

***INDIAN INSTITUTE OF  
INFORMATION TECHNOLOGY  
BHAGALPUR***



***DIGITAL SIGNAL PROCESSING  
LABORATORY REPORT (EC311)***

**NAME : SHASHI CHAURASIA**

**ROLL NO : 2001028**

**BRANCH : CSE**

**SUBJECT : EC311**

**INDIAN INSTITUTE OF INFORMATION TECHNOLOGY**

**SHASHI CHAURASIA 2001028 CSE**

## **EXPERIMENT 1**

**Aim: Write MATLAB Code to solve/plot:**

- a)  $Y = (a+b-c) (a-b-c)^2 (a+b+c)^3$ , where  
 $a=1, b=2, c=4$

```
Command Window

>> a=1;
b=2;
c=4;
x="Shashi 2001028";
f=(a+b+c);
f3=(a+b+c)^3;
f2=(a-b-c)^2;
f4=f*f2*f3;
display(x);
display(f4);

x =

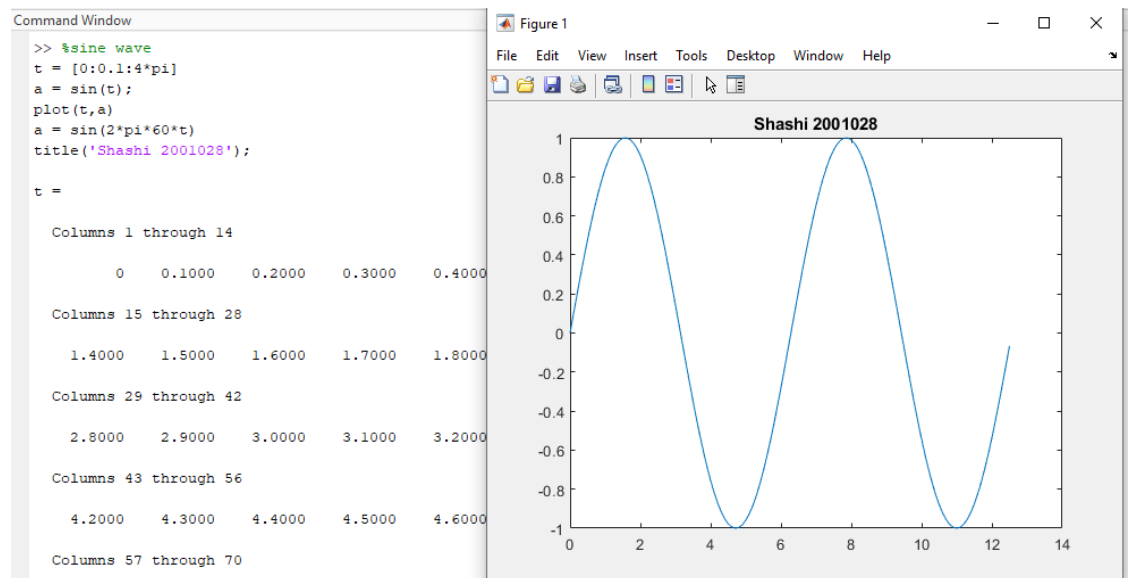
    "Shashi 2001028"

f4 =

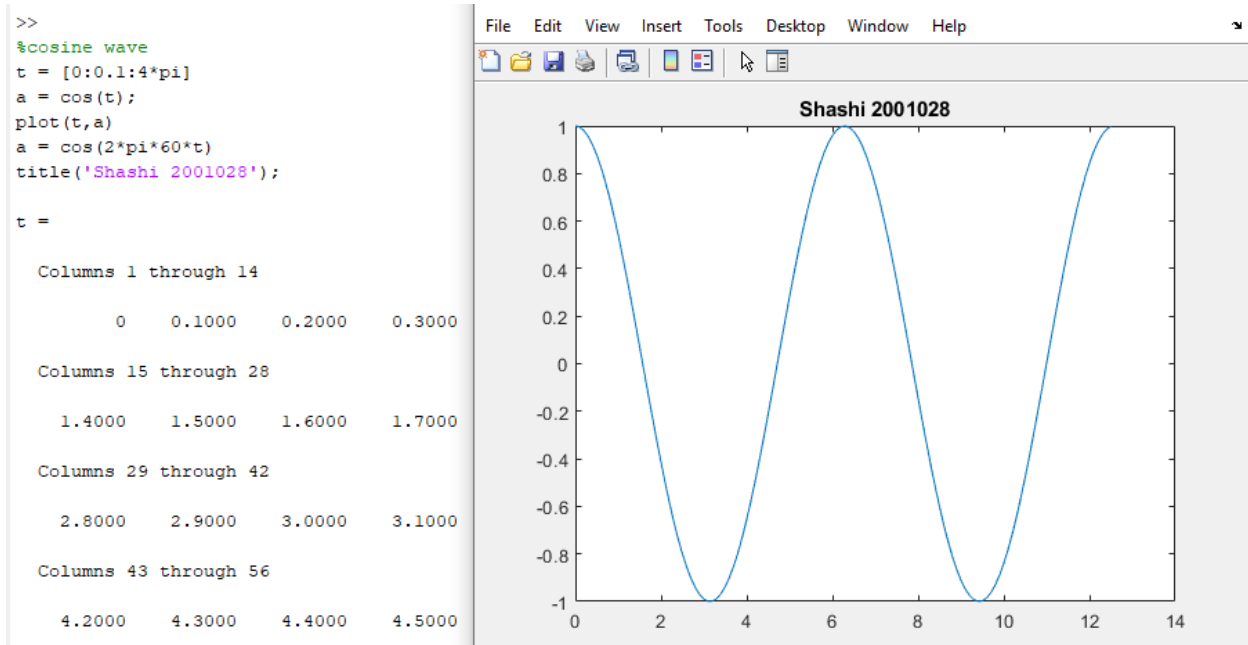
    60025

fx >> |
```

