

Ahsanullah University of Science and Technology

PROJECT REPORT

Course No: ***EEE3218***

Course Name: ***Digital Signal Processing Laboratory***

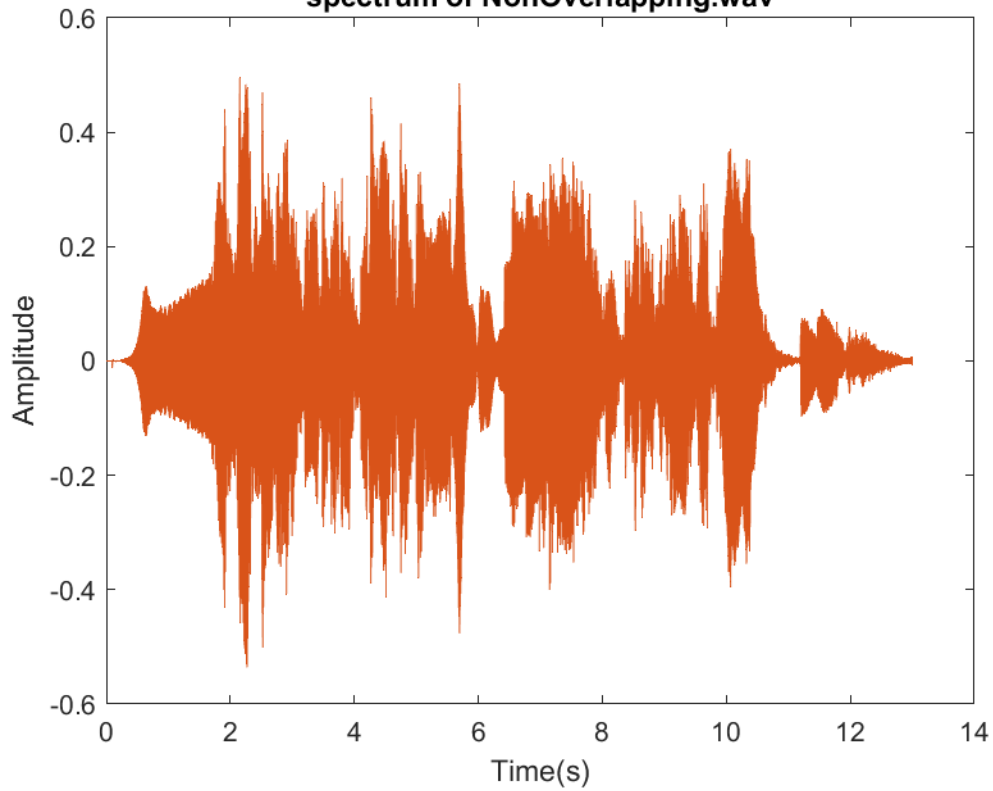
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 - ID: ***190205121***
- Year: 3 , Semester: 2
 - Section: B-2
- Department of ***EEE***

TASK 01

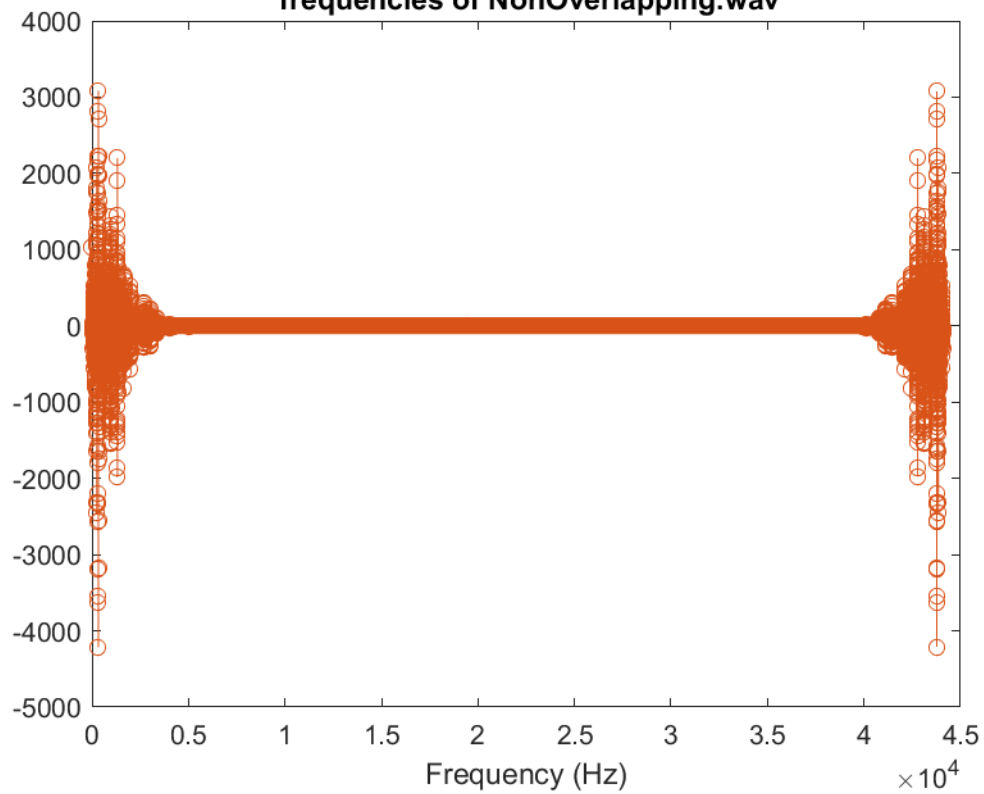
In question said that ‘NonOverlapping.wav’ contains the non-overlapping **modified** sounds of the musical instruments – *guitar, piano, trumpet and violin* (in this order)

For guitar

spectrum of NonOverlapping.wav



frequencies of NonOverlapping.wav

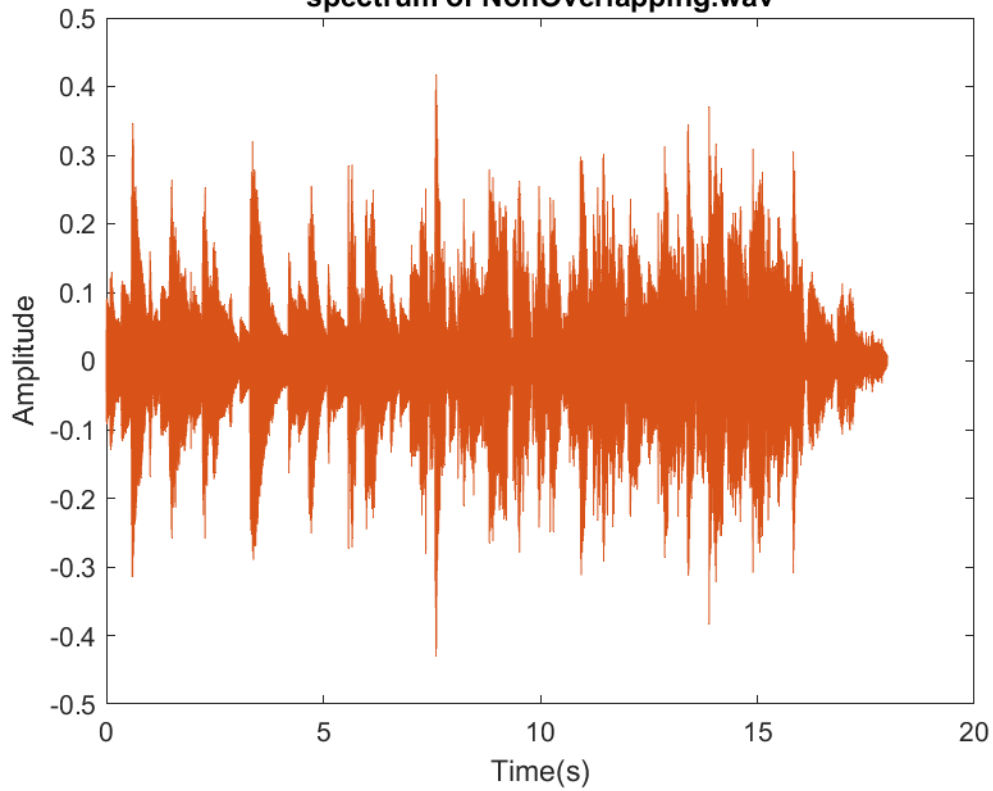


Frequency:0-2000(about)

