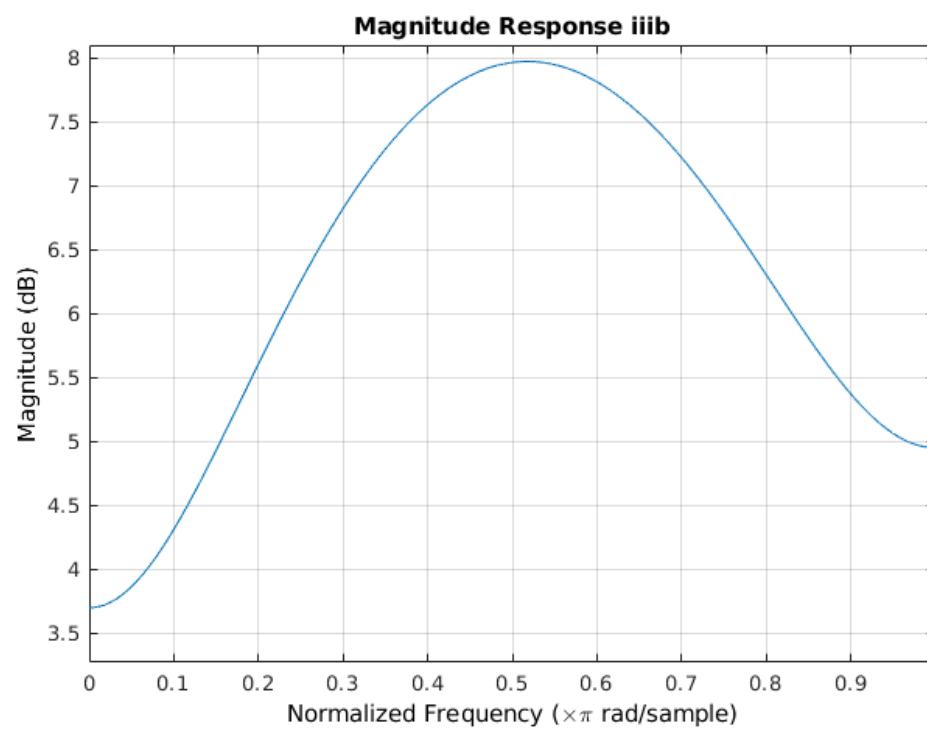
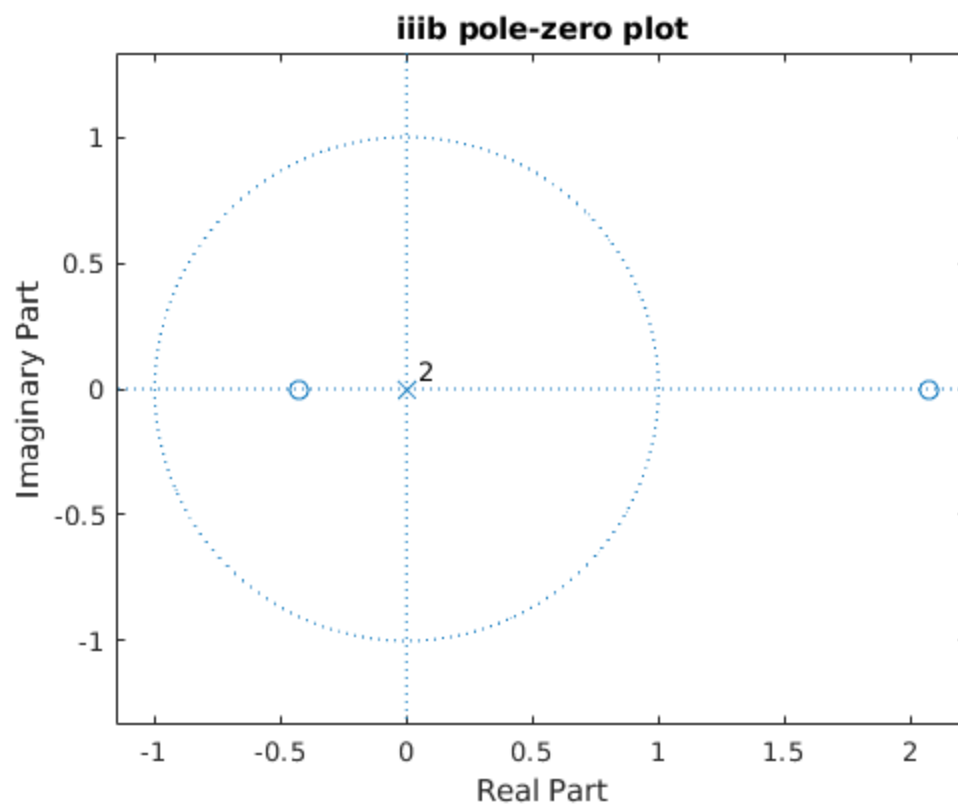












