

Hacettepe University

Department of Electrical and Electronics Engineering

ELE 409 Digital Signal Processing Laboratory

EXPERIMENT 4 – IIR FILTER PRELIMINARY WORK

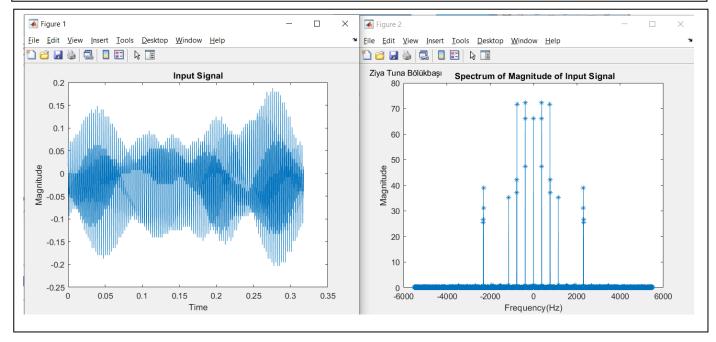
Ziya Tuna Bölükbaşı

21628166

3/12/2022

Matlab Code

```
clc;
clear all;
close all;
[x input,fs]=audioread('C:\Users\Tuna\OneDrive - hacettepe.edu.tr\Masaüstü\dersler\2022
 .dönem\dsplab\pre-exp4\sound4.wav');
fft of_x=fft(x_input)
fftshift_of_x=fftshift(abs(fft_of_x));
length fftx=length(fftshift_of_x);
length x=length(x input);
range_of_frequency=-fs/2:fs/length_fftx:fs/2-fs/length_fftx;
time=[1:length_x]/fs;
figure;
plot(time,x_input);
title('Input Signal');
xlabel('Time');
ylabel('Magnitude');
figure;
plot(range_of_frequency,fftshift_of_x,'-*');
title(' Spectrum of Magnitude of Input Signal');
xlabel('Frequency(Hz)')
ylabel('Magnitude')
gtext('Ziya Tuna Bölükbaşı')
peak_values_of_x=mypeakfinder(x_input,5,fs);
function y=mypeakfinder(x,P,Fs)
fft of x=fft(x)
fftshift_of_x=fftshift(abs(fft_of_x));
length_fftx=length(fftshift_of_x);
range_of_frequency=-Fs/2:Fs/length_fftx:Fs/2-Fs/length_fftx;
sort_fftx=sortrows(fftshift_of_x,-1);
for k=2:2:2*P
    y(k/2) = sort_fftx(k)
    for i=1:length(fftshift of x)
        if fftshift of x(i) == y(k/2)
             sort_fftx1(k)=i;
    end
    y(k/2) = range_of_frequency(sort_fftx1(k));
end
end
```



Top 5 peak values frequencies(Hz):

381.1500	762.3000	378.0000	384.3000	774.9000	
 			·		

3) The first peak value is 381.15 Hz.

wp=end of the passband

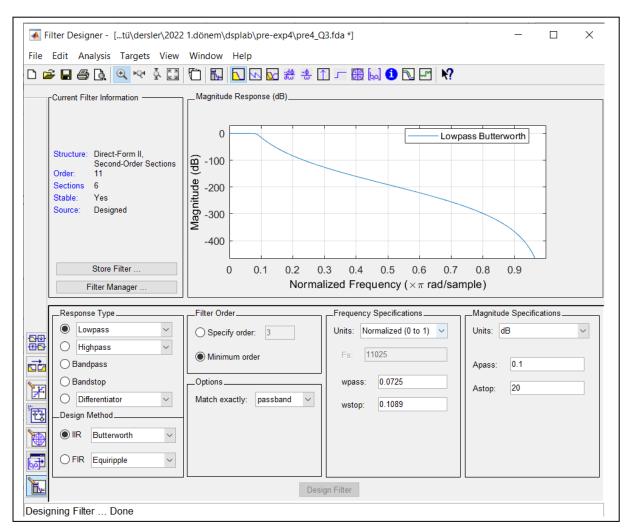
ws= beginning of stopband

wp ve ws values are found by looking at the ratio of the frequency value. Nyquist frequency value is accepted as 1 and frequency value we want is proportioned accordingly.

