

Ain Shams University
Faculty of Engineering
Discipline Programs



Signals Project Report

Computer Engineering and Software Systems (CESS)

Submitted to:

Dr. Michael

Eng. George

Submitted by:

Habiba Yasser AbdelHalim	20P3072
Nadine Hisham Hassan	20P9880
Salma Nasreldin aboelela ahmed hendawy	20p7105
Hamsa Ahmed Abdelmgeed	20p1874
Mahamad Hany Mahros Elshabory	20P9302

Table of Contents

1.0	Introduction	3
2.0	Step 1	3
3.0	Step 2	4
4.0	Step 3	4
5.0	Step 4	5
6.0	Step 5	5
7.0	Step 6	5
8.0	Step 7	6
9.0	Step 8	6
10.0	Step 9	6
11.0	Step 10	7
12.0	Step 11	7
13.0	Step 12	7
14.0	Step 13	8
15.0	Step 14	8
16.0	Step 15	8
17.0	Step 16	9
18.0	Step 17	9
19.0	Step 18	9
20.0	Step 19	10
21.0	Step 20	10
22.0	Step 21	10
23.0	Step 22	11
24.0	Step 23	11
25.0	Step 24	11
26.0	Step 25	12
27.0	Contributions	13

1.0 Introduction

This project aims to display how the music notes DO, RE, MI, FA sound originally, low pass filtered, and high pass filtered.

2.0 Step 1

First off, we calculated the different frequencies of each musical node to be able to find sampling frequency using the equation in the figure below.

```
1      n1=-9;
2      n2=-7;
3      n3=-5;
4      n4=-4;
5      F1= 440*(2^(n1/12));
6      F2= 440*(2^(n2/12));
7      F3= 440*(2^(n3/12));
8      F4= 440*(2^(n4/12));
9      %QUESTION 1
10     Fs= 10* max([F1,F2,F3,F4]);|
11     framesize= round(0.5*Fs);
12     T= 1/Fs;
13     t=(0:1:framesize-1)*T;
14     xt1=cos(2*pi*F1*t);
15     xt2=cos(2*pi*F2*t);
16     xt3=cos(2*pi*F3*t);
17     xt4=cos(2*pi*F4*t);
```

3.0 Step 2

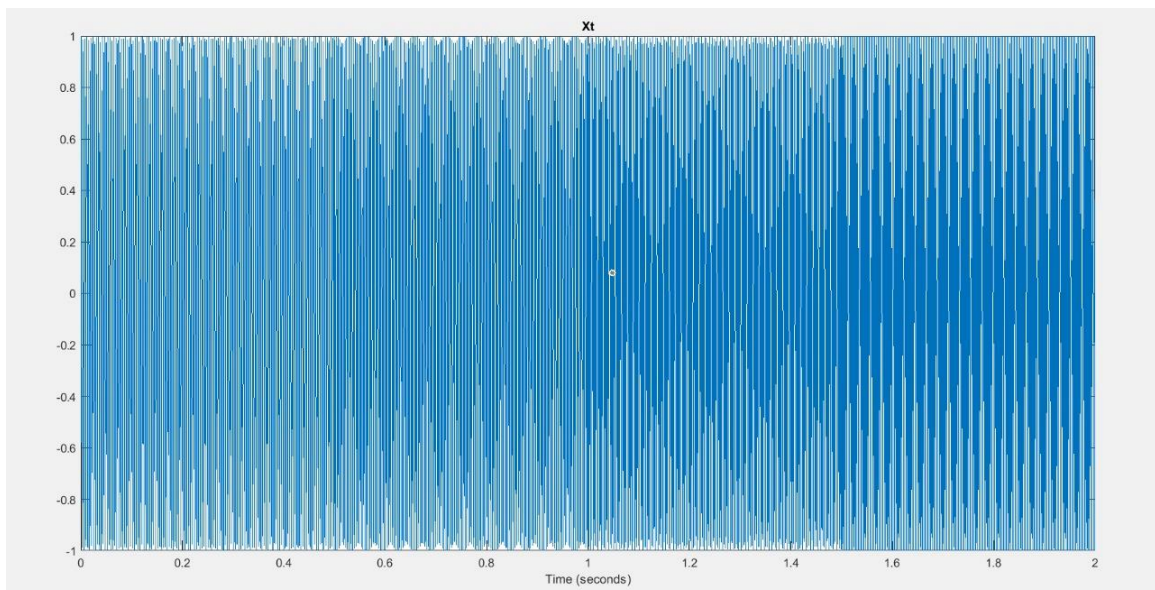
Then, we calculated $x(t)$ by combining the $x(t)$ of each musical node into one vector and generated its sound wave.

```
14      xt1=cos(2*pi*F1*t);
15      xt2=cos(2*pi*F2*t);
16      xt3=cos(2*pi*F3*t);
17      xt4=cos(2*pi*F4*t);
18      xt= [xt1 xt2 xt3 xt4];
19      framesize= framesize *4;
20      t=(0:1:framesize-1)*T;
21      %QUESTION2
22      filename= 'Xt.wav';
23      audiowrite('Xt.wav',xt, fix(Fs));
24      sound(xt,Fs);
```