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## Exercise 7.2

```
%a

[corrupt,Fs] = audioread('bad_wannabe.wav');

w = -pi:pi/2000:pi;
corrupt_dtft = dtft(corrupt,w);

figure;
subplot (2 , 1 , 1)
plot (w/pi,20*log10(abs(corrupt_dtft))) ;
grid on ;
title ( 'Magnitude Response Corrupt')
xlabel ( 'Normalized Radian Frequency (\times \pi rad/sample ) ' ) ;
ylabel ( ' Amplitude (dB) ' ) ;

h = [1 -2*cos(0.05*pi) 1];
not_corrupt = conv(corrupt,h);
audiowrite('fired_wannabe.wav',not_corrupt,Fs)
not_corrupt = dtft(not_corrupt,w);

figure;
subplot (2 , 1 , 1)
plot (w/pi,20*log10(abs(not_corrupt))) ;
grid on ;
title ( 'Magnitude Response')
xlabel ( 'Normalized Radian Frequency (\times \pi rad/sample ) ' ) ;
ylabel ( ' Amplitude (dB) ' ) ;
subplot (2 , 1 , 2)
plot ( w / pi , angle ( not_corrupt) / pi ) ;
grid on ;
title ( ' Phase Response')
xlabel ( 'Normalized Radian Frequency (\times \pi rad/sample) ');
ylabel('Phase(/times /pi rad)');

%The filtered audio quality dropped dramatically.

%b)
alp = 1;
zer = [exp(j*0.05);exp(-j*0.05)];
poles = [alp*exp(j*0.05);alp*exp(-j*0.05)];

% The filter is stable with -1 <= alpha <= 1

figure;
subplot(2,1,1);
zplane(zer,poles);
title('stable')
```

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```

alp = -2;
zer = [exp(j*0.05);exp(-j*0.05)];
poles = [alp*exp(j*0.05);alp*exp(-j*0.05)];

subplot(2,1,2)
zplane(zer,poles);
title('unstable')

alp = 0;
zer = [exp(j*0.05);exp(-j*0.05)];
poles = [alp*exp(j*0.05);alp*exp(-j*0.05)];

[b,a] = zp2tf(zer,poles,1);
fvtool(b,a);
title('alpha = 0');

alp = 0.25;
zer = [exp(j*0.05);exp(-j*0.05)];
poles = [alp*exp(j*0.05);alp*exp(-j*0.05)];

[b,a] = zp2tf(zer,poles,1);
fvtool(b,a);
title('alpha = 0.25');

alp = 0.5;
zer = [exp(j*0.05);exp(-j*0.05)];
poles = [alp*exp(j*0.05);alp*exp(-j*0.05)];

[b,a] = zp2tf(zer,poles,1);
fvtool(b,a);
title('alpha = 0.5');

alp = 0.75;
zer = [exp(j*0.05);exp(-j*0.05)];
poles = [alp*exp(j*0.05);alp*exp(-j*0.05)];

[b,a] = zp2tf(zer,poles,1);
fvtool(b,a);
title('alpha = 0.75');

alp = 1;
zer = [exp(j*0.05);exp(-j*0.05)];
poles = [alp*exp(j*0.05);alp*exp(-j*0.05)];

[b,a] = zp2tf(zer,poles,1);
fvtool(b,a);
title('alpha = 1');

% The notch seems to be exponentially decreasing as the alpha value is
% increased, with 0 notch at 1.

