EESC6360

Project 2 – FIR and IIR Filtering

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Question 1:

Write a MATLAB Program to compute and display the poles and zeros, to compute and display the factor form, and to generate the pol-zero plot of the z-transform that is a ratio of two polynomials in z^{-1} . Using this program, analyze the z-transform G(z) of Equation 1.

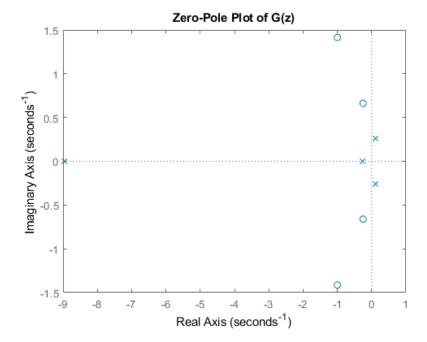
$$G(Z) = \frac{2 + 5z^{-1} + 9z^{-2} + 5z^{-3} + 3z^{-4}}{5 + 45z^{-1} + 2z^{-2} + z^{-3} + z^{-4}}$$
(1)

Solution:

```
% CODE:
num = [2,5,9,5,3];
den = [5,45,2,1,1];
[p_i1,z_i1] = question1(num,den,"G(z)");
function [poleG,zeroG] = question1(num,den,str)
syms z
poleG = roots(den);
zeroG = zeros(length(den)-1,1);
zeroG(1:length(num)-1,1) = roots(num);
disp("Zeros:")
disp(zeroG');
pause(1)
disp("Poles:");
disp(poleG');
pause(1)
denz = 1;
numz = 1;
for i = 1:length(zeroG)
    numz = numz * (1 - zeroG(i)*z^{-1});
for j = 1:length(poleG)
    denz = denz * ( 1 - poleG(j)*z^{-1});
end
disp("Factored G(z)")
pretty(numz);
if den ~= 1
   disp("-----
----")
    pretty(denz);
end
s = tf('s');
Gs = tf(num,den);
figure
pzplot(Gs*s^(length(den)-length(num)));
title("Zero-Pole Plot of " + str)
end
```

Output:

```
zeros:
 -1.0000 - 1.4142i -1.0000 + 1.4142i -0.2500 - 0.6614i -0.2500 + 0.6614i
Poles:
 -8.9576 + 0.0000i -0.2718 + 0.0000i
                                  0.1147 - 0.2627i
                                                    0.1147 + 0.2627i
Factored G(z)
                                          / sqrt(7) 1i
/ - 1 + sqrt(2) 1i
                    \setminus / 1 + sqrt(2) 1i
                       Z
                                       /\
  / sqrt(7) 1i
        4
          Z
```



Question 2:

Write a MATLAB program to compute and display the rational z-transform from its zeros, poles, and gain constant. Using this program, determine the rational form of a z-transform with the given zeros, poles, and gain constant.

Solution:

```
% CODE:
z_i = [0.3, 2.5, -0.2+1j*0.4, -0.2-1j*0.4];
p_i = [0.5, -0.75, 0.6+1j*0.7, 0.6-1j*0.7];
b0 = 3.9;
[b,a] = question2(z_i,p_i,b0)
function [b,a] = question2(z_i,pi,b0)
syms z
numz = 1;
denz = 1;
for i = 1:length(z_i)
   numz = numz*(z-z_i(i));
end
numz = expand(numz);
for j = 1:length(pi)
    denz = denz*(z-pi(j));
end
denz = expand(denz);
disp("G(z) = ");
pretty(b0*numz/denz)
b = sym2poly(numz);
a = sym2poly(denz);
end
```

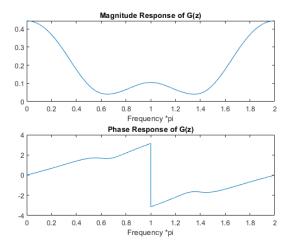
Output:

Question 3:

Write a MATLAB program to compute and display the magnitude and phase response of a filter with rational z-transform. Use this program to display the magnitude and phase response for G(z).