4.0 Step 3

Then, we plotted signal $x(t)$ versus time (t) as shown in the figure below

```
25      %QUESTION3: Plot the Xt:
26      figure;
27      plot(t,xt);
28      xlabel('Time (seconds)');
29      title('Xt');
```



5.0 Step 4

Then, we computed the energy of the signal $x(t)$ using the equation below.

```
30 %QUESTION4
31 XEnergy= sum(abs(xt.^2))*T;
```

6.0 Step 5

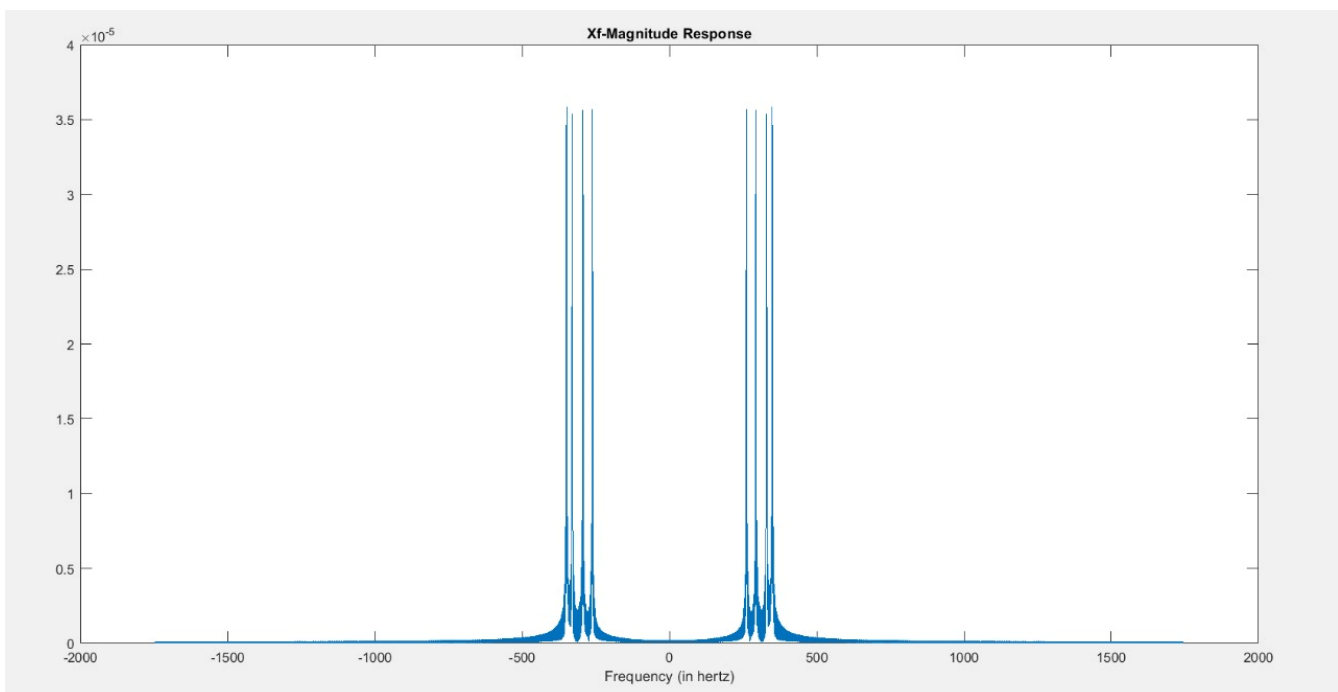
Then, we computed the frequency spectrum $X(f)$ of the signal using the equations below.

```
32 %QUESTION5
33 Xf= fftshift(fft(xt,framesize))*T;
34 dF = Fs/framesize;
35 f = -Fs/2:dF:Fs/2-dF;
```