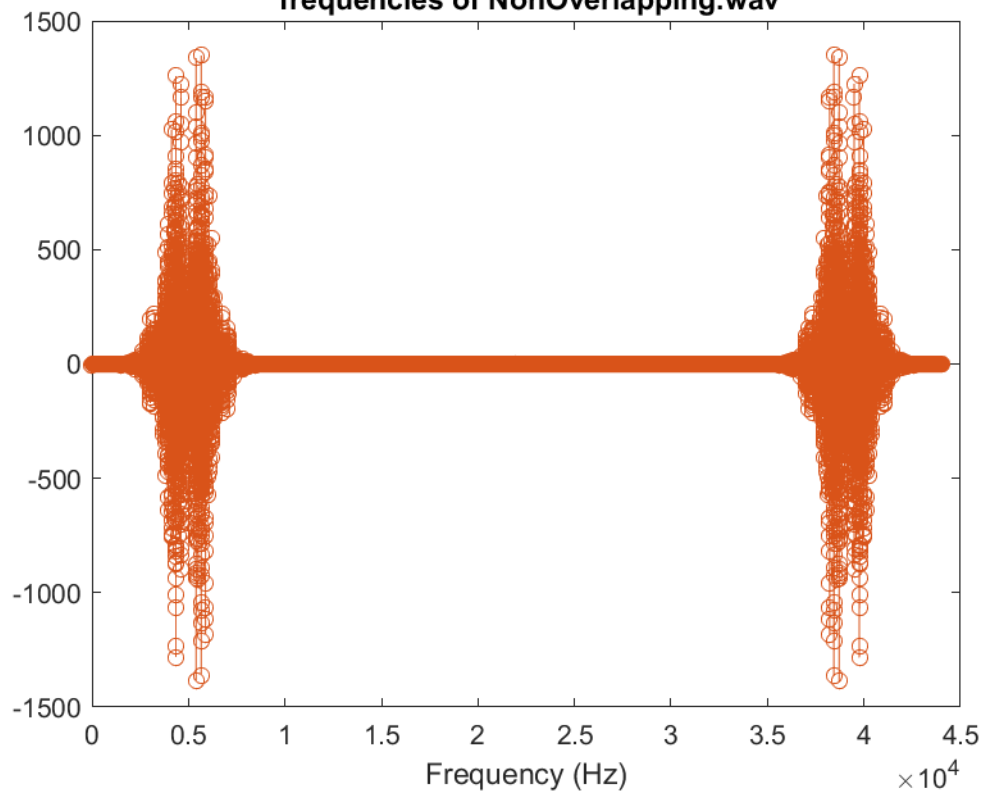
```
clc ;
clear all;
close all;
samples = [1,13*44100];%range for
[es1,Fs1]=audioread('NonOverlapping.wav',samples);
N = length(es1); % sample lenth
slength = N/Fs1; % total time span of audio signal
t = linspace(0, N/Fs1, N);
n=0:length(es1)-1;
% time domain plot
figure(1)
plot(t, es1); % plots the audio
title('spectrum of 'NonOverlapping.wav');
xlabel('Time(s) ');
ylabel('Amplitude');
%find the frequencies of the signal
es2=fft(es1);
N=length(es2);
figure(2)
f=(n*Fs1)/N;
stem(f,es2);
xlabel('Frequency (Hz) ');
title(' frequencies of NonOverlapping.wav');
```

For piano

spectrum of NonOverlapping.wav



frequencies of NonOverlapping.wav

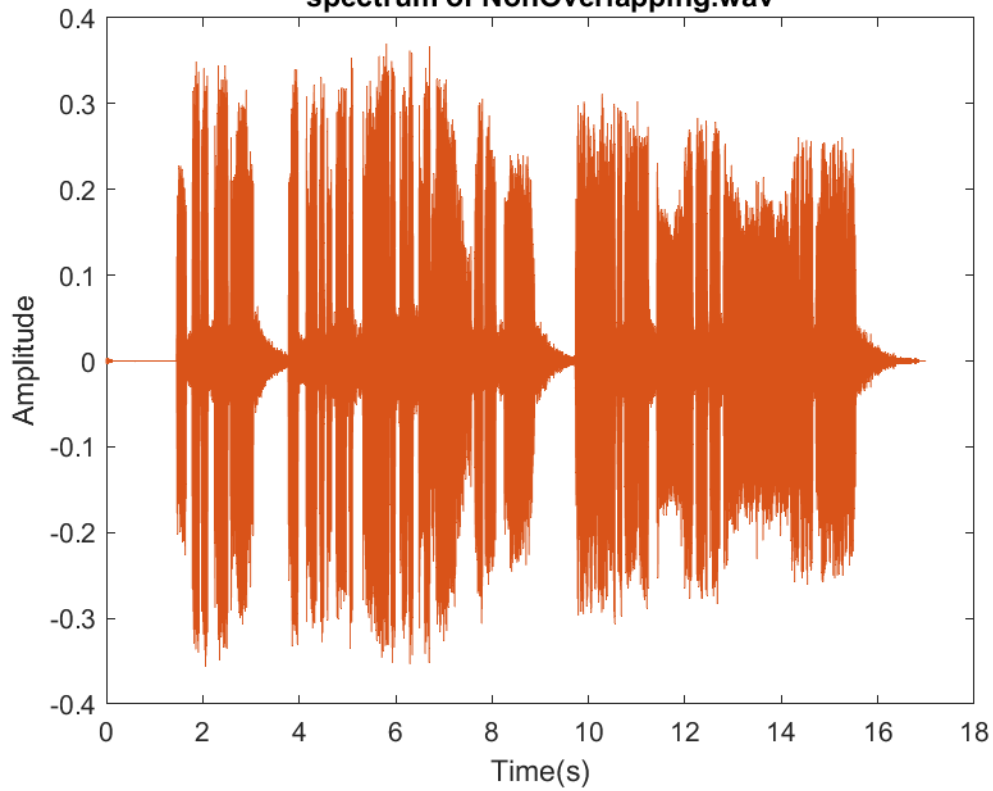


Frequency:2100-5000(about)