Solution:

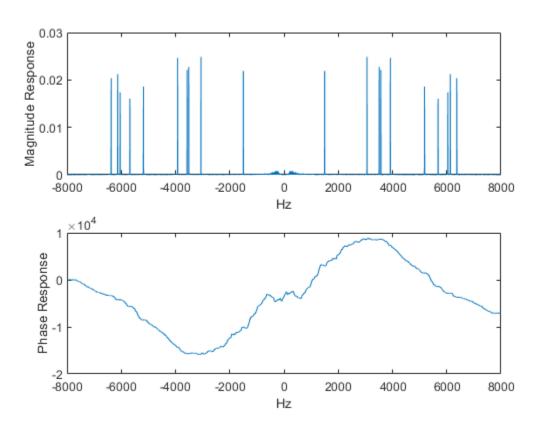
```
% CODE:
num = [2,5,9,5,3];
den = [5,45,2,1,1];
question3(num,den,"G(z)");
function question3(num,den,str)
w = 0:0.001:2*pi;
numw = 0;
denw = 0;
for i = 1:length(num)
    numw = numw + num(i).*exp(1j*w*(i-1));
end
for i = 1:length(den)
    denw = denw + den(i).*exp(1j*w*(i-1));
end
figure
subplot(211)
Hw = numw./denw;
plot(w/pi,abs(Hw))
xlabel('Frequency *pi')
title("Magnitude Response of " + str)
subplot(212)
plot(w/pi,angle(Hw))
xlabel('Frequency *pi')
title("Phase Response of " + str)
end
```



Question 4:

The audio signal in http://tinyurl.com/ja5vxce has been contaminated with multiple additive noisy signals with narrow spectral content. The goal of this project is to clean the audio by designing different notch filters

a. Plot the magnitude and phase response of the audio file. Discuss the results.



We can see 10 highly distinct peaks in the positive portion of the frequency response. Since the human voice typically is around 400-4kHz and usually is wideband, we expect every one of these peaks to correspond to additive noise. We will design notch filters to cancel out these frequencies.

b. Determine the spectral location associated with the noise. Give your results in the discrete time domain (normalizes frequencies) and continuous time domain (in Hz).

Frequencies of Noise (kHz): Columns 1 through 10

1.5000 3.0600 3.5100 3.5699 3.9200 5.1800 5.6799 6.0400 6.1299 6.3699

Discrete Normalized Frequencies of Noise:

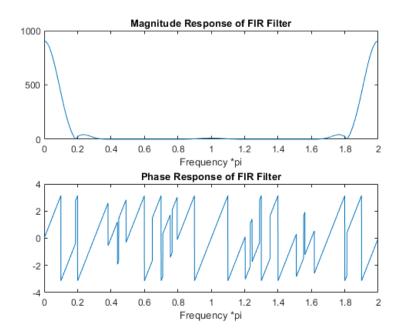
Columns 1 through 10

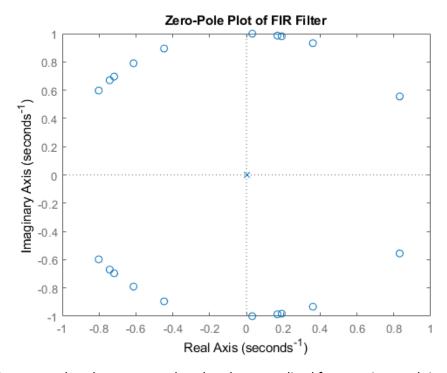
0.0937 0.1912 0.2194 0.2231 0.2450 0.3237 0.3550 0.3775 0.3831 0.3981

c. Design a notch FIR filter with only zeros. Plot the magnitude and phase response of the filter. Plot the magnitude response of the resulting signal. Create an audio file after filtering the signal.

Our FIR notch filter will have the following properties:

We will place zeros at the frequencies which contain noise and we desire to eliminate.