%c)

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```
% difference equation
%  $y[n] - 2\cos(w_0)y[n-1] + y[n-2] = x[n] - 2\alpha\cos(w_0)x[n-1] + \alpha^2 x[n-2]$ 
% alpha^2 x[n-2]
type notch

%d)
[b,a] = notch(0.05*pi, 0.99);

[bad,Fs] = audioread('bad_wannabe.wav');

good = filter(b,a,bad);

audiowrite('iired_wannabe.wav',good,Fs)

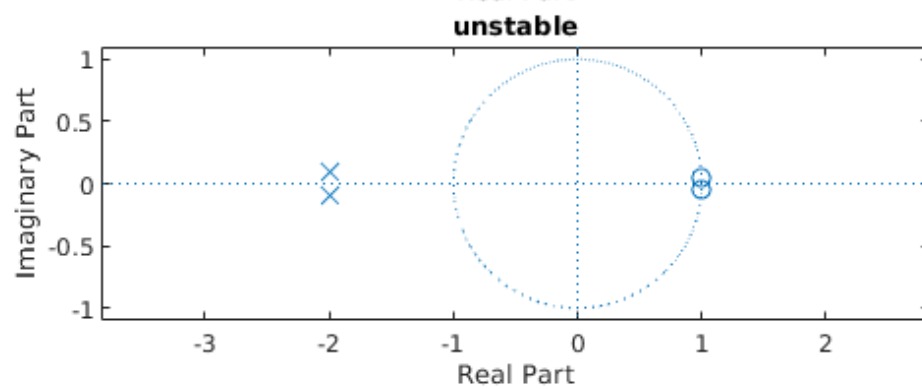
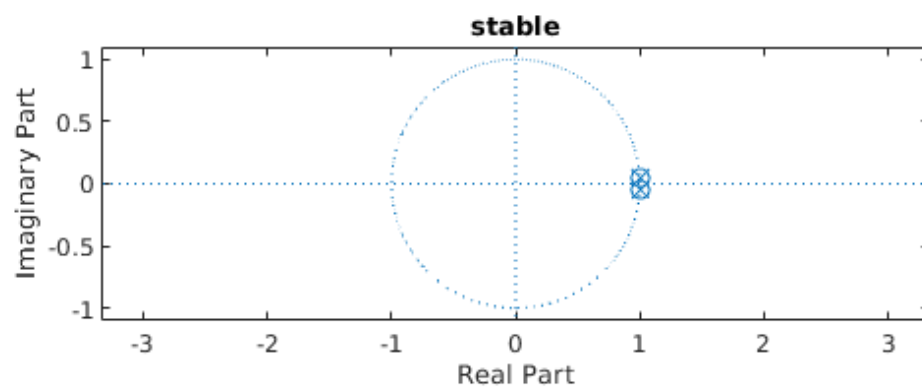
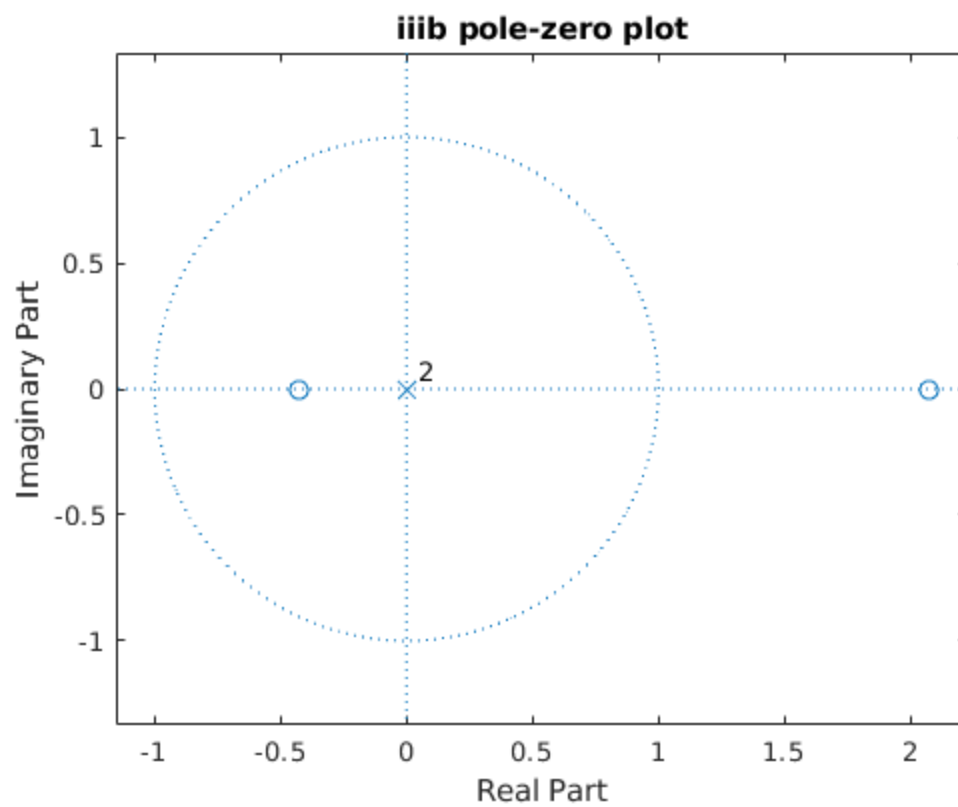
% The filtered audio is much more clear than when using then other
% nulling
% filter