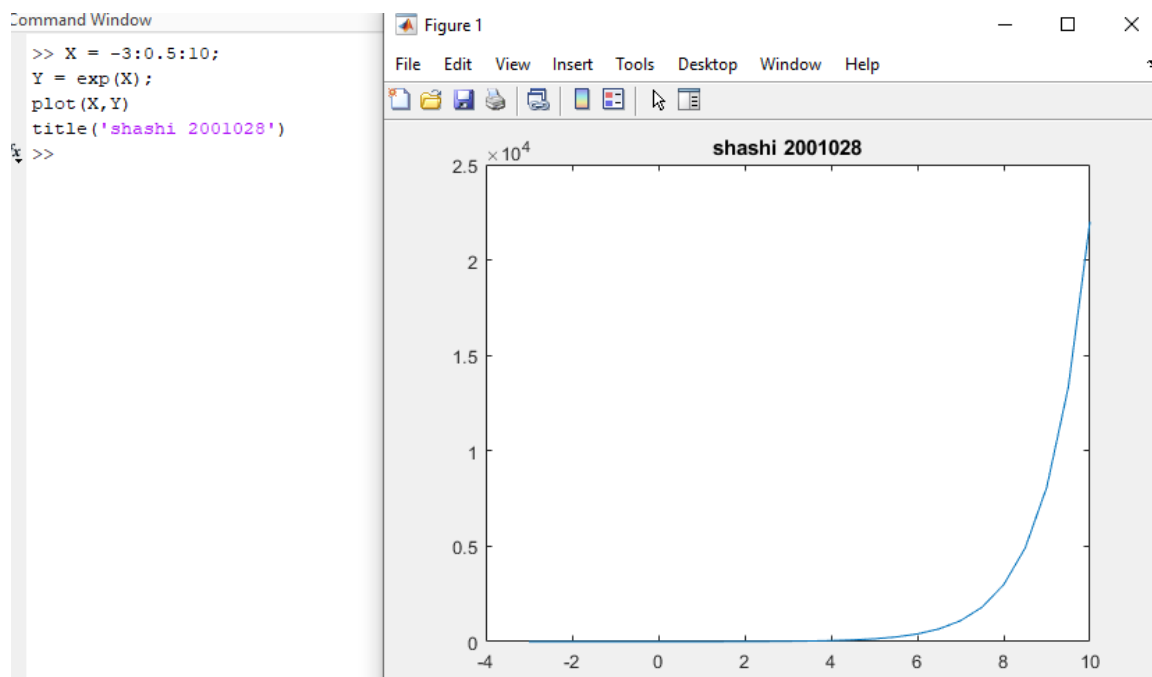
## b) Sine Wave



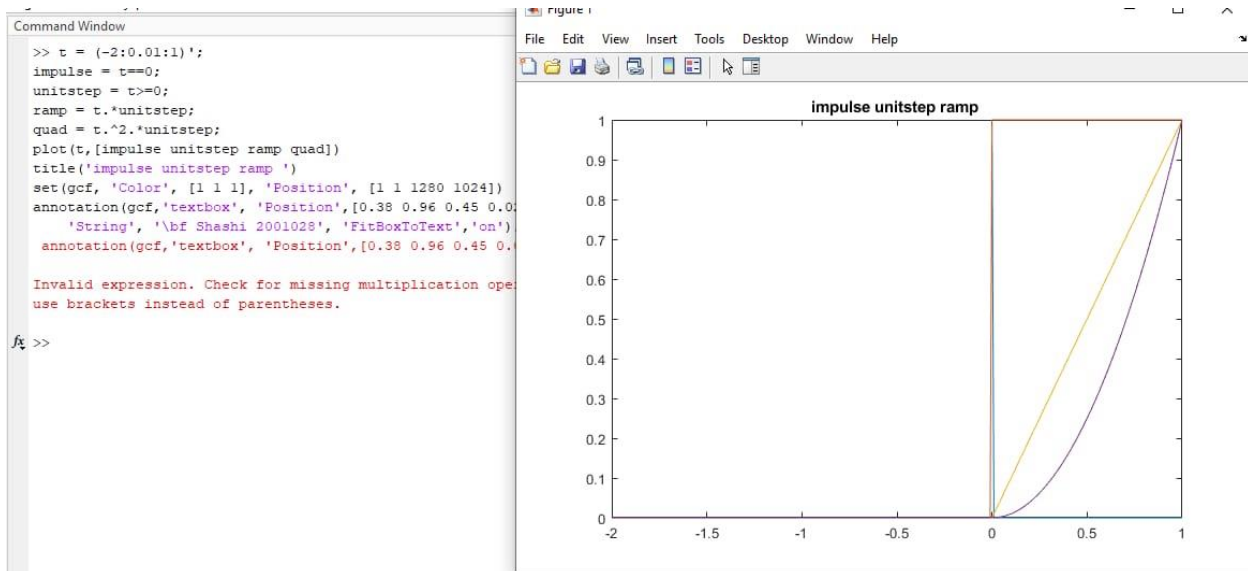
## c.) Cosine Wave



## D.) $e^x$



## E.) Unit Step ,Unit impulse, Ramp function



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AIM: Write MATLAB Function to solve/plot

## *EXPERIMENT 2*

a)  $Y = (a-b-c)^3$ , where  $a=2$ ,  $b=3$ ,  $c=4$

```

1 function [y]=equation(a,b,c)
2     y=(a+b+c)^3;
3     x="shashi 2001028";
4     display(x);
5 end

```

Command Window

```

>> a=2; b=3; c=4;
>> z= equation(a,b,c)

x =

    "shashi 2001028"

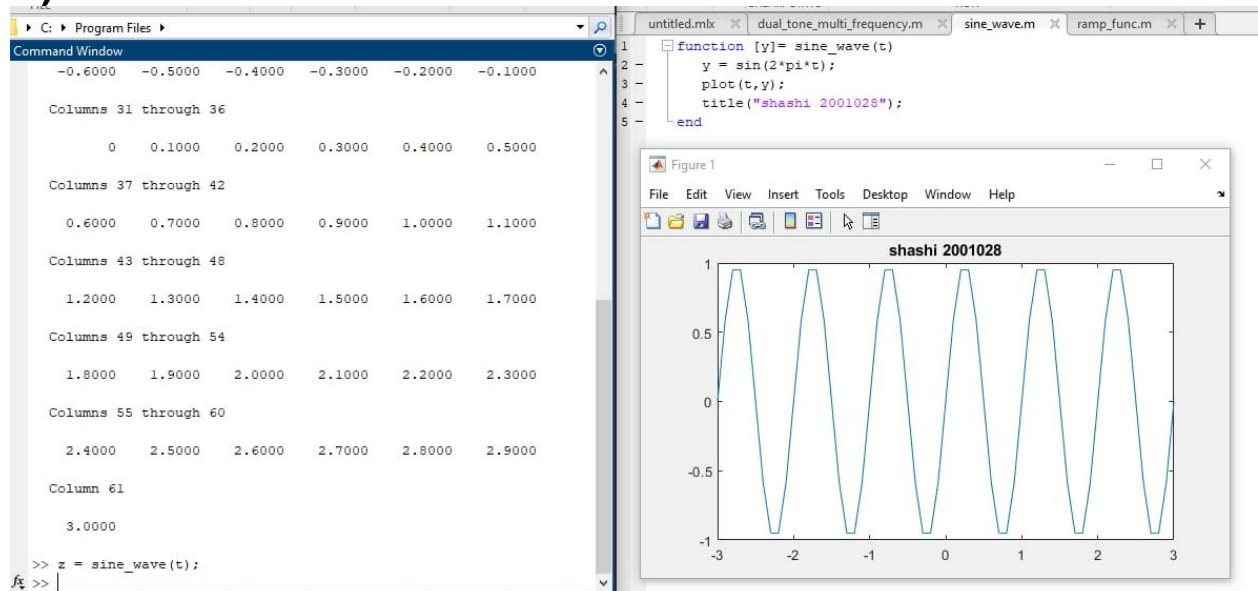
z =

    729