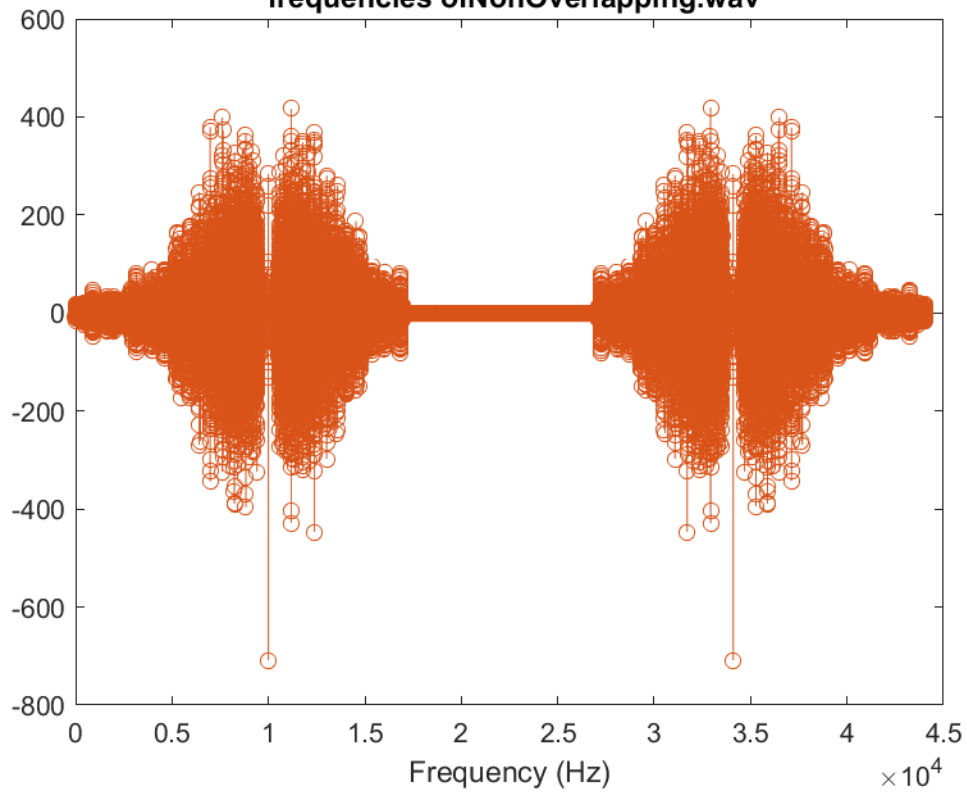
```
clc ;
clear all;
close all;
samples = [14*44100,32*44100];%range for
[es1,Fs1]=audioread('NonOverlapping.wav',samples);
N = length(es1); % sample lenth
slength = N/Fs1; % total time span of audio signal
t = linspace(0, N/Fs1, N);
n=0:length(es1)-1;
% time domain plot
figure(1)
plot(t, es1); % plots the audio
title('spectrum of NonOverlapping.wav');
xlabel('Time(s)');
ylabel('Amplitude');
%find the frequencies of the signal
es2=fft(es1);
N=length(es2);
figure(2)
f=(n*Fs1)/N;
stem(f,es2);
xlabel('Frequency (Hz)');
title(' frequencies of NonOverlapping.wav');
```

For trumpet

spectrum of NonOverlapping.wav



frequencies of NonOverlapping.wav

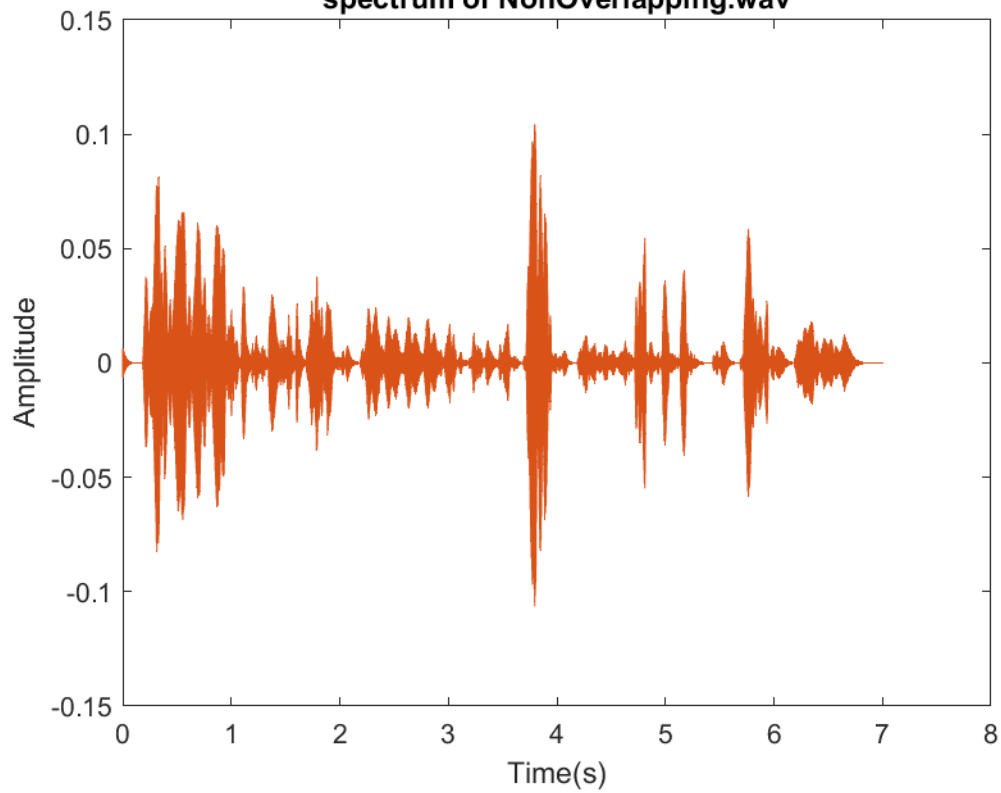


Frequency:5000-10000(about)