w = -pi:pi/2000:pi;
good =dtft(good,w);

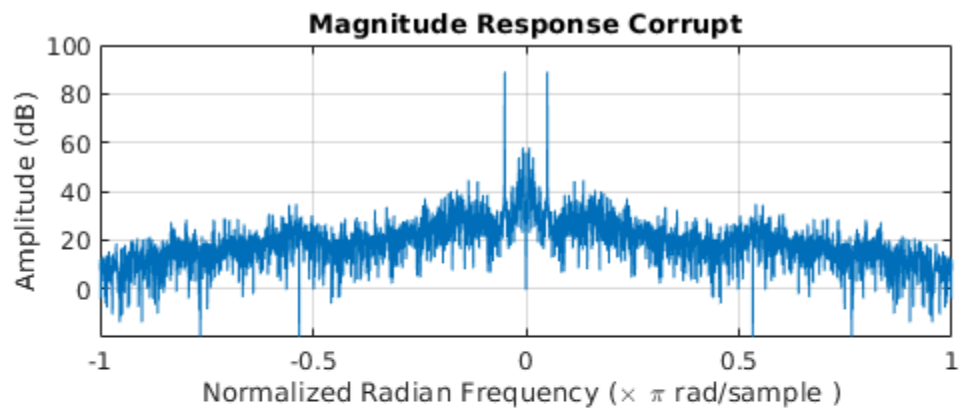
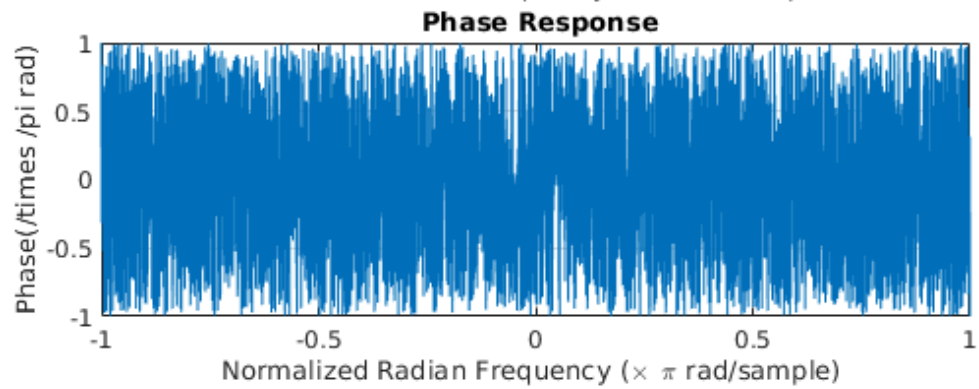
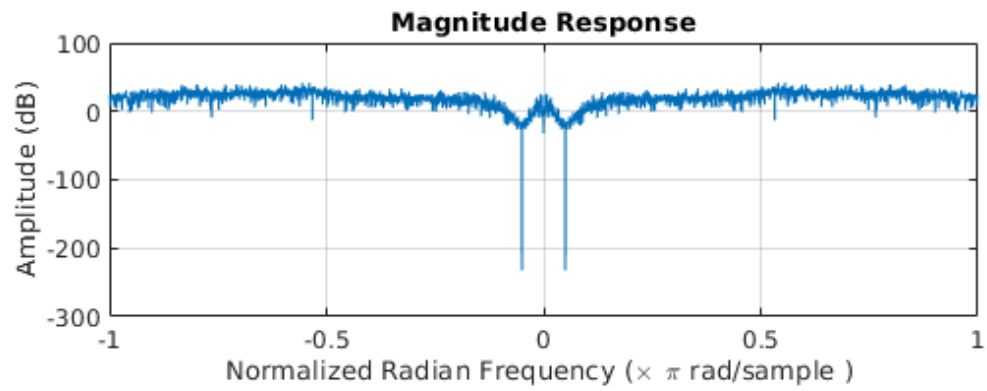
figure;
plot (w/pi,20*log10(abs(good))) ;
grid on ;
title ( 'Magnititude Response Corrupt')
xlabel ( 'Normalized Radian Frequency (\times \pi rad/sample ) ' ) ;
ylabel ( ' Amplitude (dB) ' ) ;