```

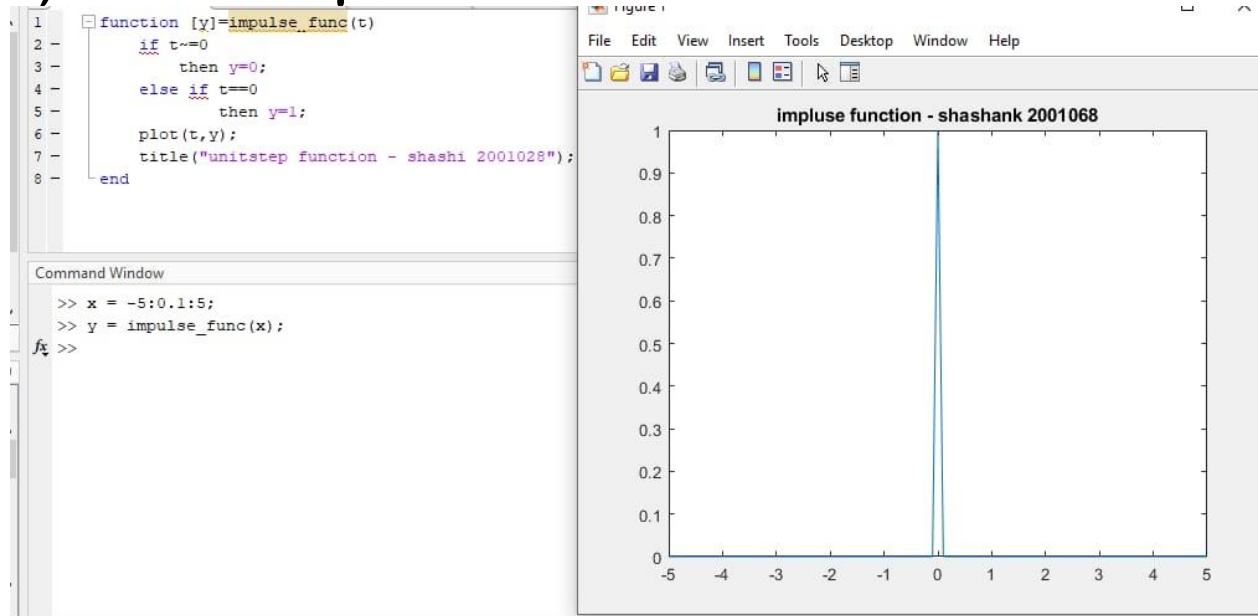
*fₓ* >> |

## b) Sine wave

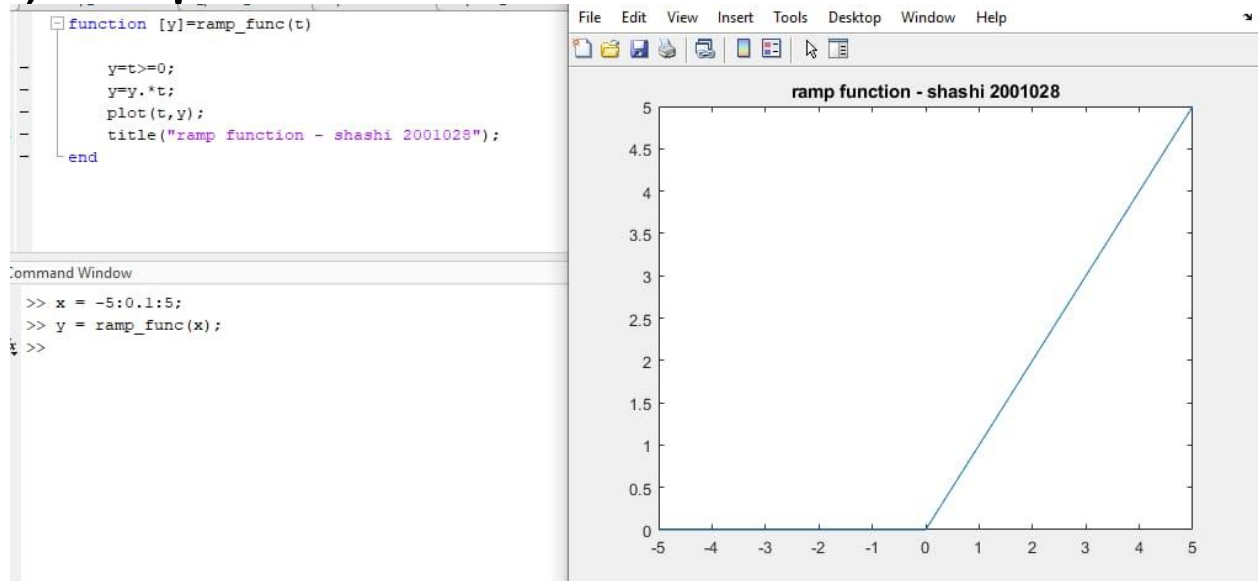




## e) Unit Impulse



## f) Ramp Function



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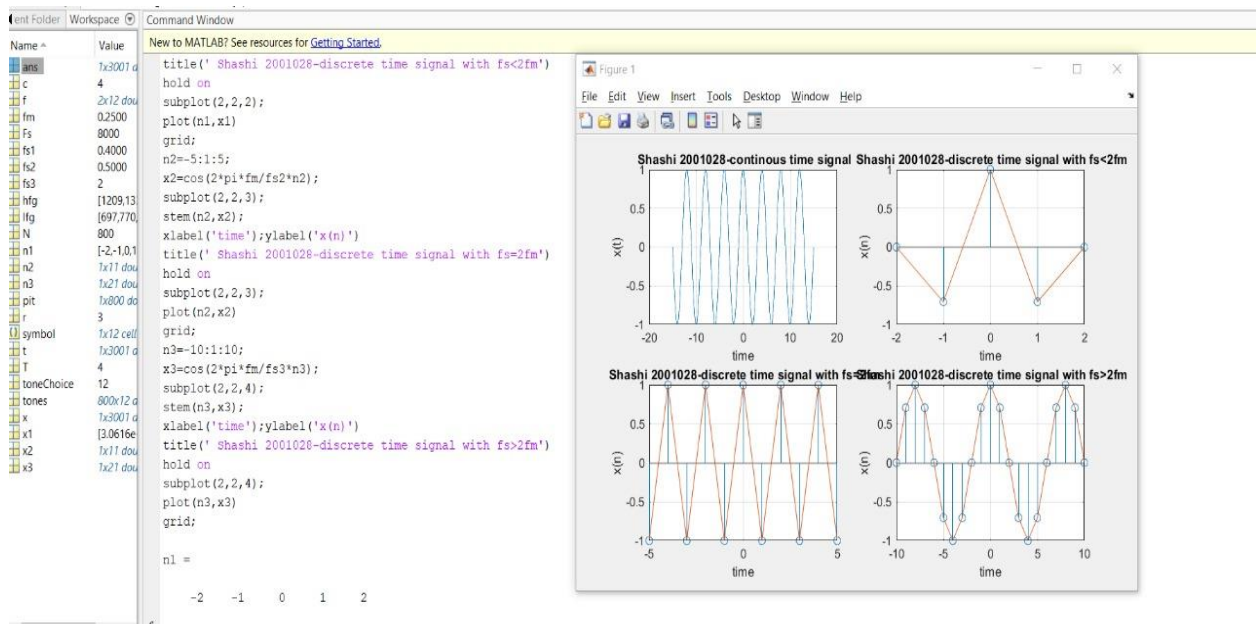
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# EXPERIMENT 3