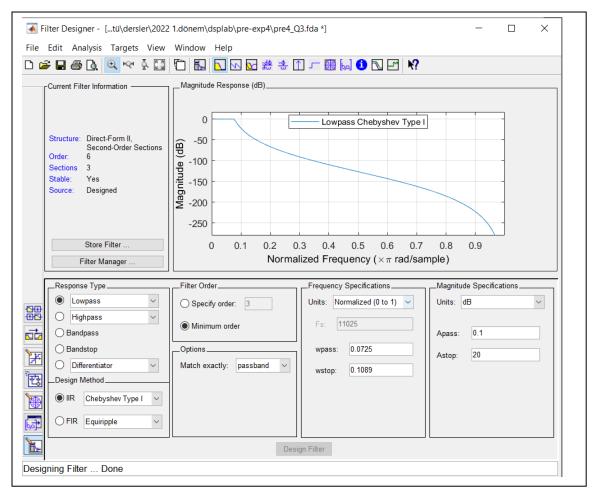
So, sample frequency is 11025 Hz and Nyquist frequency is 5512.5.

Frequency I want for wp is 400Hz.So wp is 0.0725.Frequency I want for ws is 600 Hz.So ws is 0.1089

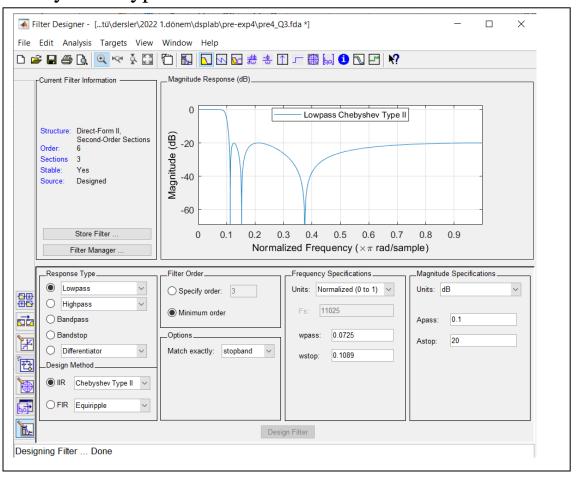
Butterworth Filter Low Pass:



Chebyshev Type 1 Low Pass Filter:

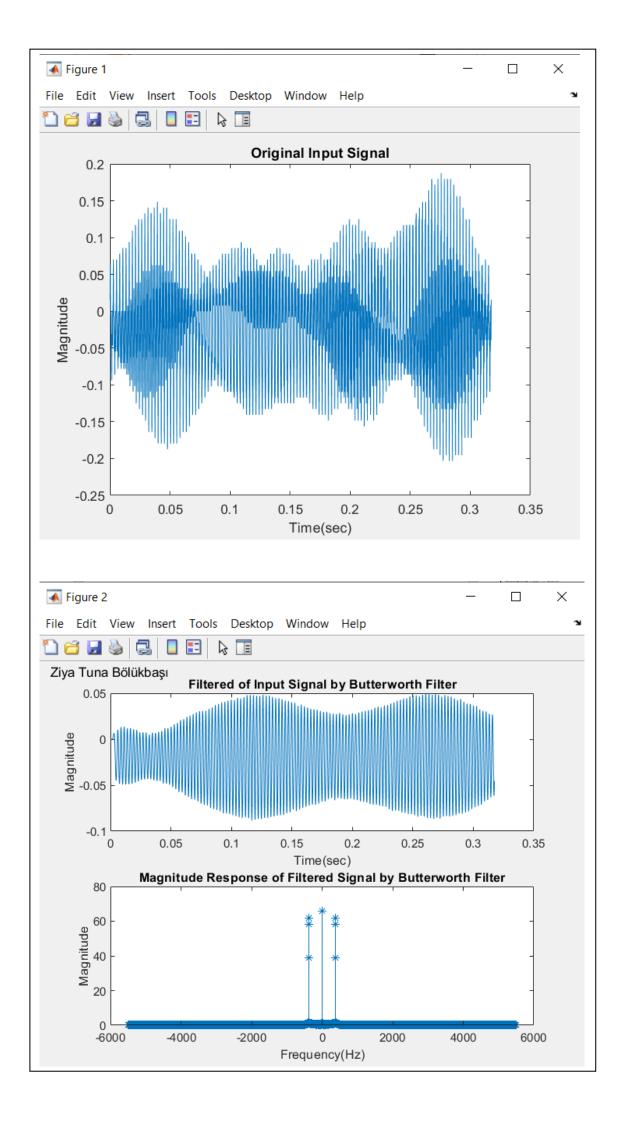


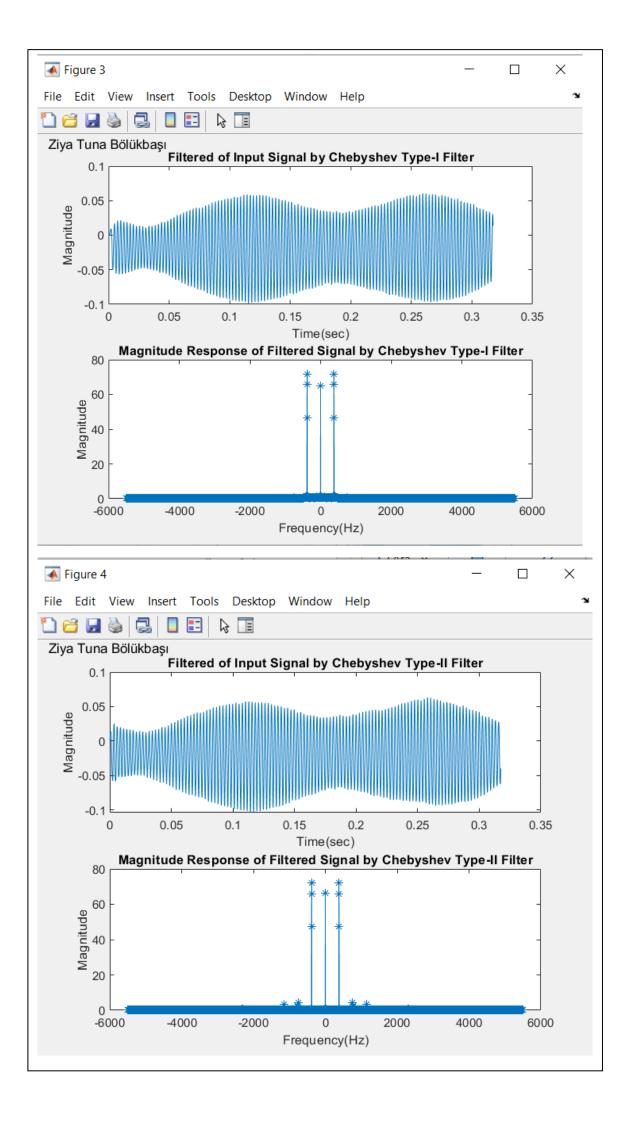
Chebyshev Type 2 Filter Low Pass:



Matlab Code

```
clear all;
close all;
[x\_input,fs] = audioread('C:\Users\Tuna\OneDrive - hacettepe.edu.tr\Masaüstü\dersler\2022) + audioread('C:\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users\Users
1.dönem\dsplab\pre-exp4\sound4.wav');
F=400:
Fs=11025;
[y,x] = butter(11,F/(Fs/2));
filtered xinput=filter(y,x,x input);
length x=length(x input);
time=[1:length_x]/fs;
figure;
plot(time, x input);
title('Original Input Signal');
xlabel('Time(sec)');
ylabel('Magnitude');
figure;
subplot(2,1,1)
plot(time, filtered xinput);
fft of fx=fft(filtered xinput)
fftshift_of_fx=fftshift(abs(fft_of_fx));
length fftfx=length(fftshift of fx);
length x=length(x input);
range_of_frequency=-fs/2:fs/length_fftfx:fs/2-fs/length_fftfx;
title('Filtered of Input Signal by Butterworth Filter');
xlabel('Time(sec)');
ylabel('Magnitude');
subplot(2,1,2)
stem(range_of_frequency,fftshift_of_fx,'-*');
title('Magnitude Response of Filtered Signal by Butterworth Filter');
xlabel('Frequency(Hz)');
ylabel('Magnitude');
gtext('Ziya Tuna Bölükbaşı');
figure;
subplot(2,1,1)
F=400;
Fs=11025;
[y,x] = \text{cheby1}(6,0.1,F/(Fs/2));
filtered xinput=filter(y,x,x input);
plot(time, filtered xinput);
title('Filtered of Input Signal by Chebyshev Type-I Filter');
xlabel('Time(sec)');
ylabel('Magnitude');
fft of fx=fft(filtered xinput)
fftshift of fx=fftshift(abs(fft of fx));
length_fftfx=length(fftshift_of_fx);
length_x=length(x_input);
range of frequency=-fs/2:fs/length fftfx:fs/2-fs/length fftfx;
subplot (2,1,2)
plot(range_of_frequency,fftshift_of_fx,'-*');
title('Magnitude Response of Filtered Signal by Chebyshev Type-I Filter');
xlabel('Frequency(Hz)');
ylabel('Magnitude');
gtext('Ziya Tuna Bölükbaşı');
figure;
subplot(2,1,1)
F=600;
Fs=11025;
[y,x] = \text{cheby2}(6,20,F/(Fs/2));
filtered xinput=filter(y,x,x input);
plot(time, filtered xinput);
fft of fx=fft(filtered xinput)
fftshift_of_fx=fftshift(abs(fft_of_fx));
length fftfx=length(fftshift of fx);
length x=length(x input);
range of frequency=-fs/2:fs/length fftfx:fs/2-fs/length fftfx;
title('Filtered of Input Signal by Chebyshev Type-II Filter');
xlabel('Time(sec)');
ylabel('Magnitude');
subplot(2,1,2)
plot(range_of_frequency,fftshift_of_fx,'-*');
title ('Magnitude Response of Filtered Signal by Chebyshev Type-II Filter');
xlabel('Frequency(Hz)');
ylabel('Magnitude');
gtext('Ziya Tuna Bölükbaşı');
```