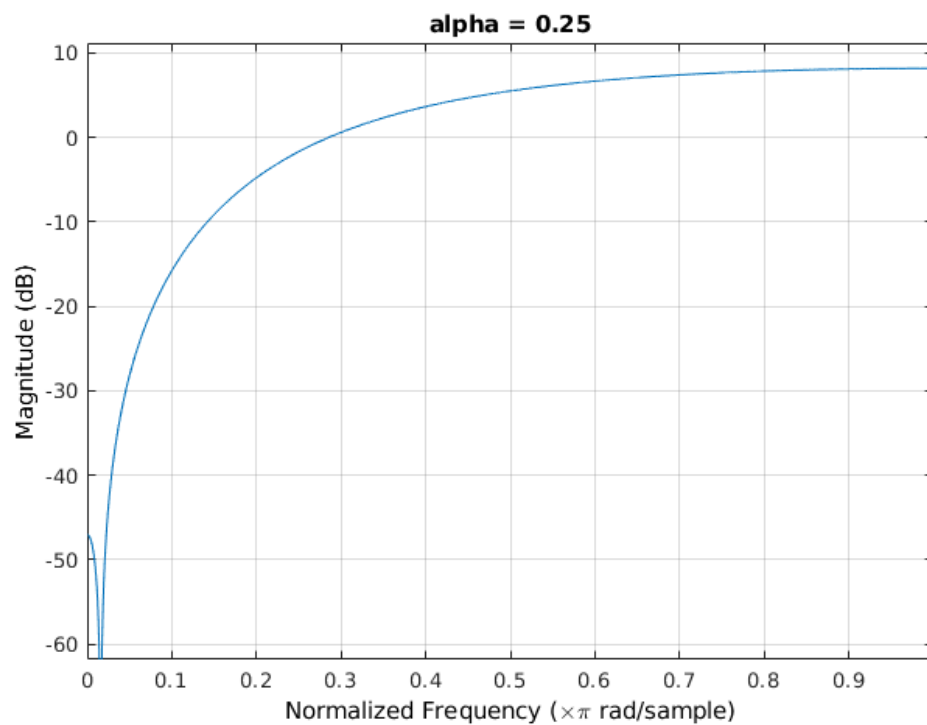
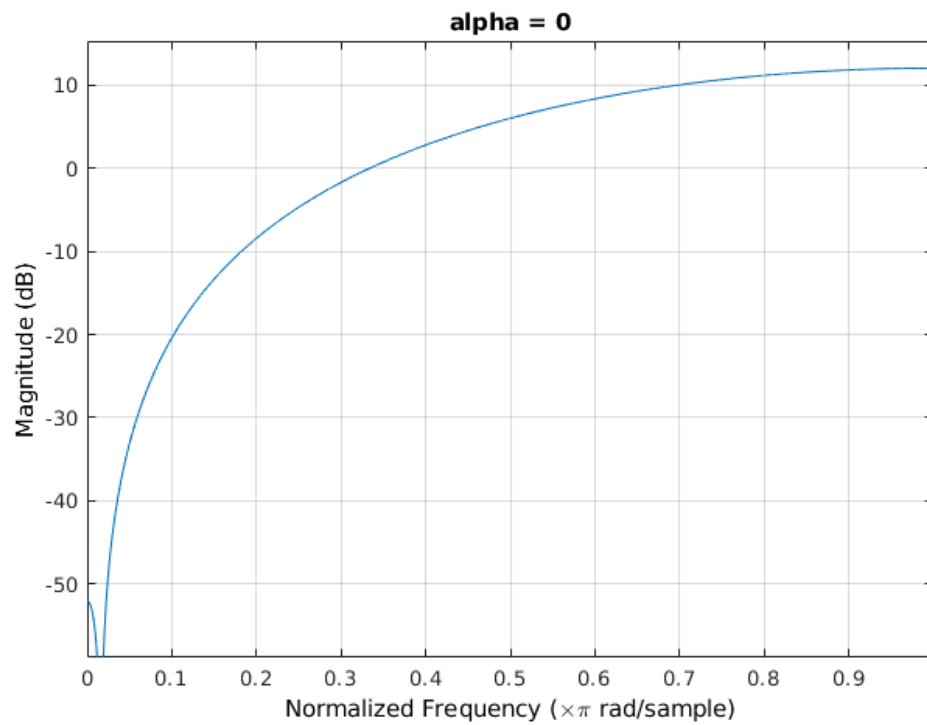
% The magnitude response shows that the notch filter did not
% completely
% eliminate the frequency at the where there was interference, but
% reduced
% it, leading to better audio quality