7.0 Step 6

Then we plotted the magnitude of $X(f)$ in the frequency range $fs/2 \leq f \leq fs/2$, where fs is the sampling frequency as shown in the figure below.

```
36 %Question6: Plot the Xf spectrum:
37 figure;
38 plot(f,abs(Xf)/framesize);
39 xlabel('Frequency (in hertz)');
40 title('Xf-Magnitude Response');
```



8.0 Step 7

Then we computed the Energy of the signal $x(t)$ from its frequency spectrum $X(f)$, and verified Parseval's Theorem using the equation below.

```
41  
42
```

```
%QUESTION7
```

```
XEnergy2=sum(abs(Xf.^2))*dF;
```

9.0 Step 8

Then we designed a Butterworth low-pass filter with filter order 20 such that when the signal $x(t)$ is applied to this filter, the output does NOT contain the MI and FA musical nodes. Moreover, we calculated the cut-off frequency by calculating the average of RE and MI musical nodes together.

```
43  
44  
45  
46
```

```
%QUESTION8
```

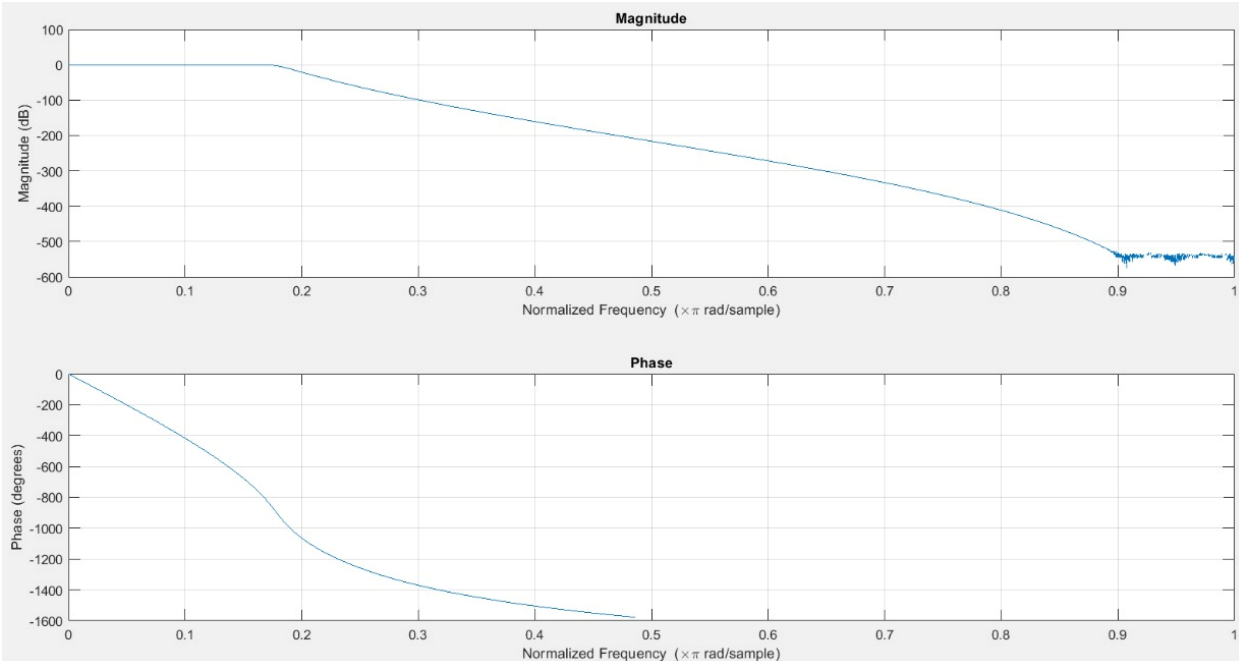
```
Fc= (F2+F3)/2;
```

```
nF= Fc/(Fs/2);
```

```
[b,a] = butter(20,nF,"low");
```