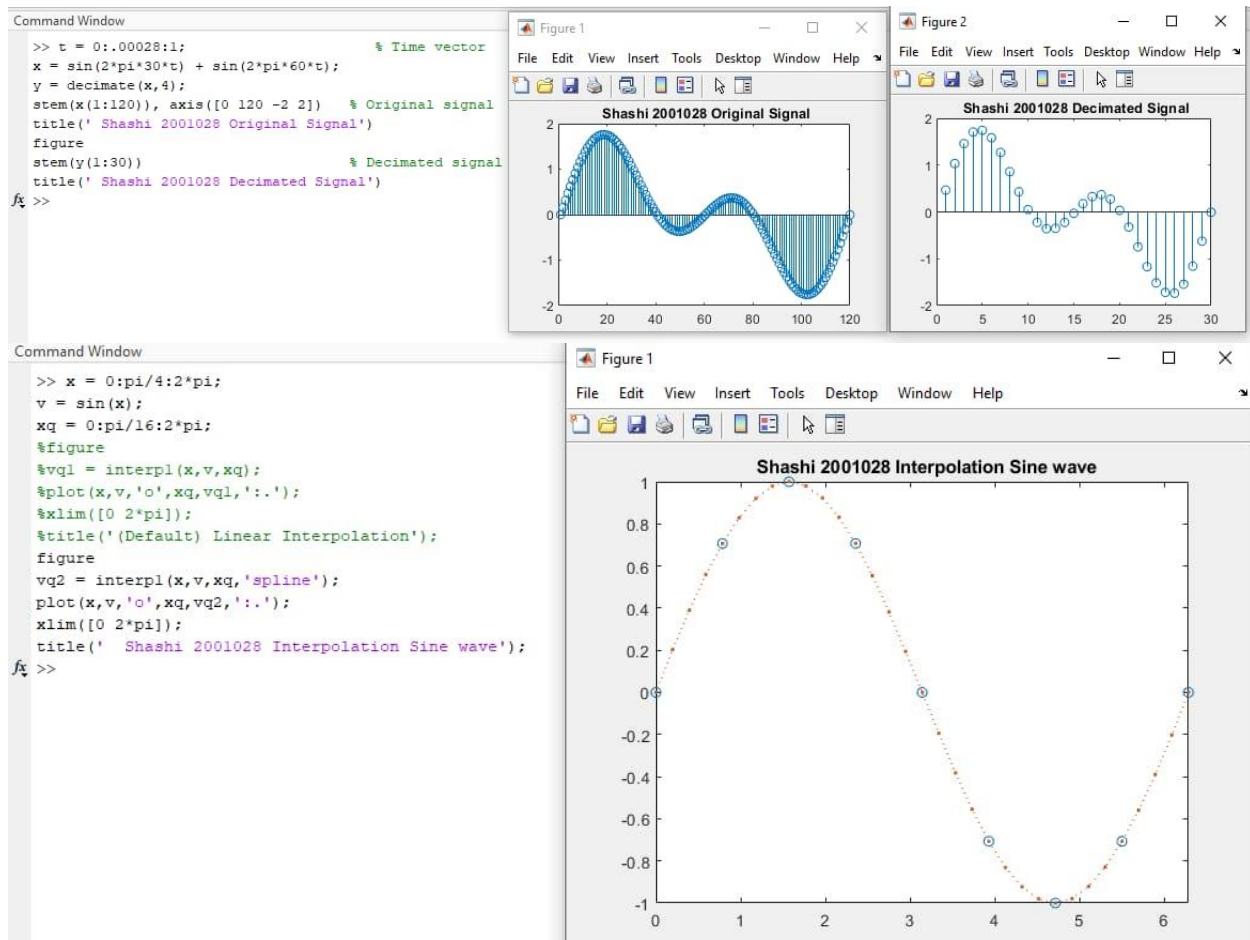
**Aim: Verify Sampling theorem through MATLAB coding.**

**Sampling Theorem:** *The sampling theorem specifies the minimum sampling rate at which a continuous-time signal needs to be uniformly sampled so that the original signal can be completely recovered or reconstructed by these samples alone. This is usually referred to as Shannon's sampling theorem in the literature.*



# *EXPERIMENT 4*

**Aim: Perform Interpolation and Decimation of any wave and verify any deformation in wave after performing both actions by same factor by using MATLAB.**