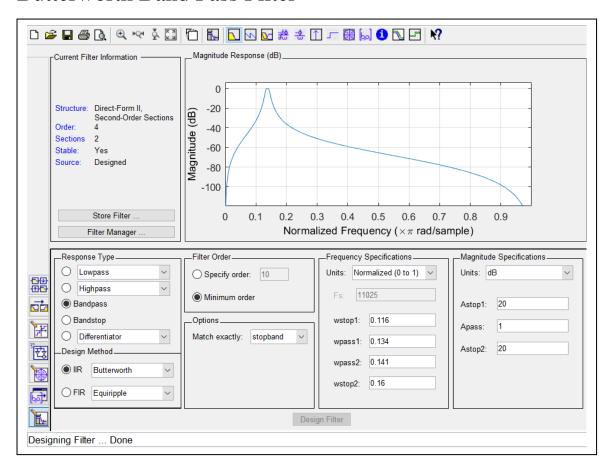
When we use the Butterworth and Chebyshev type-1 filter, we get a signal much closer to the desired output. However, in the Chebyshev type-2 filter, the attenuation is lower and fluctuations occur in the regions close to the cut-off frequency.

6) The second dominant peak value is 762.3 Hz

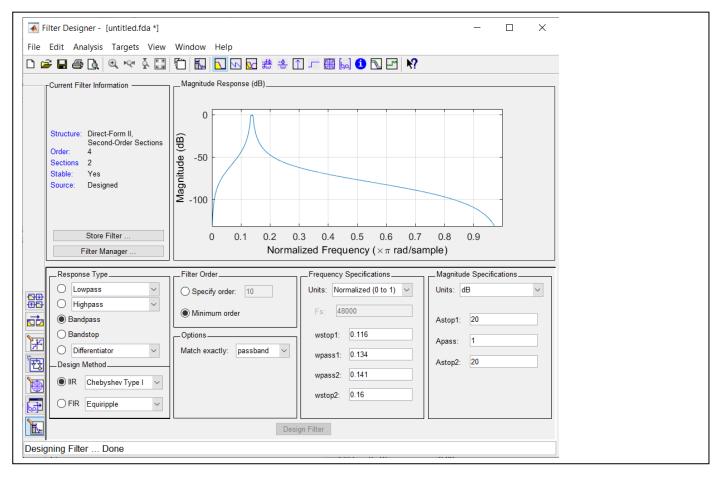
wstop1=0.116 wpass1=0.134 wpass2=0.141 wstop2=0.16

7)