10.0 Step 9

Then we plotted the magnitude and phase response of the Butterworth low pass filter as shown in the figure below.



11.0 Step 10

Then, we applied the signal $x(t)$ to this Butterworth LPF and denoted the output signal as $y_1(t)$

```
50 %QUESTION10
51 Y1t= filter(b,a,xt);
```

12.0 Step 11

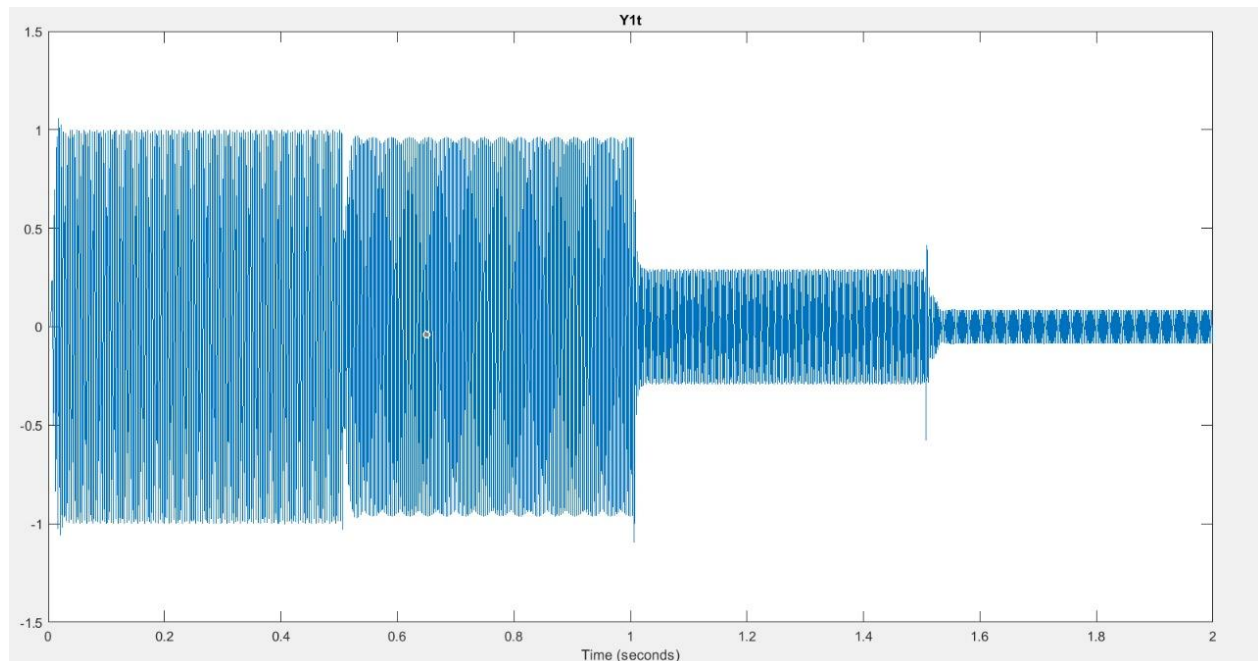
Then we stored the generated signal $y_1(t)$ as an audio file with extension (*.wav)

```
52 %QUESTION11
53 filename= 'Yt.wav';
54 audiowrite('Yt.wav',Y1t, fix(Fs));
55 sound(Y1t,Fs);
```