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# *EXPERIMENT 5*

# Aim: Write MATLAB code to plot graph for Dual Tone Multi Frequency for: 1, 2,3,4,5,6,7,8,9,0, \*, #

```

1 - symbol = {'1','2','3','4','5','6','7','8','9','*','0','#'};
2 - lfg = [697 770 852 941]; % Low frequency group
3 - hfg = [1209 1336 1477]; % High frequency group
4 - f = [];
5 - for c=1:4,
6 -     for r=1:3,
7 -         f = [ f [lfg(c);hfg(r)] ];
8 -     end
9 - end
10 - Fs = 8000; % Sampling frequency 8 kHz
11 - N = 800; % Tones of 100 ms
12 - t = (0:N-1)/Fs; % 800 samples at Fs
13 - pit = 2*pi*t;
14 - tones = zeros(N,size(f,2));
15 - for toneChoice=1:12,
16 -     % Generate tone
17 -     tones(:,toneChoice) = sum(sin(f(:,toneChoice)*pit));
18 -     % Plot tone
19 -     subplot(4,3,toneChoice),plot(t*1e3,tones(:,toneChoice));
20 -     title(['Symbol "', symbol{toneChoice}, "": [' ',num2str(f(1,toneChoice))',' ',num2str(f(2,toneChoice)),'']]);
21 -     set(gca, 'Xlim', [0 25]);
22 -     ylabel('Amplitude');
23 -     if toneChoice>9, xlabel('Time (ms)'); end
24 - end
25 - set(gcf, 'Color', [1 1 1], 'Position', [1 1 1280 1024])
26 - annotation(gcf,'textbox', 'Position',[0.38 0.96 0.45 0.026],...
27 -     'EdgeColor',[1 1 1],...

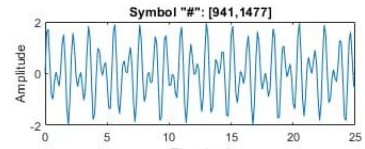
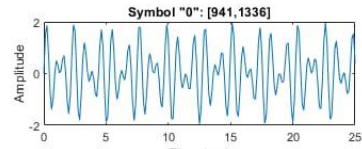
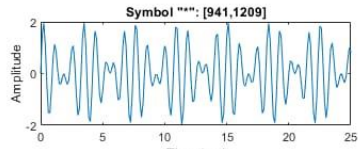
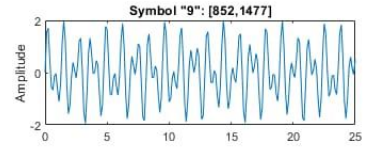
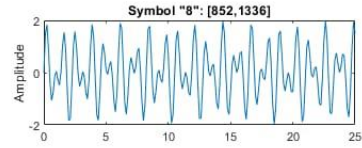
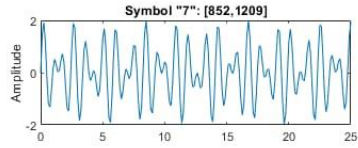
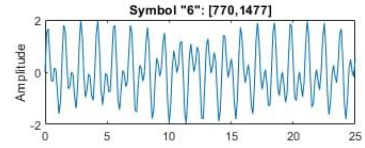
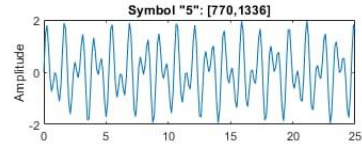
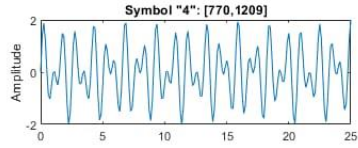
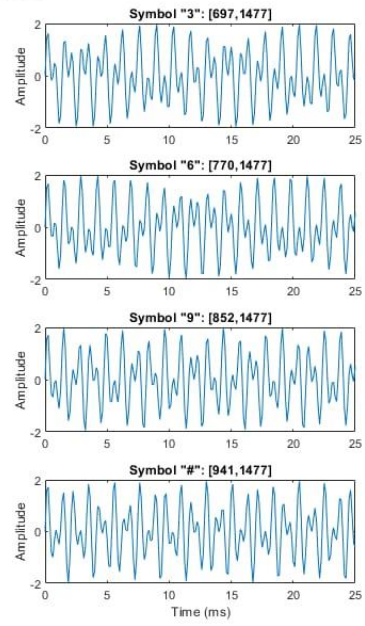
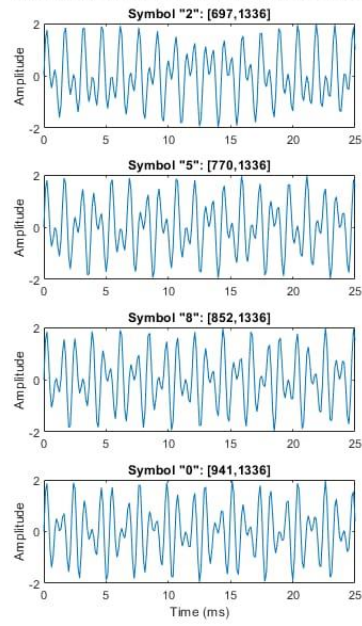
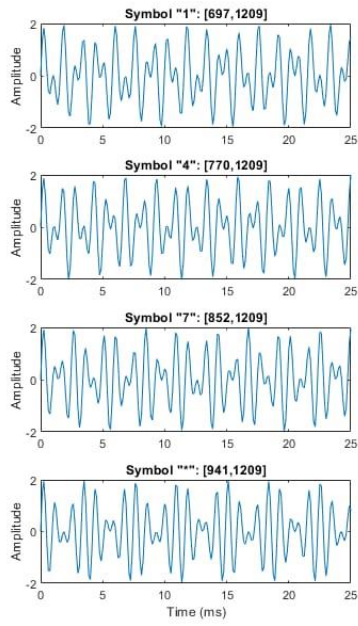
```

```

27 -     'EdgeColor',[1 1 1],...
28 -     'String', '\bf shashi 2001028 Time response of each tone of the telephone pad', ...
29 -     'FitBoxToText','on');
30 - symbol = {'1','2','3','4','5','6','7','8','9','*','0','#'};
31 - lfg = [697 770 852 941]; % Low frequency group
32 - hfg = [1209 1336 1477]; % High frequency group
33 - f = [];
34 - for c=1:4,
35 -     for r=1:3,
36 -         f = [ f [lfg(c);hfg(r)] ];
37 -     end
38 - end
39 - Fs = 8000; % Sampling frequency 8 kHz
40 - N = 800; % Tones of 100 ms
41 - t = (0:N-1)/Fs; % 800 samples at Fs
42 - pit = 2*pi*t;
43 - tones = zeros(N,size(f,2));
44 - for toneChoice=1:12,
45 -     % Generate tone
46 -     tones(:,toneChoice) = sum(sin(f(:,toneChoice)*pit));
47 -     % Plot tone
48 -     subplot(4,3,toneChoice),plot(t*1e3,tones(:,toneChoice));
49 -     title(['Symbol "', symbol{toneChoice}, "": [' ',num2str(f(1,toneChoice))',' ',num2str(f(2,toneChoice)),'']]);
50 -     set(gca, 'Xlim', [0 25]);
51 -     ylabel('Amplitude');
52 -     if toneChoice>9, xlabel('Time (ms)'); end
53 - end

```

shashi 2001028 Time response of each tone of the telephone pad



\_\_\_\_\_

## Question 6

(a) Write Matlab codes to find convolution, auto and cross-correlation of a given sequence and analyze output

A= [1 2 5 7]

B= [3 1 2 4]

**Answer-**

**For Convolution-**

MATLAB code-

% linear convolution

close all

x=input('Enter x: ');

h=input('Enter h: ');

m=length(x);

n=length(h);

X=[x,zeros(1,n)];