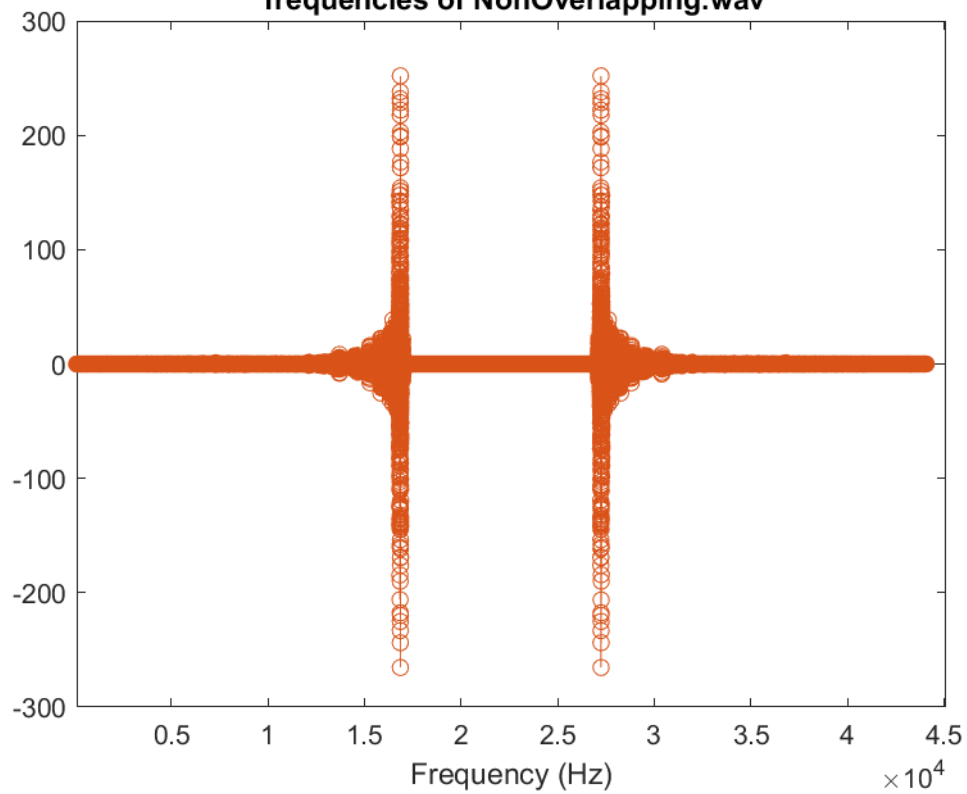
```
clc ;
clear all;
close all;
samples = [33*44100,50*44100];%range for
[es1,Fs1]=audioread('NonOverlapping.wav',samples);
N = length(es1); % sample lenth
slength = N/Fs1; % total time span of audio signal
t = linspace(0, N/Fs1, N);
n=0:length(es1)-1;
% time domain plot
figure(1)
plot(t, es1); % plots the audio
title('spectrum of NonOverlapping.wav');
xlabel('Time(s) ');
ylabel('Amplitude');
%find the frequencies of the signal
es2=fft(es1);
N=length(es2);
figure(2)
f=(n*Fs1)/N;
stem(f,es2);
xlabel('Frequency (Hz) ');
title(' frequencies of NonOverlapping.wav');
```


For violin

spectrum of NonOverlapping.wav



frequencies of NonOverlapping.wav

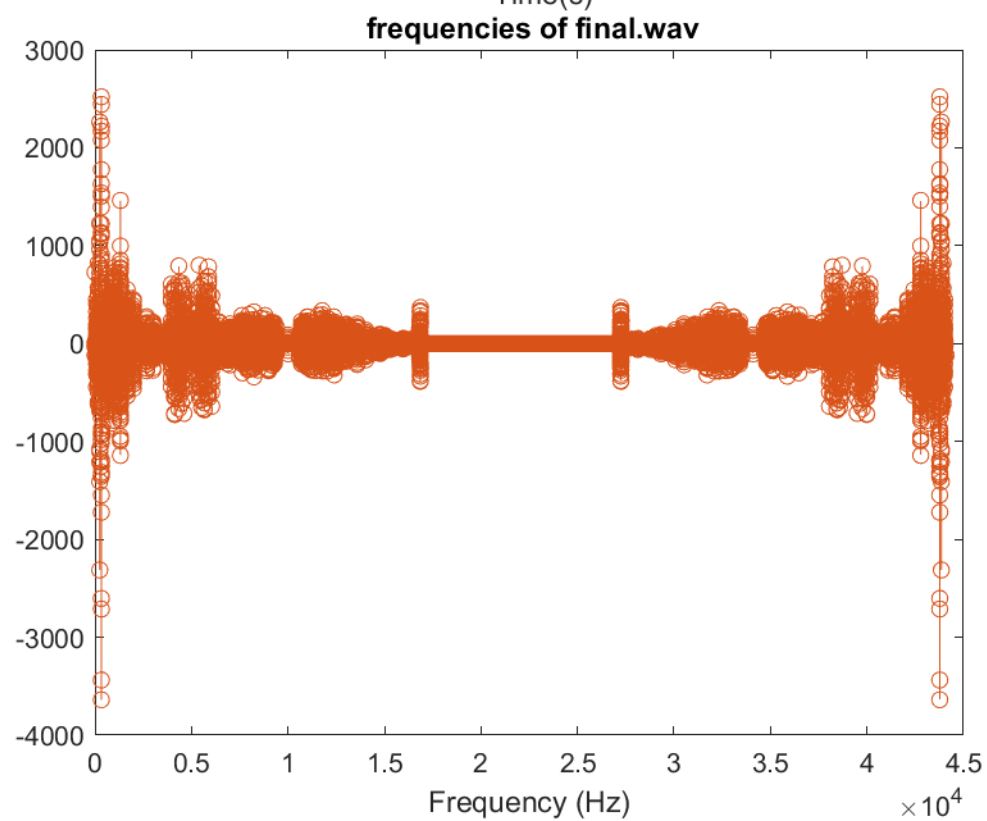
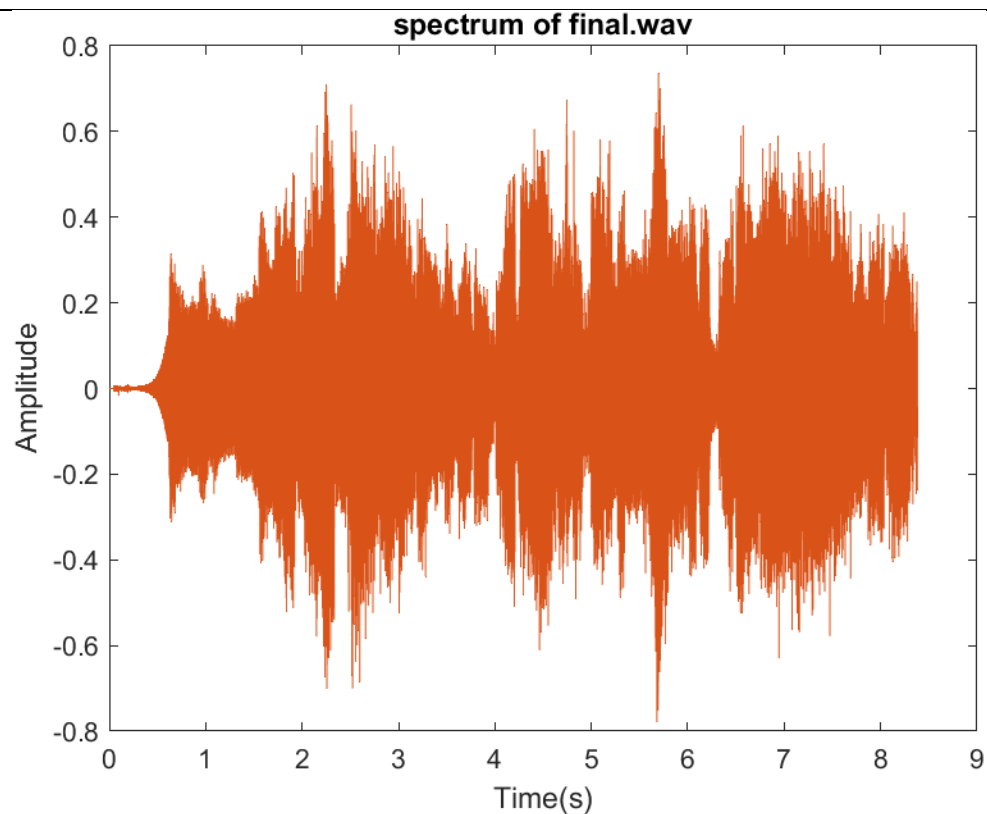