We can see that the zeros we placed at the normalized frequencies result in a 0 magnitude response of the filter at those normalized frequencies.

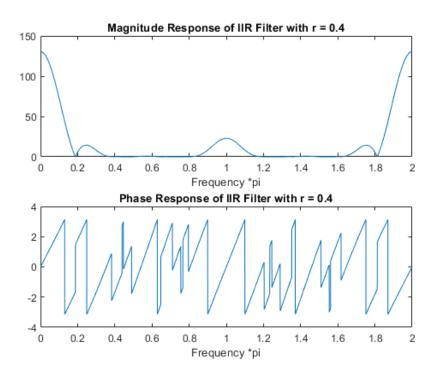
```
zeros:
 Columns 1 through 4
  Columns 5 through 8
  0.1912 - 0.9815i 0.1912 + 0.9815i 0.1681 - 0.9858i 0.1681 + 0.9858i
 Columns 9 through 12
  0.0314 - 0.9995i 0.0314 + 0.9995i -0.4470 - 0.8945i -0.4470 + 0.8945i
 Columns 13 through 16
 -0.8020 - 0.5973i -0.8020 + 0.5973i -0.7423 - 0.6701i -0.7423 + 0.6701i
 Columns 17 through 20
 -0.7181 - 0.6959i -0.7181 + 0.6959i -0.6129 - 0.7902i -0.6129 + 0.7902i
Factored G(z)
 566085731962705 9002750976819333
 18014398509481984 9007199254740992
   566085731962705 9002750976819333
    18014398509481984 9007199254740992
  ----i
9007199254740992 4503599627370496
    5520343083264219 3558632133896769
    9007199254740992 4503599627370496
    1513968334038139 4439525264545937
    9007199254740992 4503599627370496
    1513968334038139 4439525264545937
    9007199254740992 4503599627370496
    234039014659009 2502038957673595
    281474976710656 4503599627370496
    234039014659009 2502038957673595
    281474976710656 4503599627370496
    6685597574028107 6035927724275377
    9007199254740992 9007199254740992
                   Z
```

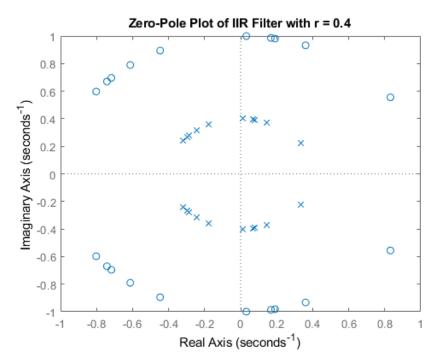
```
6685597574028107 6035927724275377
9007199254740992 9007199254740992
3445177072878327 8840946055072791
-----i
18014398509481984 9007199254740992
3445177072878327 8840946055072791
18014398509481984 9007199254740992
6468296862995579 6268235326380601
9007199254740992 9007199254740992
6468296862995579 6268235326380601
9007199254740992 9007199254740992
2013000991858733 8057353563157265
4503599627370496 9007199254740992
2013000991858733 8057353563157265
4503599627370496 9007199254740992
812504570253995 4200208911103973
2251799813685248 4503599627370496
812504570253995 4200208911103973
2251799813685248 4503599627370496
3611975720343689 5379978066609391
4503599627370496 9007199254740992
3611975720343689 5379978066609391
4503599627370496 9007199254740992
```

d. Design a notch IIR filter with zeros and poles. Set the magnitude of the poles to r = 0.4 (pk = $r*e^(jw_k)$). Plot the magnitude and phase response of the filter. Plot the magnitude response of the resulting signal. Create an audio file after filtering the signal.

Our IIR notch filter will have the following properties:

We will keep the zeros in the same locations as for the FIR filter, but we will at poles at the same frequencies to narrow the bandwidth of the notch. For this IIR filter, we will place our poles at a radial distance of 0.4 from the origin of the z-plane.