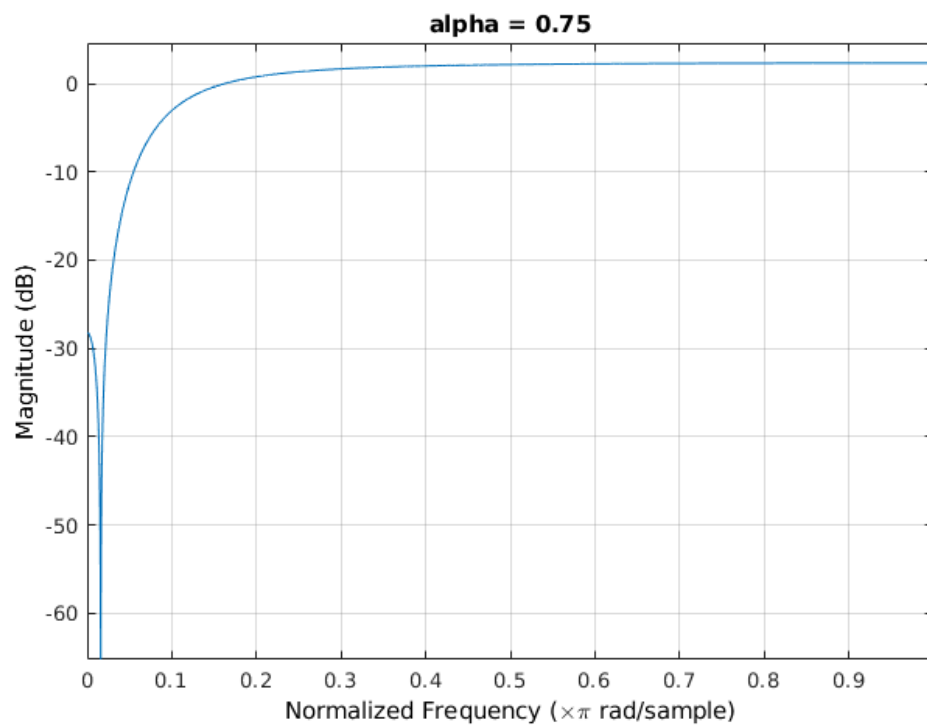
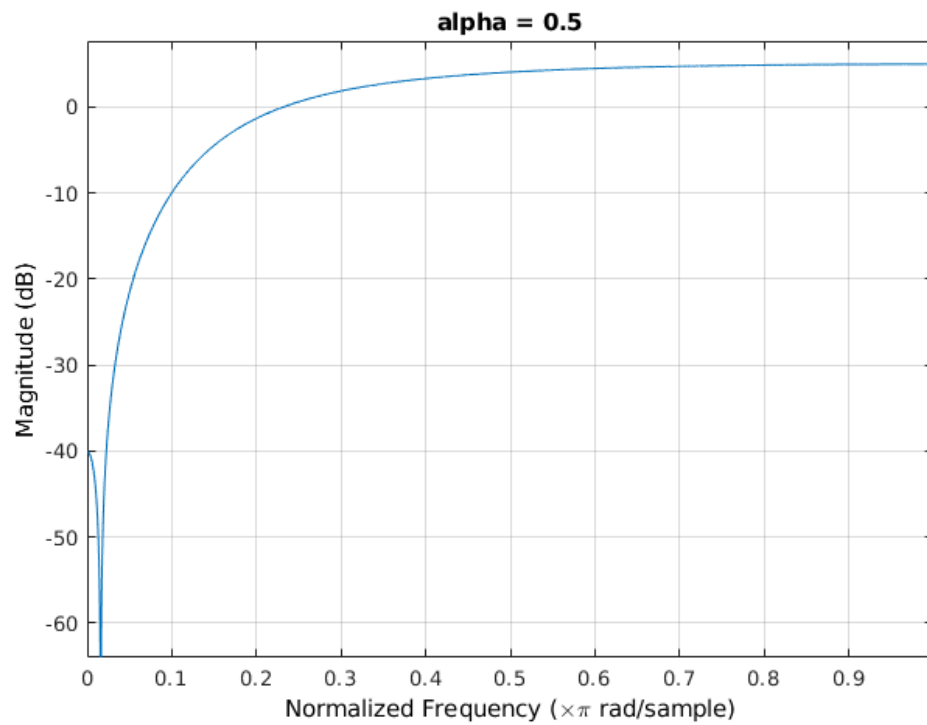
function [b,a] = notch(w0 , alpha)