Frequency:16500-17000(about)

```
clc ;
clear all;
close all;
samples = [51*44100,58*44100];%range for
[es1,Fs1]=audioread('NonOverlapping.wav',samples);
N = length(es1); % sample lenth
slength = N/Fs1; % total time span of audio signal
t = linspace(0, N/Fs1, N);
n=0:length(es1)-1;
% time domain plot
figure(1)
plot(t, es1); % plots the audio
title('spectrum of NonOverlapping.wav');
xlabel('Time(s)');
ylabel('Amplitude');
%find the frequencies of the signal
es2=fft(es1);
N=length(es2);
figure(2)
f=(n*Fs1)/N;
stem(f,es2);
xlabel('Frequency (Hz)');
title(' frequencies of NonOverlapping.wav');
```

TASK 02

```
clc ;
clear all;
close all;
%samples = [51*44100,58*44100];%range for
[es1,Fs1]=audioread('final.wav');
N = length(es1); % sample length
slength = N/Fs1; % total time span of audio signal
t = linspace(0, N/Fs1, N);
n=0:length(es1)-1;
% time domain plot
figure(1)
plot(t, es1); % plots the audio
title('spectrum of final.wav');
xlabel('Time(s)');
ylabel('Amplitude');
%find the frequencies of the signal
es2=fft(es1);
N=length(es2);
figure(2)
f=(n*Fs1)/N;
stem(f,es2);
xlabel('Frequency (Hz)');
title(' frequencies of final.wav');
```



Frequency Range 00hz to 17k

TASK 03

Filter no 1

Low pass FIR filter

Method: Window(Hamming)

Order:1000

Fc:2000;

Filter no 2

Band pass FIR filter

Method: Window(Blackman)

Order:2426

Fc:2000-5000;

Filter no 3

Band pass FIR filter

Method: Window(Blackman)

Order:2426

Fc:5000-10000;

Filter no 4

Band pass FIR filter

Method: Window(blackman)

Order:2426

Fc:14000-16000;

TASK 04

```
clc;
clear all;
close all;
[es1,Fs1]=audioread('final.wav');
%% low pass FIR filter
TW = 0.11;
PBE = 2000;
M1=1000;
corner = PBE+TW/2;
wn = 2*corner/Fs1;
a1 = fir1(M1, wn, 'low', hamming(M1+1));
%% Filtering the Audio Data
filtered_audio_data_1 = filter(a1,1,es1) ;
audiowrite('output1.wav',filtered_audio_data_1 ,Fs1);
%% FIR Bandpass filter 1
Fs = Fs1;
TW = 100;
PBE1 = 2000;
PBE2 = 5000;
delf = TW/Fs;
M2 = round(5.5/delf);
corner1 = PBE1+TW/2;
corner2 = PBE2+TW/2;
wn1 = 2*corner1/Fs;
wn2 = 2*corner2/Fs;
wn=[wn1,wn2];
a2= fir1(M2, wn, 'bandpass', blackman(M2+1));
%% Filtering the Audio Data
filtered_audio_data_2 = filter(a2,1,es1) ;
audiowrite('output2.wav',filtered_audio_data_2,Fs1);
%% FIR Band pass filter 2
TW3 = 100;
PBE31 = 5000;
PBE32 = 10000;
delf3 = TW3/Fs;
M3 = round(5.5/delf3);
corner31 = PBE31+TW3/2;
corner32 = PBE32+TW3/2;
```

```

wn31 = 2*corner31/Fs;
wn32 = 2*corner32/Fs;
wn3=[wn31,wn32];
a3= fir1(M3, wn3, 'bandpass', blackman(M3+1));
%% Filtering the Audio Data
filtered_audio_data_3 = filter(a3,1,es1) ;
audiowrite('output3.wav',filtered_audio_data_3,Fs1);
%% FIR Band pass filter 3
TW4 = 100;
PBE41 = 14000;
PBE42 = 16000;
delf4 = TW3/Fs;
M4 = round(5.5/delf4);
corner41 = PBE41+TW4/2;
corner42 = PBE42+TW4/2;
wn41 = 2*corner41/Fs;
wn42 = 2*corner42/Fs;
wn4=[wn41,wn42];
a4= fir1(M4, wn4, 'bandpass', blackman(M4+1));
%% Filtering the Audio Data
filtered_audio_data_4 = filter(a4,1,es1) ;
audiowrite('output4.wav',filtered_audio_data_4,Fs1);