13.0 Step 12

Then we plotted the signal $y_1(t)$ versus time (t) as shown in the figure below.

```
56 %QUESTION12
57 figure;
58 plot(t,Y1t);
59 xlabel('Time (seconds)');
60 title('Y1t');
```



14.0 Step 13

Then we computed the energy of the signal $y_1(t)$ using the equation below.

```
61 %QUESTION13
62 Y1Energy= sum(abs(Y1t.^2))*T;
```

15.0 Step 14

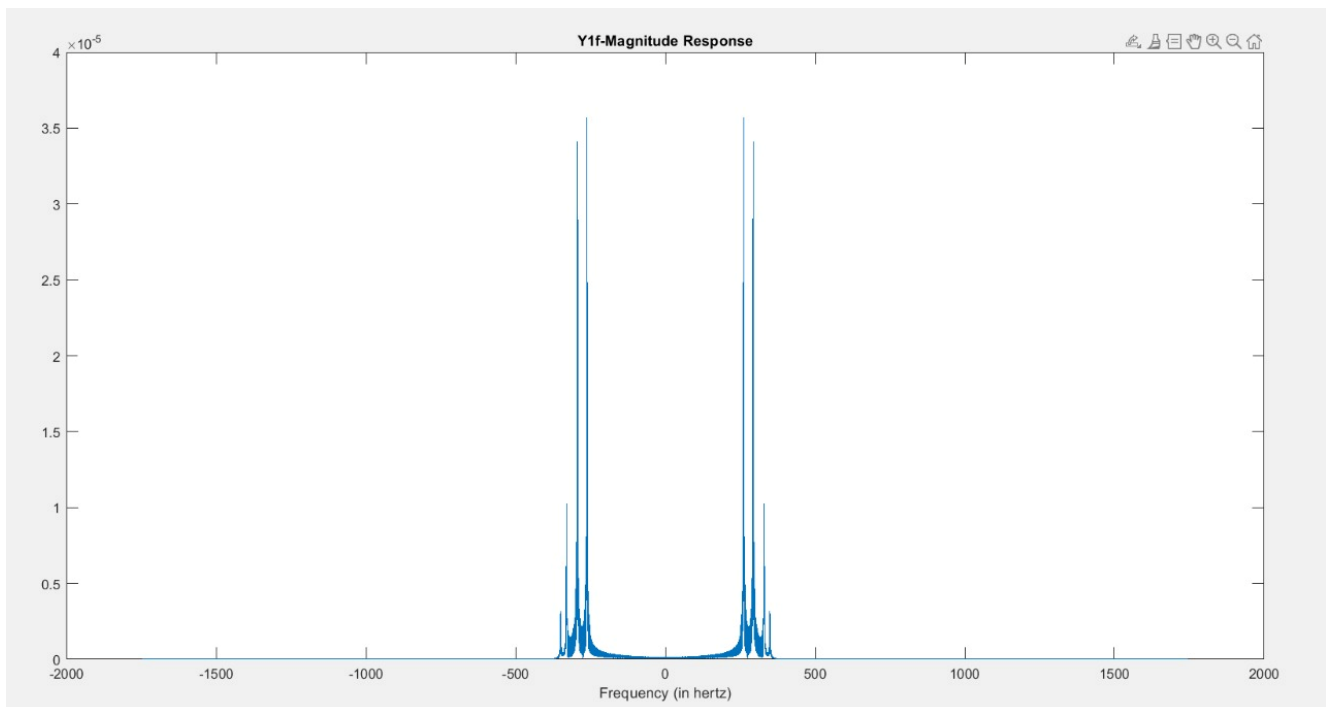
Then we computed the frequency spectrum $y_1(f)$ of this signal.

```
63 %QUESTION14
64 Y1f= fftshift(fft(Y1t,framesize))*T;
```