H=[h,zeros(1,m)];

for i=1:n+m-1

Y(i)=0;

for j=1:i

Y(i)=Y(i)+X(j)\*H(i-j+1);

end

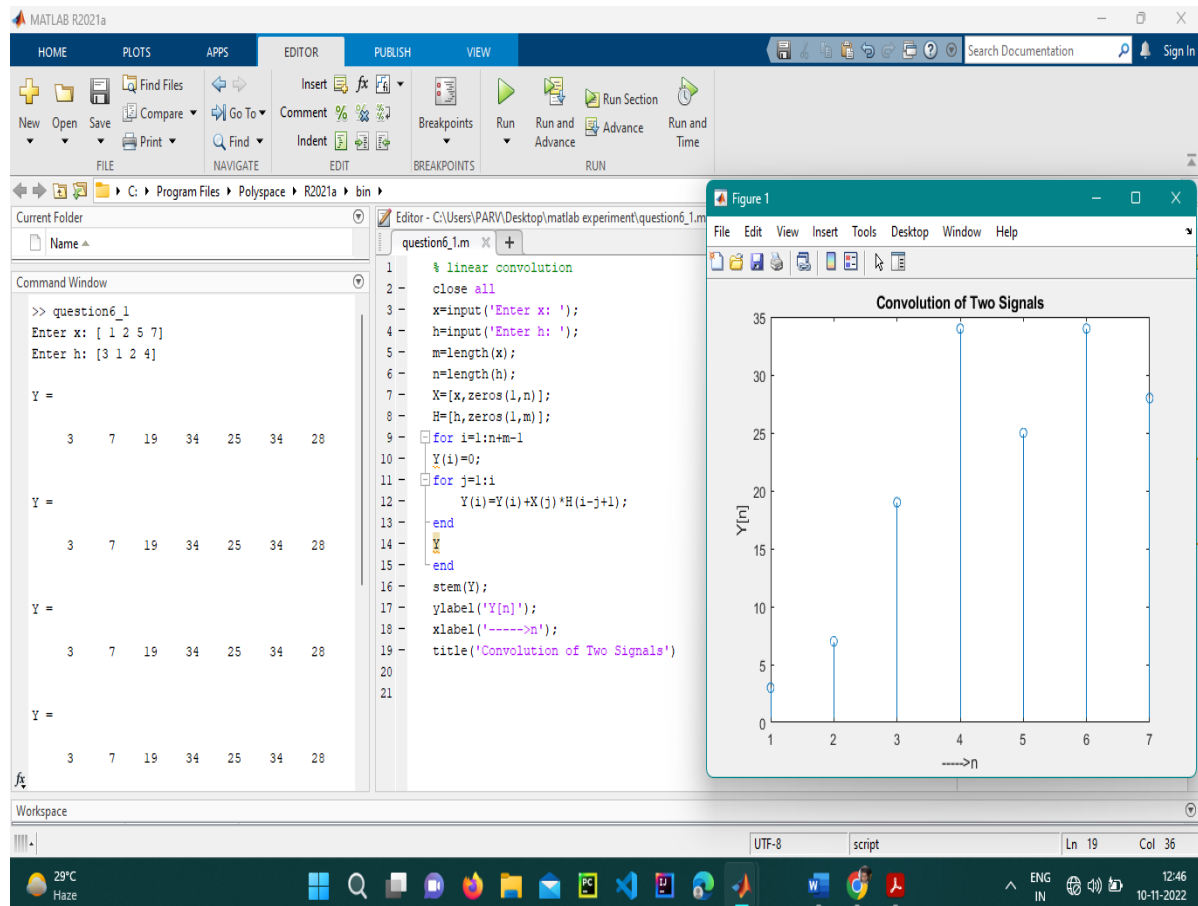
Y

end

stem(Y);

ylabel('Y[n]');

```
xlabel('----->n');
title('Convolution of Two Signals')
```



**For Auto Correlation-**

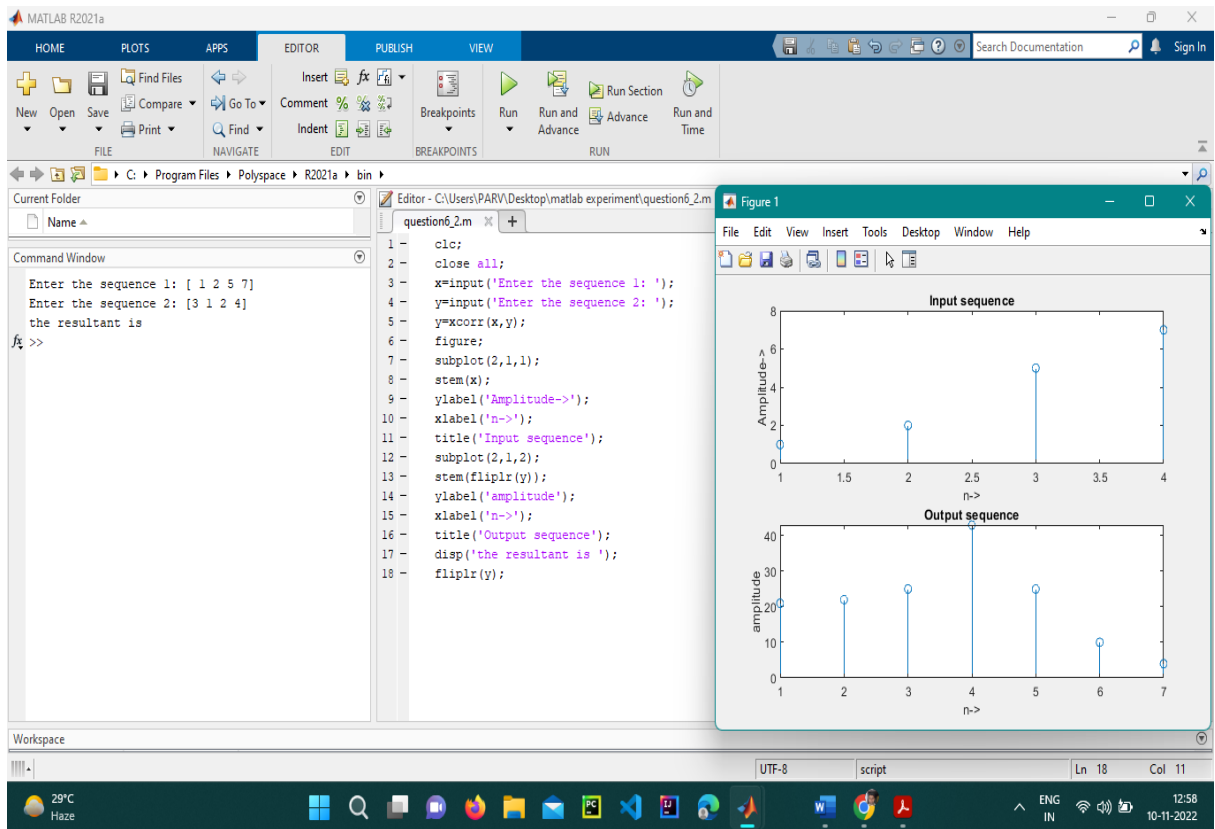
**MATLAB code-**

```
clc;
close all;
x=input('Enter the sequence 1: ');
y=input('Enter the sequence 2: ');
y=xcorr(x,y);
figure;
```

```

subplot(2,1,1);
stem(x);
ylabel('Amplitude->');
xlabel('n->');
title('Input sequence');
subplot(2,1,2);
stem(fliplr(y));
ylabel('amplitude');
xlabel('n->');
title('Output sequence');
disp('the resultant is ');
fliplr(y);