By adding poles at the same normalized frequencies as the zeroes, we increase the quality factor (decrease the bandwidth) of the notch filter at each of the normalized frequencies.

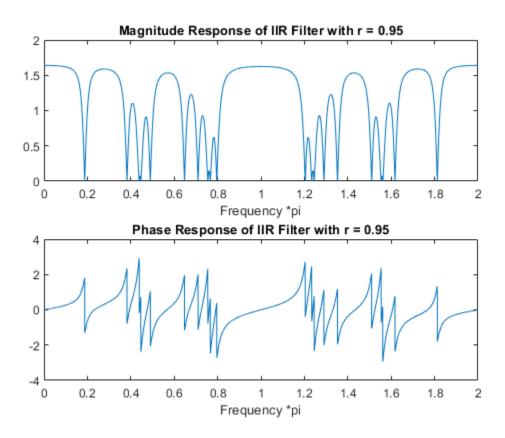
```
zeros:
 Columns 1 through 4
  0.8315 - 0.5556i
                 Columns 5 through 8
  0.1912 - 0.9815i 0.1912 + 0.9815i 0.1681 - 0.9858i
                                                  0.1681 + 0.9858i
 Columns 9 through 12
  0.0314 - 0.9995i
                  0.0314 + 0.9995i -0.4470 - 0.8945i -0.4470 + 0.8945i
 Columns 13 through 16
 -0.8020 - 0.5973i -0.8020 + 0.5973i -0.7423 - 0.6701i -0.7423 + 0.6701i
 Columns 17 through 20
 -0.7181 - 0.6959i -0.7181 + 0.6959i -0.6129 - 0.7902i -0.6129 + 0.7902i
Poles:
 Columns 1 through 4
  0.1443 + 0.3731i
 Columns 5 through 8
  0.0765 - 0.3926i
                 0.0765 + 0.3926i 0.0672 - 0.3943i 0.0672 + 0.3943i
 Columns 9 through 12
  0.0126 - 0.3998i \quad 0.0126 + 0.3998i \quad -0.1788 - 0.3578i \quad -0.1788 + 0.3578i
 Columns 13 through 16
 -0.3208 - 0.2389i -0.3208 + 0.2389i -0.2969 - 0.2680i -0.2969 + 0.2680i
 Columns 17 through 20
 -0.2873 - 0.2784i -0.2873 + 0.2784i -0.2452 - 0.3161i -0.2452 + 0.3161i
Factored G(z)
 566085731962705
                  9002750976819333
 18014398509481984 9007199254740992
 ----- - 1 |
   566085731962705 9002750976819333
                     -----i
    18014398509481984 9007199254740992
    5520343083264219 3558632133896769
    9007199254740992 4503599627370496
    5520343083264219 3558632133896769
    9007199254740992 4503599627370496
    -----+ 1
    1513968334038139 4439525264545937
    9007199254740992 4503599627370496
```

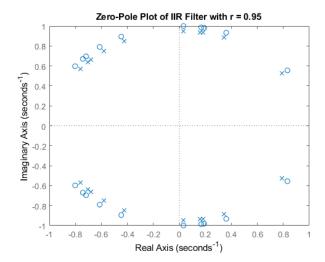
```
1513968334038139 4439525264545937
 9007199254740992 4503599627370496
 234039014659009 2502038957673595
 281474976710656 4503599627370496
 234039014659009 2502038957673595
 6685597574028107 6035927724275377
 9007199254740992 9007199254740992
 6685597574028107 6035927724275377
 9007199254740992 9007199254740992
 3445177072878327 8840946055072791
 18014398509481984 9007199254740992
 3445177072878327 8840946055072791
                  -----i
 18014398509481984 9007199254740992
 6468296862995579 6268235326380601
 9007199254740992 9007199254740992
 6468296862995579 6268235326380601
 9007199254740992 9007199254740992
 -----+ 1
 2013000991858733 8057353563157265
 4503599627370496 9007199254740992
 2013000991858733 8057353563157265
 4503599627370496 9007199254740992
 812504570253995 4200208911103973
 2251799813685248 4503599627370496
 812504570253995 4200208911103973
2251799813685248 4503599627370496
```

e. Repeat part (d) with r = 0.95

Our IIR notch filter will have the following properties:

We will keep the zeros in the same locations as for the FIR filter, but we will at poles at the same frequencies to narrow the bandwidth of the notch. For this IIR filter, we will place our poles at a radial distance of 0.4 from the origin of the z-plane.