16.0 Step 15

Then we plotted the magnitude of $y_1(f)$ in the frequency range $fs/2 \leq f \leq fs/2$ as shown in the figure below

```
65 %QUESTION15
66 figure;
67 plot(f,abs(Y1f)/framesize);
68 xlabel('Frequency (in hertz)');
69 title('Y1f-Magnitude Response');
```



17.0 Step 16

Then we computed the Energy of the signal $y_1(t)$ from its frequency spectrum $y_1(f)$ and verified Parseval's Theorem using the equation below.

```
70 %QUESTION16
71 Y1Energy2=sum(abs(Y1f.^2))*dF;
```

18.0 Step 17

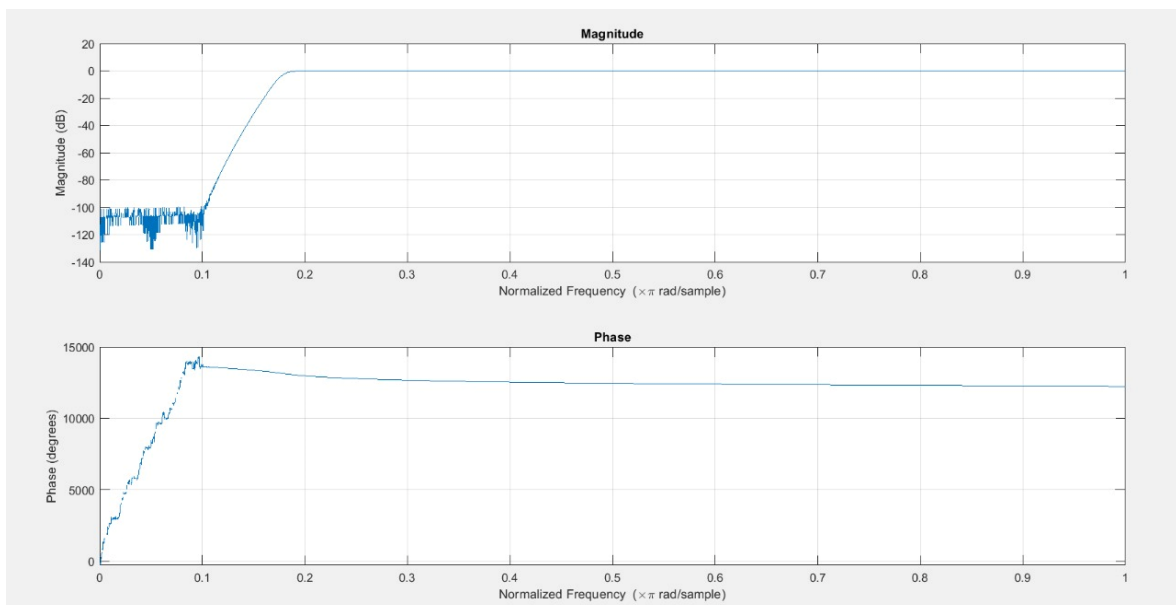
Then we designed a Butterworth high-pass filter with filter order 20 such that when the signal $x(t)$ is applied to this filter, the output does NOT contain the DO and RE musical nodes as shown in the figure below. Moreover, we calculated the cut-off frequency by calculating the average of RE and MI musical nodes together.

```
72 %QUESTION17
73 [d,c] = butter(20,nF,"high");
```

19.0 Step 18

Then we plotted the magnitude and phase response of the Butterworth HPF as shown in the figure below.

```
74 %QUESTION18
75 figure;
76 freqz(d, c, framesize);
```



20.0 Step 19

Then we applied the signal $x(t)$ to this Butterworth HPF and denoted the output as $y2(t)$.

```
77 %QUESTION19
78 Y2t= filter(d,c,xt);
```