```

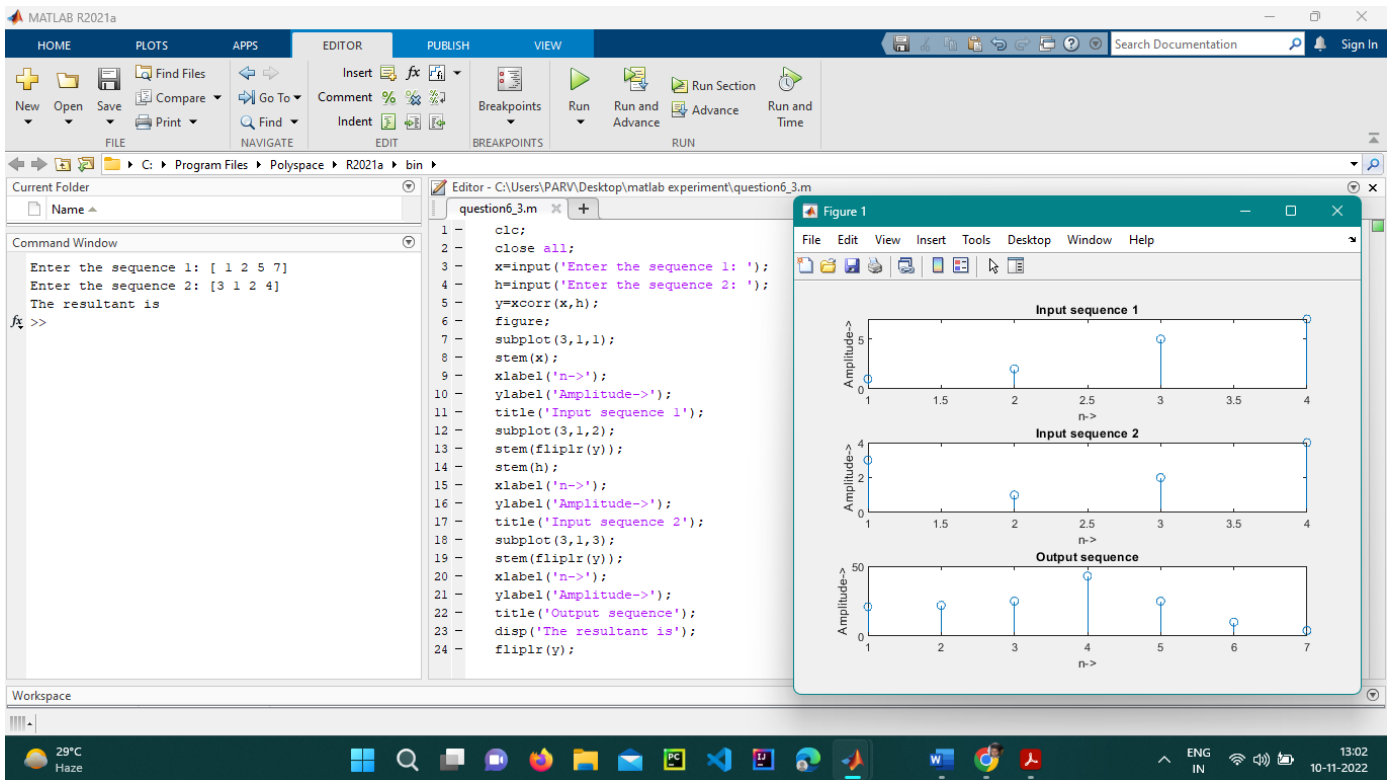


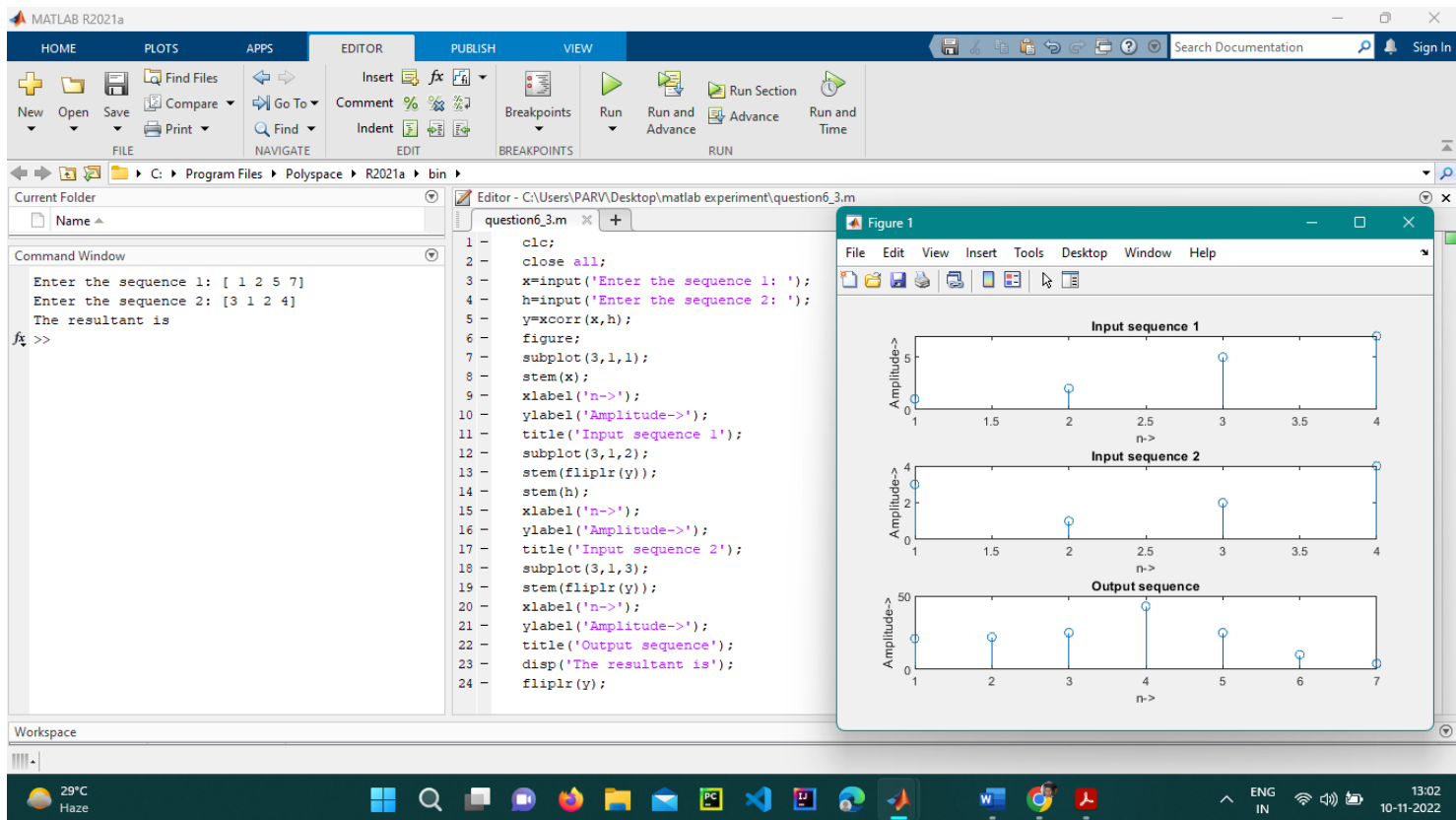
**For cross-correlation-**

**MATLAB code-**

```
x=input('Enter the sequence 1: ');
h=input('Enter the sequence 2: ');
y=xcorr(x,h);
figure;
subplot(3,1,1);
stem(x);
xlabel('n->');
ylabel('Amplitude->');
title('Input sequence 1');
subplot(3,1,2);
stem(fliplr(y));
stem(h);
xlabel('n->');
ylabel('Amplitude->');
title('Input sequence 2');
subplot(3,1,3);
stem(fliplr(y));
xlabel('n->');
ylabel('Amplitude->');
title('Output sequence');
disp('The resultant is');
fliplr(y);
```







## Question6.

(b) Write Matlab codes to find either convolution or auto correlation or cross-correlation without using inbuilt codes.

## Answer-

### MATLAB code-

```

clc;
close;
disp('enter the length of the first sequence m=');
m=input("");
disp('enter the first sequence x[m]=');