By increasing the radial magnitude of the poles and placing them closer to the unit circle, we can see that the quality factor is drastically increased and the notch filter stops the desired normalized frequencies and allows other frequencies to pass.

```
zeros:
  Columns 1 through 4
  0.8315 - 0.5556i
                     0.8315 + 0.5556i
                                        0.3608 - 0.9326i
                                                            0.3608 + 0.9326i
 Columns 5 through 8
  0.1912 - 0.9815i
                     0.1912 + 0.9815i
                                        0.1681 - 0.9858i
                                                            0.1681 + 0.9858i
  Columns 9 through 12
  0.0314 - 0.9995i
                     0.0314 + 0.9995i -0.4470 - 0.8945i
                                                          -0.4470 + 0.8945i
 Columns 13 through 16
  -0.8020 - 0.5973i -0.8020 + 0.5973i -0.7423 - 0.6701i -0.7423 + 0.6701i
 Columns 17 through 20
  -0.7181 - 0.6959i -0.7181 + 0.6959i -0.6129 - 0.7902i
                                                          -0.6129 + 0.7902i
Poles:
 Columns 1 through 4
  0.7899 - 0.5278i
                     0.7899 + 0.5278i
                                        0.3428 - 0.8860i
                                                            0.3428 + 0.8860i
 Columns 5 through 8
  0.1817 - 0.9325i
                     0.1817 + 0.9325i
                                        0.1597 - 0.9365i
                                                            0.1597 + 0.9365i
 Columns 9 through 12
  0.0299 - 0.9495i
                     0.0299 + 0.9495i - 0.4246 - 0.8498i
                                                          -0.4246 + 0.8498i
 Columns 13 through 16
  -0.7619 - 0.5674i -0.7619 + 0.5674i -0.7051 - 0.6366i -0.7051 + 0.6366i
 Columns 17 through 20
  -0.6822 - 0.6611i - 0.6822 + 0.6611i - 0.5822 - 0.7507i - 0.5822 + 0.7507i
Factored G(z)
```

```
566085731962705
                9002750976819333
18014398509481984 9007199254740992
  566085731962705
                   9002750976819333
  18014398509481984 9007199254740992
  5520343083264219 3558632133896769
   9007199254740992 4503599627370496
  5520343083264219 3558632133896769
   ----i
9007199254740992 + 4503599627370496
  1513968334038139 4439525264545937
   9007199254740992 4503599627370496
  1513968334038139 4439525264545937
   9007199254740992 4503599627370496
  234039014659009 2502038957673595
   281474976710656 4503599627370496
   234039014659009 2502038957673595
   281474976710656 4503599627370496
  6685597574028107 6035927724275377
   9007199254740992 9007199254740992
   6685597574028107 6035927724275377
   9007199254740992 9007199254740992
   3445177072878327 8840946055072791
   18014398509481984 9007199254740992
   3445177072878327 8840946055072791
  18014398509481984 9007199254740992
  6468296862995579 6268235326380601
  9007199254740992 9007199254740992
```

```
-----+ 1 |
6468296862995579 6268235326380601
9007199254740992 9007199254740992
2013000991858733 8057353563157265
4503599627370496 9007199254740992
2013000991858733 8057353563157265
4503599627370496 9007199254740992
-----+ 1
812504570253995 4200208911103973
2251799813685248 4503599627370496
812504570253995 4200208911103973
2251799813685248 4503599627370496
3611975720343689 5379978066609391
4503599627370496 9007199254740992
3611975720343689 5379978066609391
4503599627370496 9007199254740992
```