b = [1 -2*cos(w0) 1];
a = [1 -2*alpha*cos(w0) alpha*alpha];
```

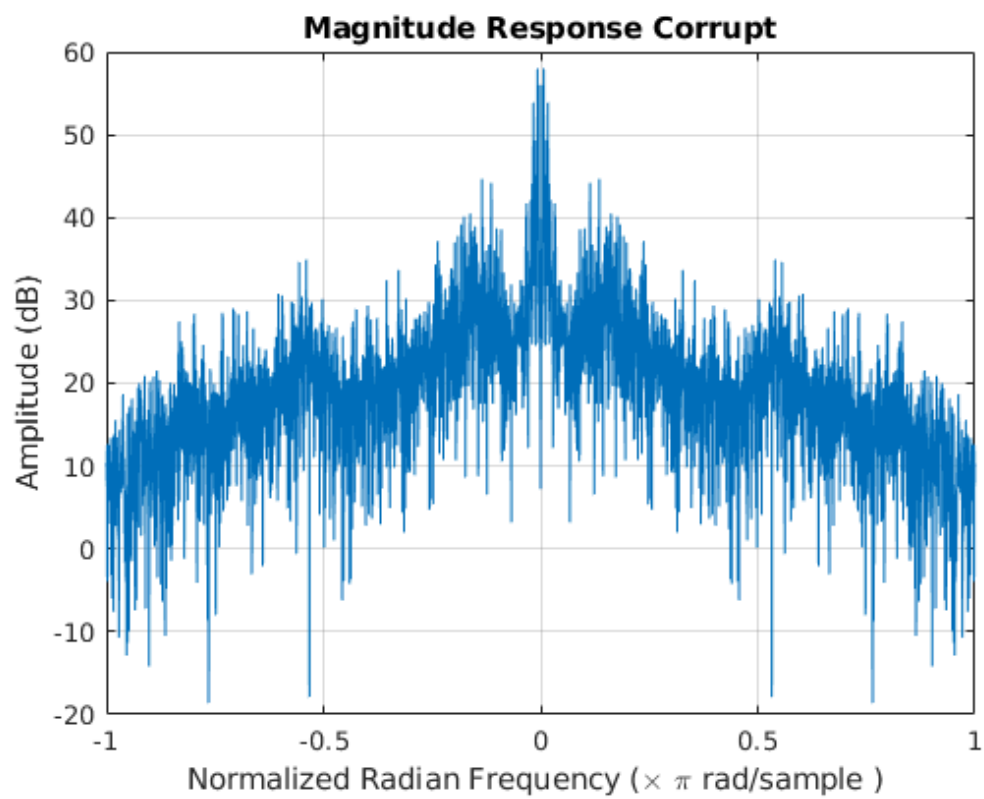