21.0 Step 20

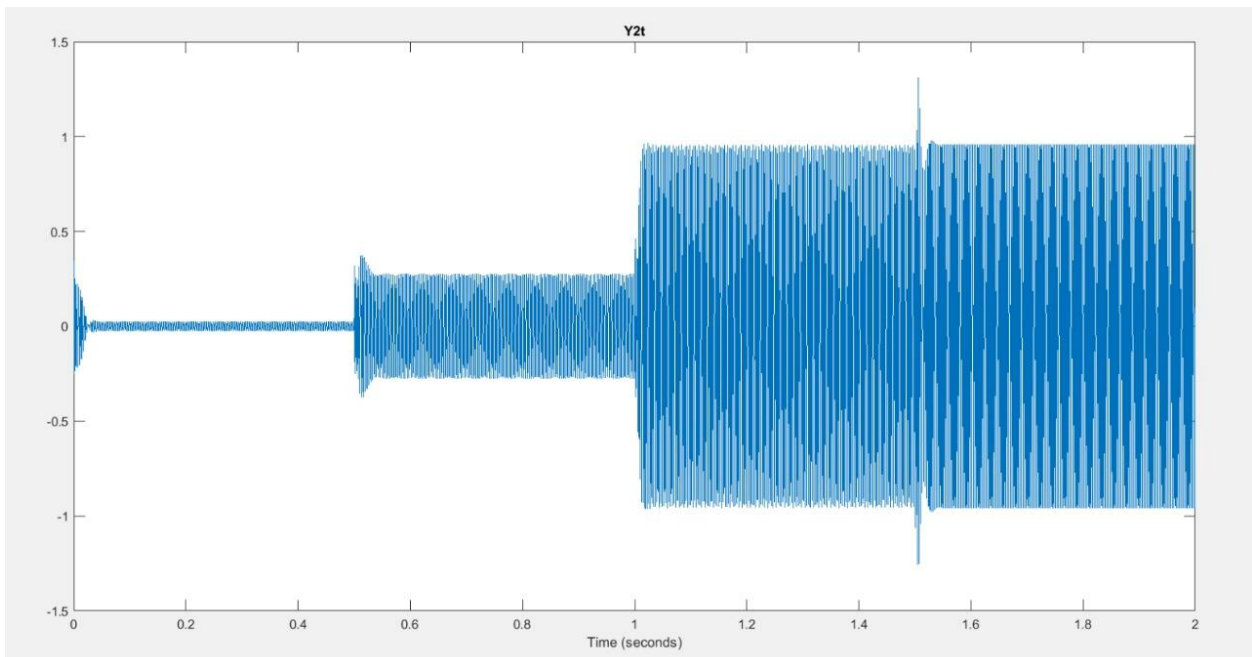
Then we stored the generated signal $y2(t)$ as an audio file with extension (*.wav).

```
79 %QUESTION20
80 filename= 'Y2t.wav';
81 audiowrite('Y2t.wav',Y2t, fix(Fs));
82 sound(Y2t,Fs);
```

22.0 Step 21

Then we plot the signal $y2(t)$ versus time (t) as shown in the figure below.

```
83 %QUESTION21
84 figure;
85 plot(t,Y2t);
86 xlabel('Time (seconds)');
87 title('Y2t');
```



23.0 Step 22

Then we computed the energy of the signal $y_2(t)$ using the equation below.

88

%QUESTION22

89

```
Y2Energy= sum(abs(Y2t.^2))*T;
```

24.0 Step 23

Then we computed the frequency spectrum $Y_2(f)$ of this signal.

90

%QUESTION23

91

```
Y2f= fftshift(fft(Y2t,framesize))*T;
```

25.0 Step 24

Then we plotted the magnitude of $y_2(f)$ in the frequency range $f_s/2 \leq f \leq f_s/2$ as shown in the figure below.