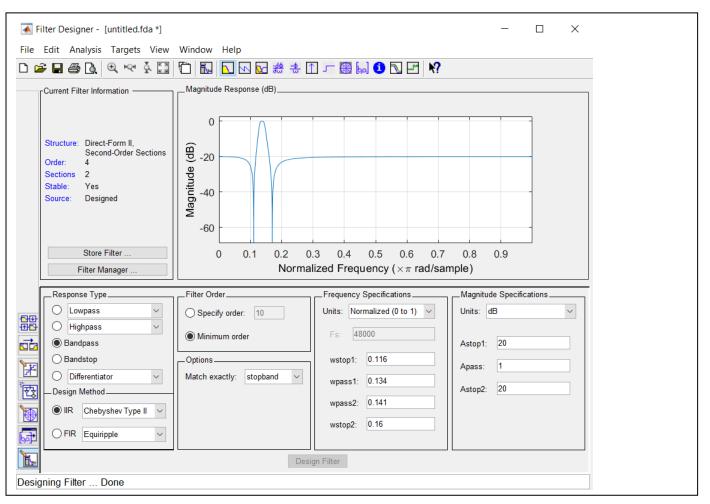
Butterworth Band Pass Filter



Chebyshev Type 1 Band Pass Filter:

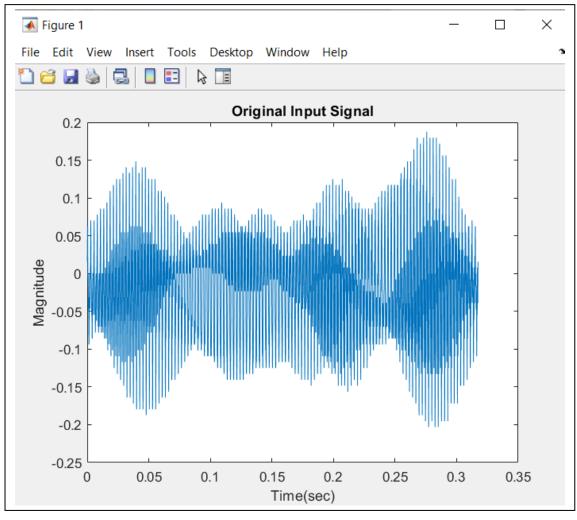


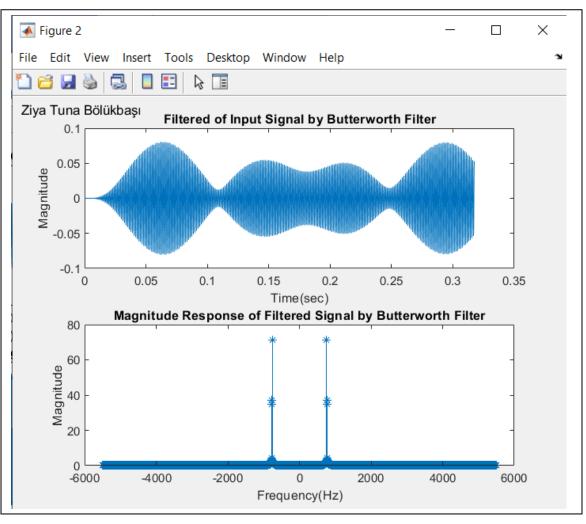
Chebyshev Type 2 Band Pass Filter:

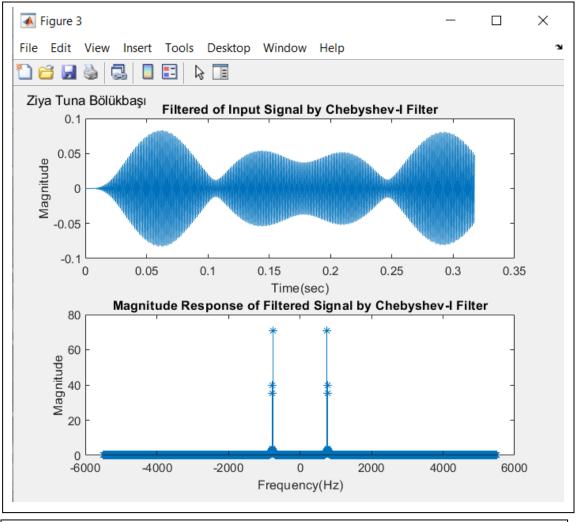


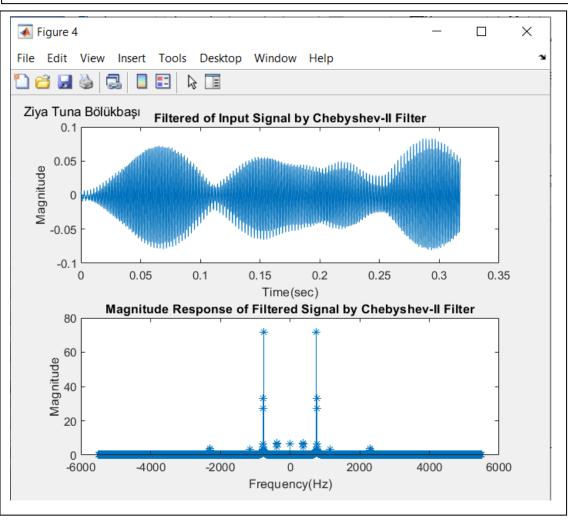
8)Matlab Code

```
clc;
clear all;
close all;
[x_input,fs]=audioread('C:\Users\Tuna\OneDrive - hacettepe.edu.tr\Masaüstü\dersler\2022 1.dönem\dsplab\pre-exp4\sound4.wav');
Fs=11025:
figure;
length_x=length(x_input);
time=[1:length_x]/fs;
plot(time,x_input);
title('Original Input Signal');
xlabel('Time(sec)');
ylabel('Magnitude');
[y,x]=butter(4,[740,780]/(Fs/2),'bandpass');%(order,fc/fs/2,'bandpass')
filtered_xinput=filter(y,x,x_input);
figure;
subplot(2,1,1)
plot(time,filtered_xinput);
fft_of_fx=fft(filtered_xinput)
fftshift_of_fx=fftshift(abs(fft_of_fx));
length_fftfx=length(fftshift_of_fx);
length_x=length(x_input);
range\_of\_frequency = -fs/2:fs/length\_fftfx:fs/2-fs/length\_fftfx;
title('Filtered of Input Signal by Butterworth Filter');
xlabel('Time(sec)');
ylabel('Magnitude');
subplot(2,1,2)
stem(range_of_frequency,fftshift_of_fx,'-*');
title('Magnitude Response of Filtered Signal by Butterworth Filter');
xlabel('Frequency(Hz)');
ylabel('Magnitude');
gtext('Ziya Tuna Bölükbaşı');
%chebs1
[y,x] = cheby 1 (4,0.1,[740,780]/(Fs/2)); \% (order,ripple-Apass,Fc/fs/2)
filtered_xinput=filter(y,x,x_input);
length_x=length(x_input);
time=[1:length_x]/fs;
figure;
subplot(2,1,1)
plot(time,filtered_xinput);
fft_of_fx=fft(filtered_xinput)
fftshift_of_fx=fftshift(abs(fft_of_fx));
length_fftfx=length(fftshift_of_fx);
length_x=length(x_input);
range_of_frequency=-fs/2:fs/length_fftfx:fs/2-fs/length_fftfx;
title('Filtered of Input Signal by Chebyshev-I Filter');
xlabel('Time(sec)');
ylabel('Magnitude');
subplot(2,1,2)
stem(range_of_frequency,fftshift_of_fx,'-*');
title('Magnitude Response of Filtered Signal by Chebyshev-I Filter');
xlabel('Frequency(Hz)');
ylabel('Magnitude');
gtext('Ziya Tuna Bölükbaşı');
%chebs2
[y,x]=cheby2(4,20,[740,780]/(Fs/2)); %(order,Astop,Fc/fs/2)
filtered_xinput=filter(y,x,x_input);
length_x=length(x_input);
time=[1:length_x]/fs;
figure;
subplot(2,1,1)
plot(time,filtered_xinput);
fft_of_fx=fft(filtered_xinput)
fftshift_of_fx=fftshift(abs(fft_of_fx));
length_fftfx=length(fftshift_of_fx);
length_x=length(x_input);
range_of_frequency=-fs/2:fs/length_fftfx:fs/2-fs/length_fftfx;
title('Filtered of Input Signal by Chebyshev-II Filter');
xlabel('Time(sec)');
ylabel('Magnitude');
subplot(2,1,2)
stem(range_of_frequency,fftshift_of_fx,'-*');
title('Magnitude Response of Filtered Signal by Chebyshev-II Filter');
xlabel('Frequency(Hz)');
ylabel('Magnitude');
gtext('Ziya Tuna Bölükbaşı');
```









Frequency of the output signal is 762.3 Hz. Butterworth Bandpass and Chebyshev-I Bandpass outputs are almost same. But Chebyshev-II Bandpass output has some ripples.