```

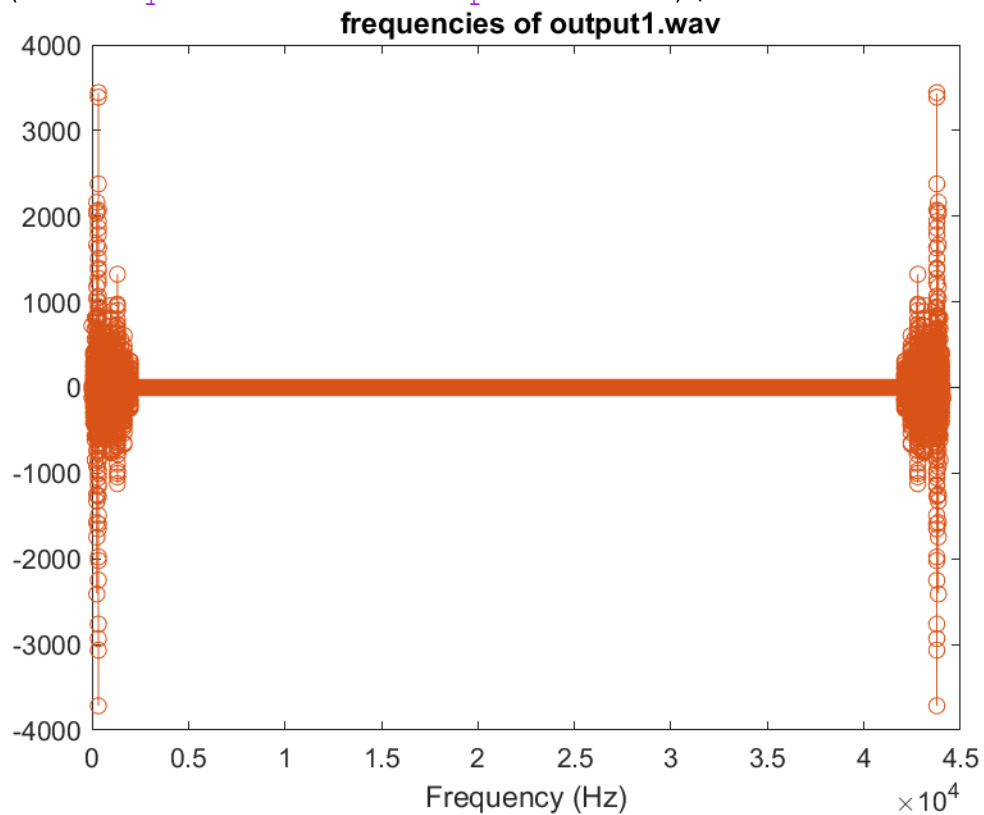
TASK05

```

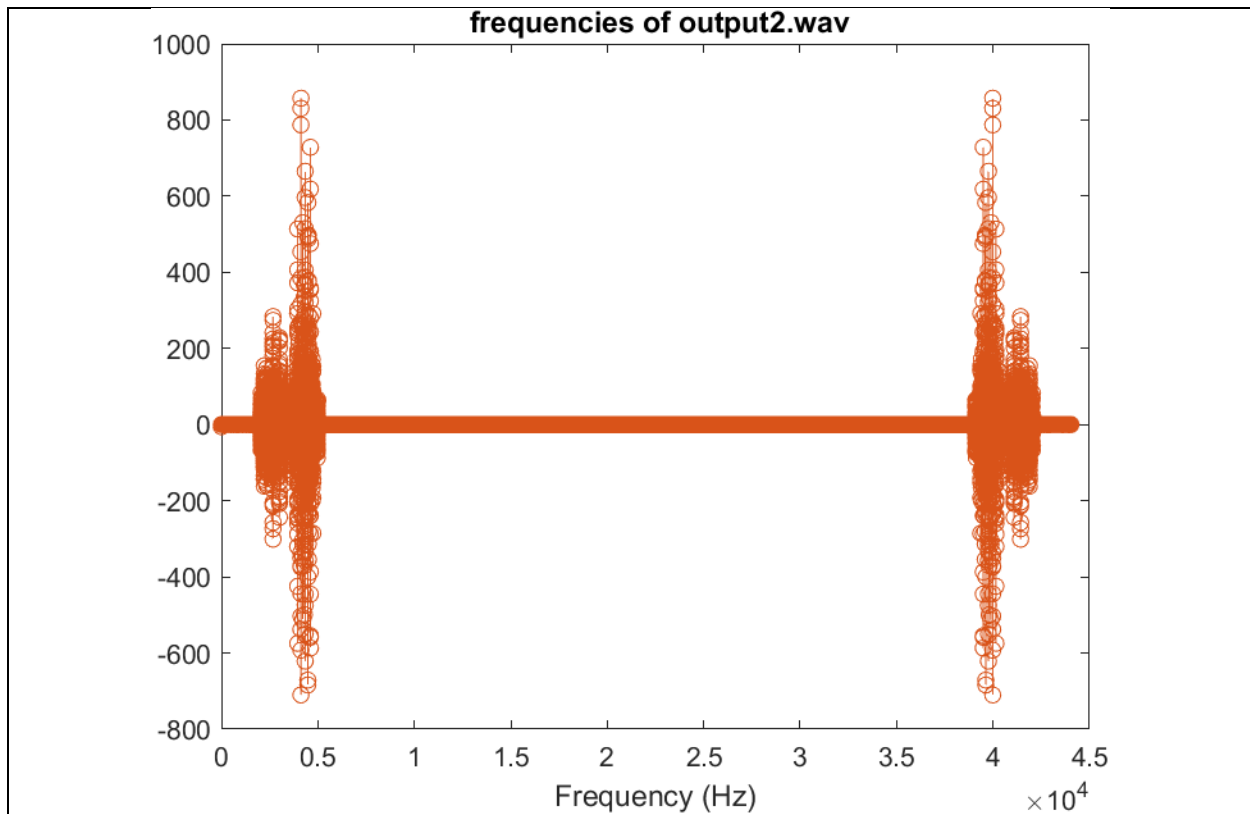
clc;
close all;
clear all;
[es1,Fs1]=audioread('output1.wav');
Ts=1/Fs1;
N = length(es1); % sample lenth
slength = N/Fs1; % total time span of audio signal
t = linspace(0, N/Fs1, N);
n=0:length(es1)-1;
%frequencies Plot
es2=fft(es1);
N=length(es2);
figure(2)
f=(n*Fs1)/N;

```

```
stem(f,es2);
xlabel('Frequency (Hz)');
title(' frequencies of output1.wav');
```