92

%QUESTION24

93

```
figure;
```

94

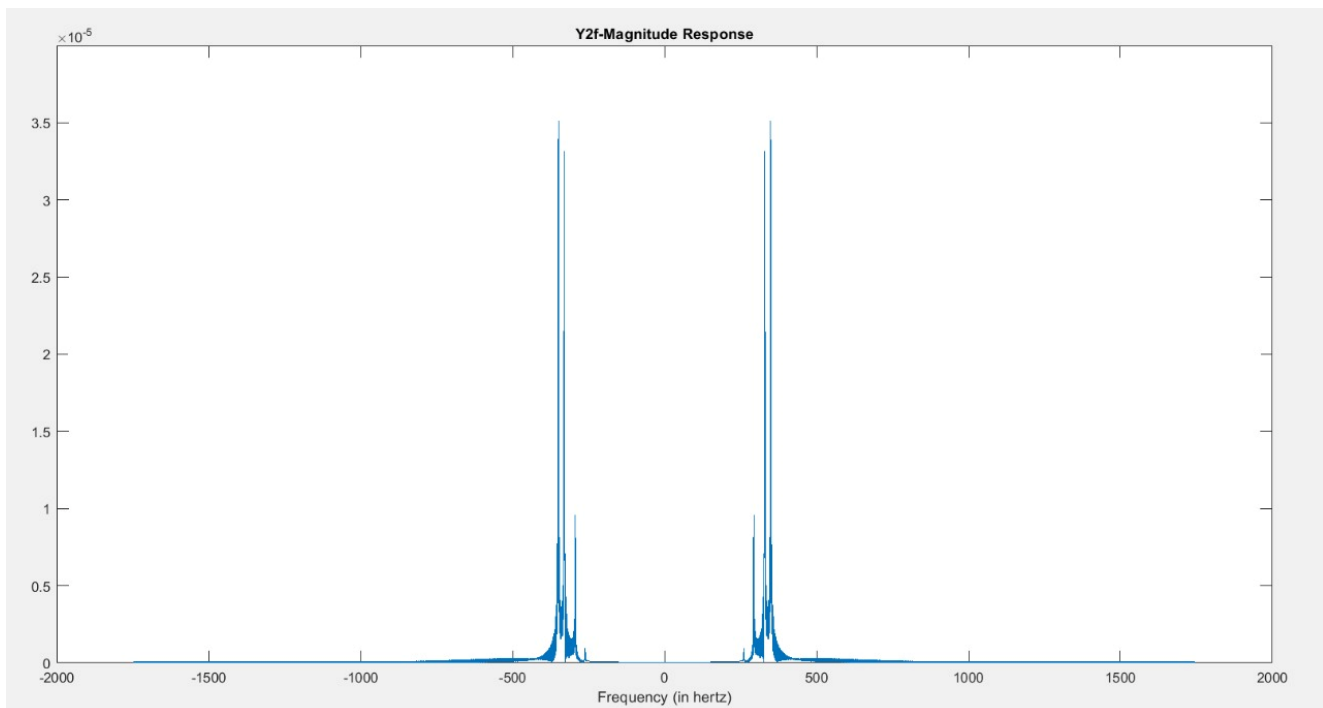
```
plot(f,abs(Y2f)/framesize);
```

95

```
xlabel('Frequency (in hertz)');
```

96

```
title('Y2f-Magnitude Response');
```



26.0 Step 25

Then we computed the Energy of the signal $y_2(t)$ from its frequency spectrum $y_2(f)$ and verified Parseval's Theorem using the equation below.

97

%QUESTION25

98

$Y2Energy2 = \sum (abs(Y2f.^2)) * dF;$

27.0 Contributions

	Habiba	Nadine	Salma	Hamsa	Mahamad
Responsibility: Indicate the specific part of preparing the report that each individual was responsible for.	2,8,9,10 17,18,19	4,5,7,13 16,22,25	1,3,6,12 15,21,24	Shared in the rest	0
Cooperation: (10 points) Able to work within team. Willingly performed tasks.	10	10	10	10	0
Punctuality (5 points) On time for team meetings.	5	5	5	5	0
Evaluative (10 points) Offer constructive criticism and helpful evaluation of work.	10	10	10	Medical Excuse	0
	Total: 25	Total:25	Total:25	Total:15 Medical Excuse (Tried to help as best as possible)	Total: 0