9) I can say that chebyshev-I filter does not provide meaningful filtering compared to other filters, butterworth filter provides meaningful filtering in all cases and chebyshev-II filter is more suitable for bandpass filter. I think chebyshev-II filter is the best filter.

10)Matlab Code

```
clc;
clear all;
close all;
[x input,fs]=audioread('C:\Users\Tuna\OneDrive - hacettepe.edu.tr\Masaüstü\dersler\2022 1.dönem\dsplab\pre-
Fs=11025;
%x input
sound(x input)
length_x=length(x_input);
time=[1:length_x]/fs;
figure;
plot(time,x_input);
title('Original x input Signal');
xlabel('Time(sec)');
ylabel('Magnitude');
fft of x=fft(x input)
fftshift_of_x=fftshift(abs(fft_of_x));
length_fftx=length(fftshift_of_x);
range_of_frequency=-fs/2:fs/length_fftx:fs/2-fs/length_fftx;
plot(range_of_frequency,fftshift_of_x,'-*');
title(' Spectrum of Magnitude of Input Signal');
xlabel('Frequency(Hz)')
ylabel('Magnitude')
gtext('Ziya Tuna Bölükbaşı')
%butterworth
Fc=2000
[y,x] = butter(11,Fc/(Fs/2));
filtered_xinput=filter(y,x,x_input);
figure:
subplot (2,1,1)
plot(time, filtered_xinput);
fft_of_fx=fft(filtered_xinput)
fttshift_of_fx=fftshift(abs(fft_of_fx));
length_fftfx=length(fftshift_of_fx);
range_of_frequency=-fs/2:fs/length_fftfx:fs/2-fs/length_fftfx;
title('Filtered of Input Signal by Butterworth Filter');
xlabel('Time(sec)');
ylabel('Magnitude');
subplot(2,1,2)
{\tt stem} \, ({\tt range\_of\_frequency}, {\tt fftshift\_of\_fx}, {\tt '-*'}) \, ;
title('Magnitude Response of Filtered Signal by Butterworth Filter');
xlabel('Frequency(Hz)');
ylabel('Magnitude');
gtext('Ziya Tuna Bölükbaşı');
sound(filtered_xinput)
%chebvshev-I
figure;
subplot(2,1,1)
Fc=2000;
Fs=11025;
[y,x] = cheby1(6,0.1,Fc/(Fs/2));
filtered_xinput=filter(y,x,x_input);
plot(time,filtered_xinput);
title('Filtered of Input Signal by Chebyshev Type-I Filter');
xlabel('Time(sec)');
ylabel('Magnitude');
fft of fx=fft(filtered xinput)
fftshift_of_fx=fftshift(abs(fft_of_fx));
length_fftfx=length(fftshift_of_fx);
range_of_frequency=-fs/2:fs/length_fftfx:fs/2-fs/length_fftfx;
subplot(2,1,2)
subplot(2,1,2,)
plot(range_of_frequency,fftshift_of_fx,'-*');
title('Magnitude Response of Filtered Signal by Chebyshev Type-I Filter');
xlabel('Frequency(Hz)');
ylabel('Magnitude');
gtext('Ziya Tuna Bölükbaşı');
sound(filtered_xinput)
%chebyshev-II
figure:
subplot(2,1,1)
Fc=2000;
Fs=11025;
[y,x] = cheby2(6,20,Fc/(Fs/2));
filtered_xinput=filter(y,x,x_input);
plot(time, filtered_xinput);
fft_of_fx=fft(filtered_xinput)
fftshift_of_fx=fftshift(abs(fft_of_fx));
length_fftfx=length(fftshift_of_fx);
range_of_frequency=-fs/2:fs/length_fftfx:fs/2-fs/length_fftfx;
title('Filtered of Input Signal by Chebyshev Type-II Filter');
xlabel('Time(sec)');
ylabel('Magnitude');
subplot(2,1,2)
plot(range_of_frequency,fftshift_of_fx,'-*');
title('Magnitude Response of Filtered Signal by Chebyshev Type-II Filter');
xlabel('Frequency(Hz)');
ylabel('Magnitude');
gtext('Ziya Tuna Bölükbaşı');
sound(filtered_xinput)
```

I filtered the original signal with three of this filter. I listened both the original signal and other filtered signals. As for me the more understandable and the cleanest signal I listened between three signals is the signal filtered by Chebyshev-II Filter.