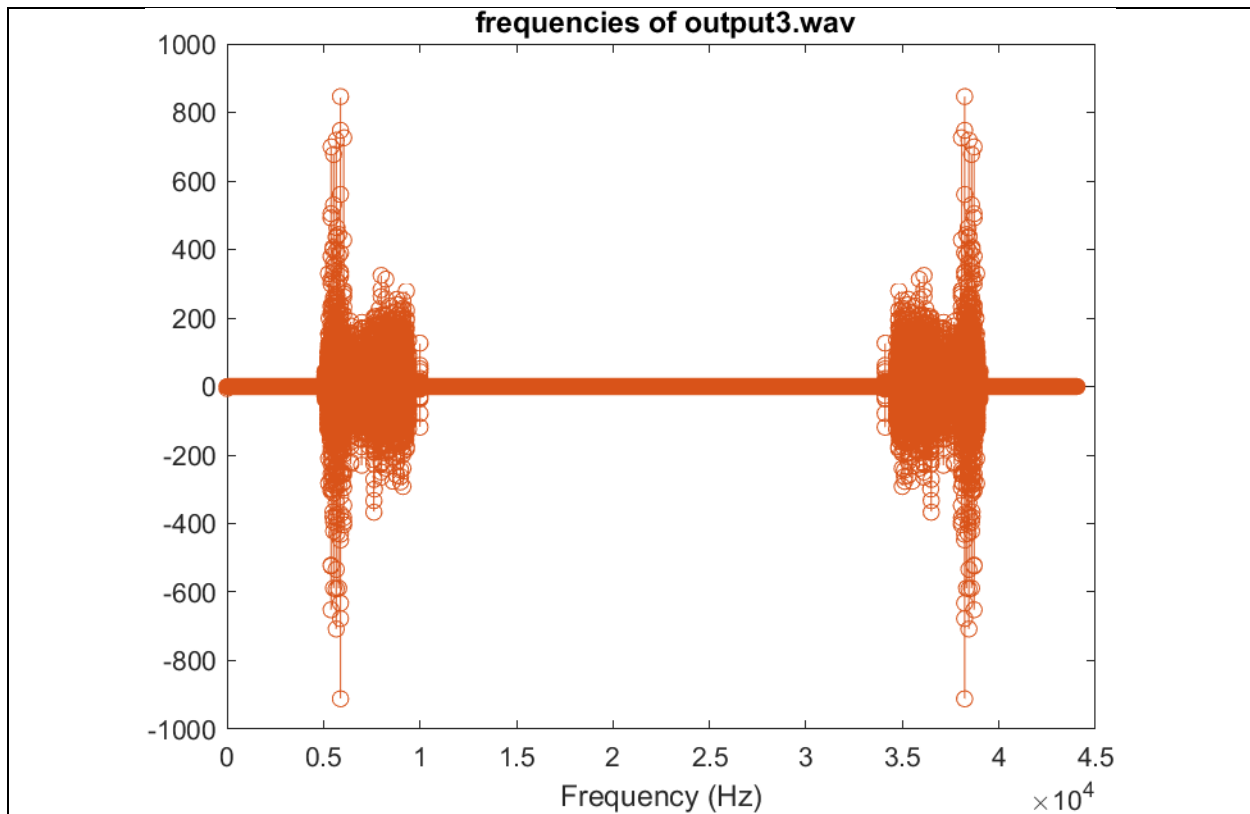
```
clc;
close all;
clear all;
[es1,Fs1]=audioread('output2.wav');
Ts=1/Fs1;
N = length(es1); % sample length
slength = N/Fs1; % total time span of audio signal
t = linspace(0, N/Fs1, N);
n=0:length(es1)-1;
%frequencies Plot
es2=fft(es1);
N=length(es2);
figure(2)
f=(n*Fs1)/N;
stem(f,es2);
xlabel('Frequency (Hz)');
title(' frequencies of output2.wav');
```

```

clc;
close all;
clear all;
[es1,Fs1]=audioread('output3.wav');
Ts=1/Fs1;
N = length(es1); % sample lenth
slength = N/Fs1; % total time span of audio signal
t = linspace(0, N/Fs1, N);
n=0:length(es1)-1;
%frequencies Plot
es2=fft(es1);
N=length(es2);
figure(2)
f=(n*Fs1)/N;
stem(f,es2);
xlabel('Frequency (Hz)');
title(' frequencies of output3.wav');

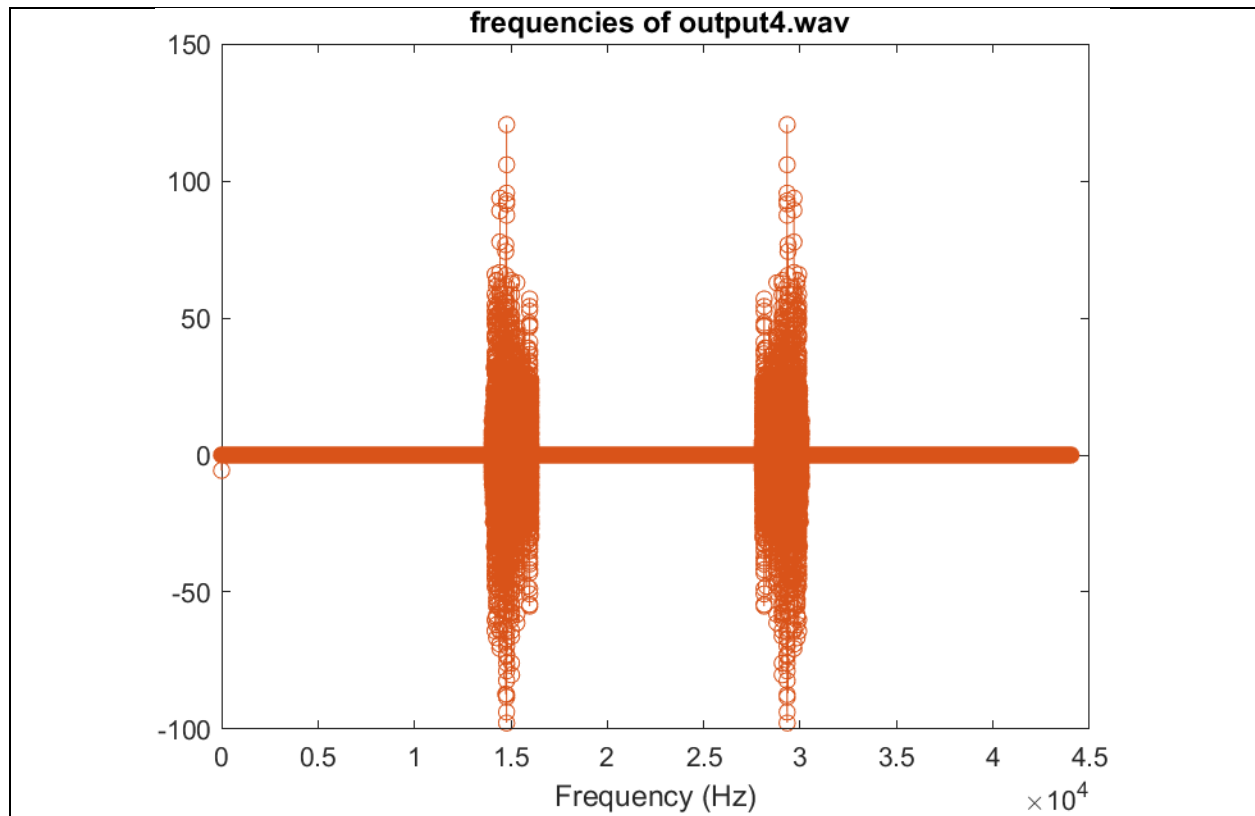
```



```

clc;
close all;
clear all;
[es1,Fs1]=audioread('output4.wav');
Ts=1/Fs1;
N = length(es1); % sample lenth
slength = N/Fs1; % total time span of audio signal
t = linspace(0, N/Fs1, N);
n=0:length(es1)-1;
%frequencies Plot
es2=fft(es1);
N=length(es2);
figure(2)
f=(n*Fs1)/N;
stem(f,es2);
xlabel('Frequency (Hz)');
title(' frequencies of output4.wav');

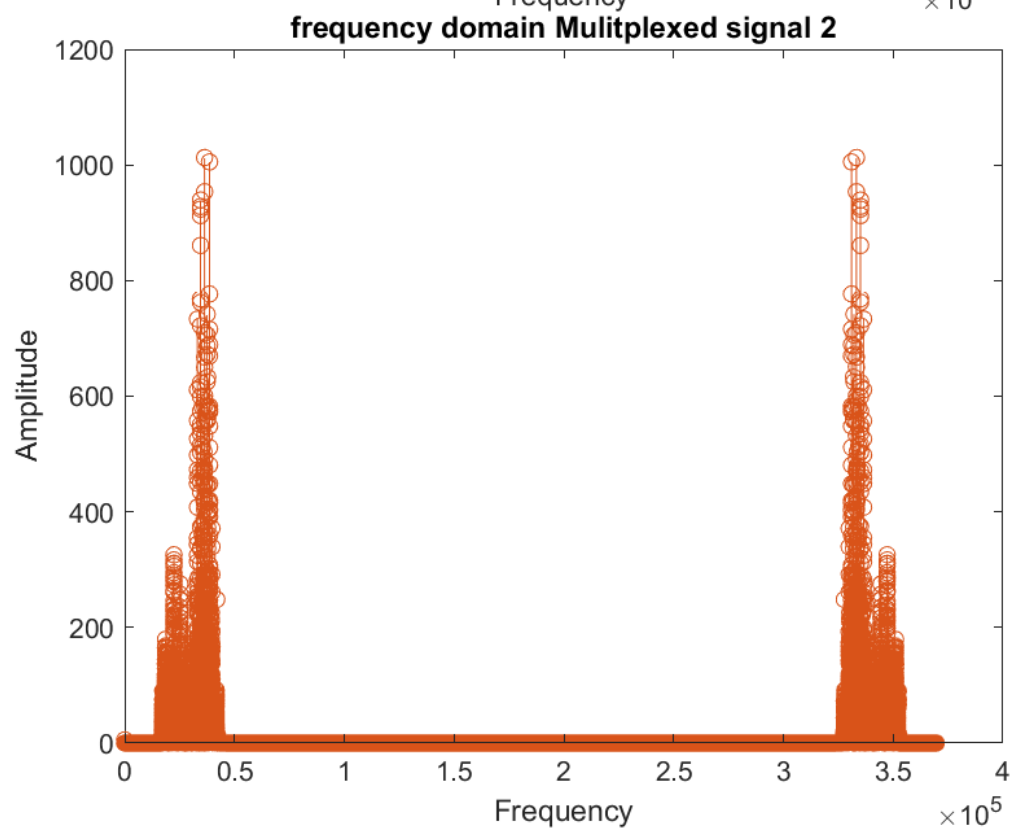
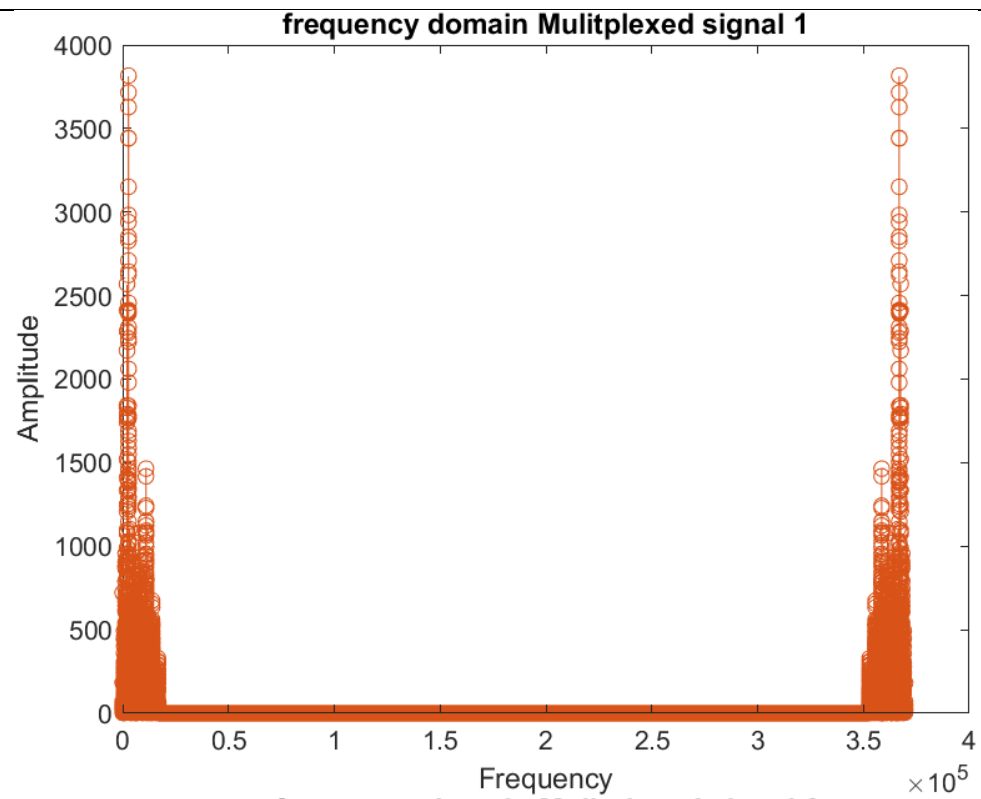
```