Code for question 4:

```
[y,Fs] = audioread('project2audio.wav');
[y_new, y_new1, y_new2] = question4(y, Fs, 0.4, 0.95);
soundsc(y_new2,Fs)
function [y_new,y_new1,y_new2] = question4(y,Fs,r1,r2)
L = length(y);
yf = fftshift(fft(y))/L;
f = -Fs/2:Fs/L:Fs/2-Fs/L;
figure
subplot(211);
plot(f,abs(yf));
ylabel('Magnitude Response')
xlabel('Hz')
% plot phase
subplot(212);
plot(f,unwrap(angle(yf)));
ylabel('Phase Response')
xlabel('Hz')
```

```
[peaks,noi\_loc] = findpeaks(abs(yf((length(yf)+1)/2 : length(yf))),f((length(yf)+1)/2 : length(yf))
length(yf) ), 'MinPeakHeight', 0.01);
disp("Frequencies of Noise (Hz):")
disp(noi_loc);
disp("Discrete Normalized Frequencies of Noise:")
disp(noi_loc/Fs);
syms z
% Make an FIR filter with zeros at noise frequencies
numz = 1;
%noi_loc = mean(noi_loc);
for i = 1:length(noi_loc)
   numz = numz .* (z^2-2*cos(2*pi*noi_loc(i)/Fs)*z+1);
numHz = sym2poly(numz);
question3(numHz,1,"FIR Filter");
question1(numHz,1,"FIR Filter");
y_new = filter(numHz,1,y);
% Make an IIR filter with r1
denz = 1;
for i = 1:length(noi_loc)
   denz = denz .* (z^2-2*r1*cos(2*pi*noi_loc(i)/Fs)*z+r1^2);
end
denHz1 = sym2poly(denz);
question3(numHz,denHz1,"IIR Filter with r = 0.4");
question1(numHz,denHz1,"IIR Filter with r = 0.4");
y_new1 = filter(numHz,denHz1,y);
% Make an IIR filter with r2
denz = 1;
for i = 1:length(noi_loc)
   denz = denz .* (z^2-2*r^2*cos(2*pi*noi_loc(i)/Fs)*z+r^2);
denHz2 = sym2poly(denz);
question3(numHz,denHz2,"IIR Filter with r = 0.95");
question1(numHz,denHz2,"IIR Filter with r = 0.95");
y_new2 = filter(numHz,denHz2,y);
function question3(num,den,str)
w = 0:0.001:2*pi;
numw = 0;
denw = 0;
for i = 1:length(num)
    numw = numw + num(i).*exp(1j*w*(i-1));
end
```

```
for i = 1:length(den)
    denw = denw + den(i).*exp(1j*w*(i-1));
end
figure
subplot(211)
Hw = numw./denw;
plot(w/pi,abs(Hw))
xlabel('Frequency *pi')
title("Magnitude Response of " + str)
subplot(212)
plot(w/pi,angle(Hw))
xlabel('Frequency *pi')
title("Phase Response of " + str)
end
function [poleG,zeroG] = question1(num,den,str)
syms z
poleG = roots(den);
zeroG = zeros(length(den)-1,1);
zeroG(1:length(num)-1,1) = roots(num);
disp("Zeros:")
disp(zeroG');
pause(1)
disp("Poles:");
disp(poleG');
pause(1)
denz = 1;
numz = 1;
for i = 1:length(zeroG)
    numz = numz * (1 - zeroG(i)*z^{-1});
end
for j = 1:length(poleG)
    denz = denz * ( 1 - poleG(j)*z^{-1});
end
disp("Factored G(z)")
pretty(numz);
if den \sim= 1
    disp("-----
----")
    pretty(denz);
end
s = tf('s');
Gs = tf(num,den);
pzplot(Gs*s^(length(den)-length(num)));
title("Zero-Pole Plot of " + str)
% Audiofiles written
```