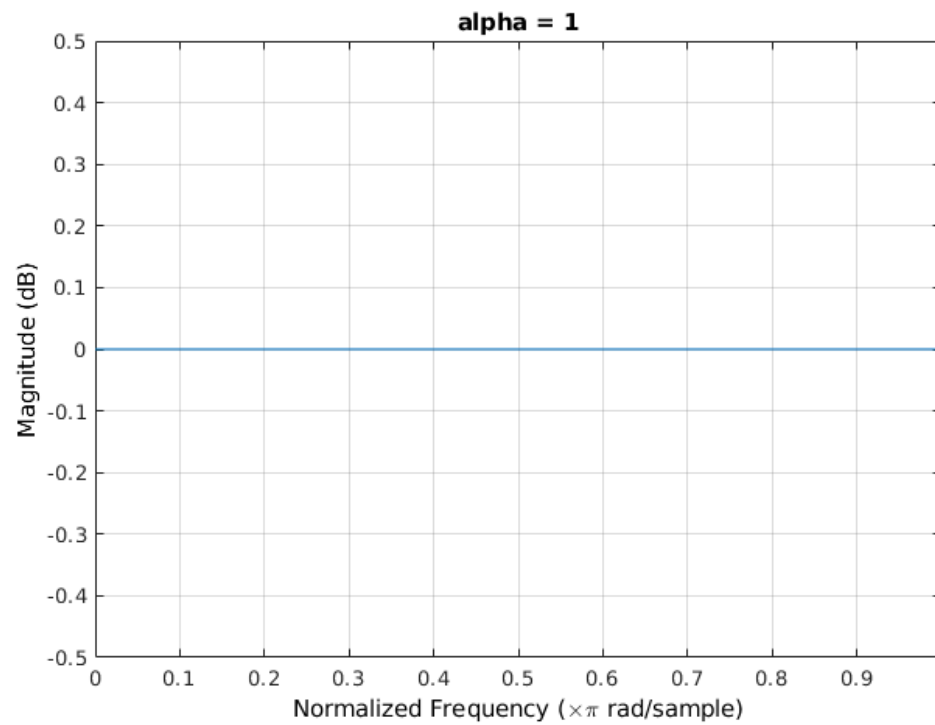












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