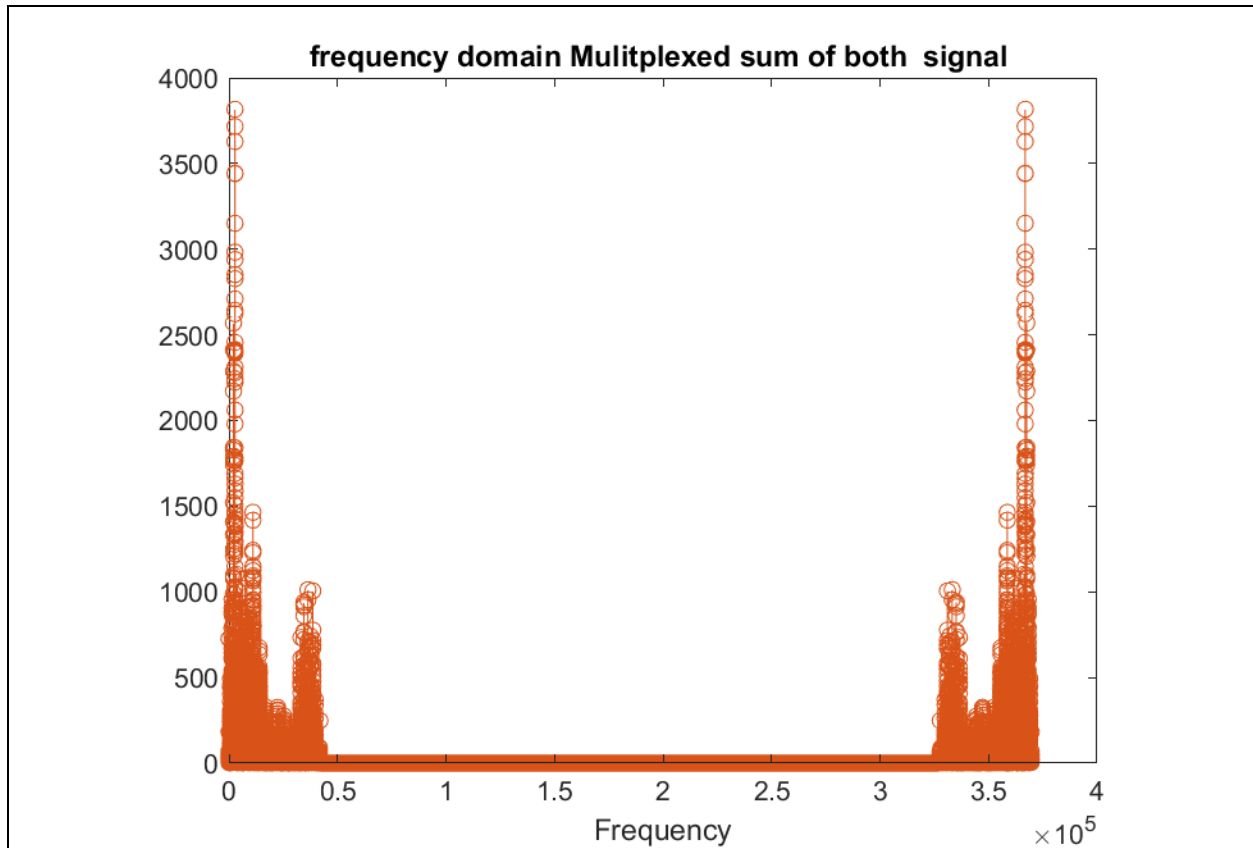


Pass the individual wav files separately using “Time-division multiplexing (TDM)”.

TASK06

```
clc;
clear all;
close all;
%% FDM multiplexer
fc1=5121;
[m,fs1]=audioread('output1.wav');
[n,fs2]=audioread('output2.wav');
t1=[1:length(m)]';
ct1=cos(2*pi*fc1*t1);
Xfdm1 = m.*ct1;
Xfdm_fft1=abs(fft(Xfdm1));
figure
stem(Xfdm_fft1);
title('frequency domain Mulitplexed signal 1');
xlabel('Frequency');
ylabel('Amplitude');
fc2=6000; %chosse another carrier
t2=[1:length(n)]';
ct2=cos(2*pi*fc2*t2);
Xfdm2 = n.*ct2;
Xfdm_fft2=abs(fft(Xfdm2));
figure
stem(Xfdm_fft2);
title('frequency domain Mulitplexed signal 2');
xlabel('Frequency');
ylabel('Amplitude');
figure
stem(Xfdm_fft1+Xfdm_fft2);
title('frequency domain Mulitplexed sum of both
signal ');
xlabel('Frequency');
```





TASK07

Convert this into an overlapping yet melodious one with proper synchronization of octaves.

```
clc;
close all;
clear all;
[es1,Fs1]=audioread('output1.wav');
[es2,Fs2]=audioread('output2.wav');
[es3,Fs3]=audioread('output3.wav');
[es4,Fs4]=audioread('output4.wav');

final=es1+es2+es3+es4;
% convertthis into an overlapping yet melodious ...
... one with proper synchronization of octaves
audiowrite('mFinal.wav',final,Fs1);
sound(final,Fs1)
```