for i=1:m
x(i)=input("");
end
disp('enter the length of the second sequence n=');
n=input("");

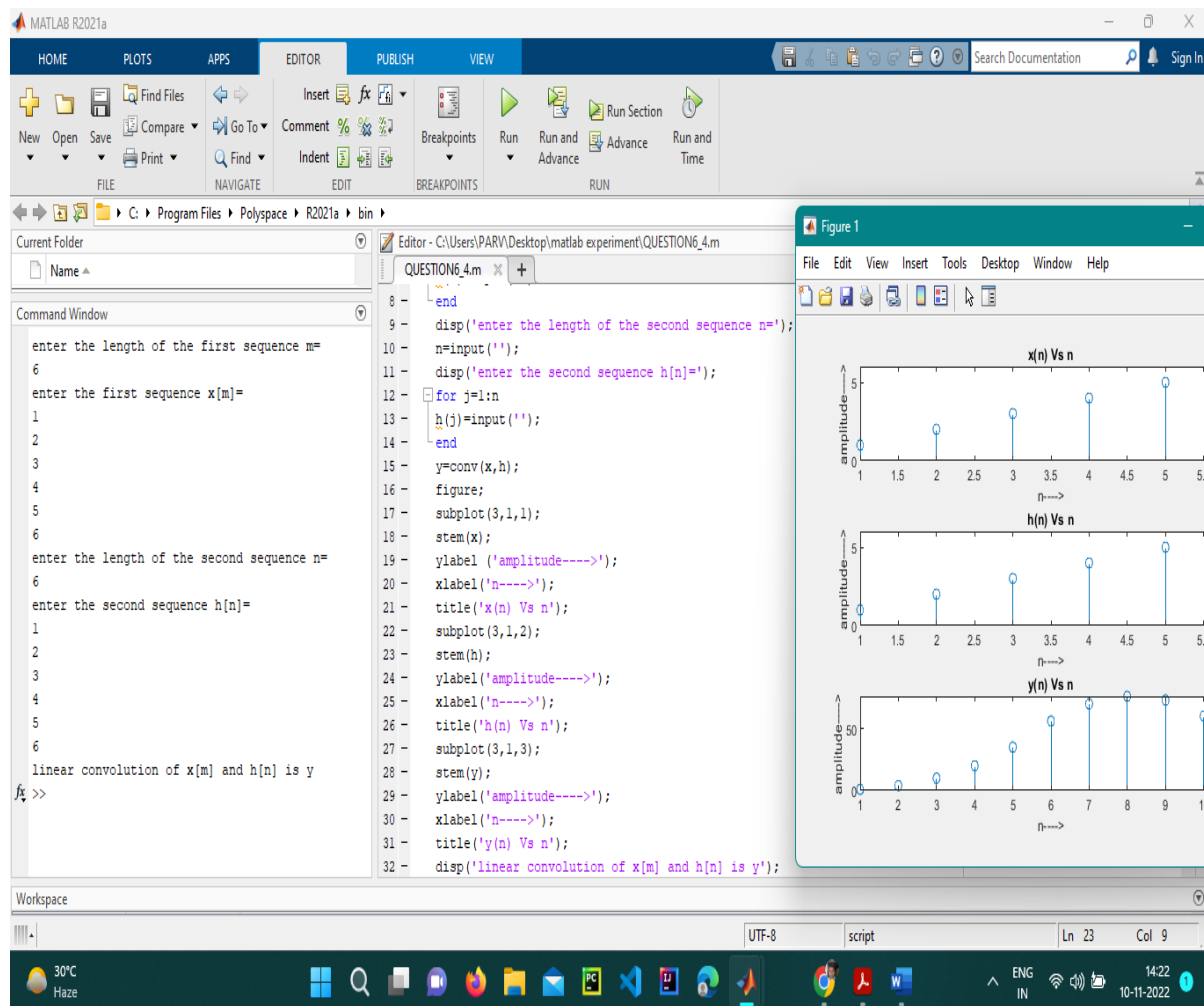
```

```

disp('enter the second sequence h[n]=');
for j=1:n
h(j)=input("");
end
y=conv(x,h);
figure;
subplot(3,1,1);
stem(x);
ylabel ('amplitude---->');
xlabel('n---->');
title('x(n) Vs n');
subplot(3,1,2);
stem(h);
ylabel('amplitude---->');
xlabel('n---->');
title('h(n) Vs n');
subplot(3,1,3);
stem(y);
ylabel('amplitude---->');
xlabel('n---->');

title('y(n) Vs n');disp('linear convolution of x[m] and h[n] is y');

```



## Question 7

Write a MATLAB code to design IIR Butterworth low pass, high pass, band pass and band stop filter and verify its characteristics for given cut-off frequency:

low pass- 1200 Hz

Highpass-1200 Hz

Band pass- 1200 Hz to 1800Hz

Band stop- 1200 Hz to 1800Hz

**Answer-**

**MATLAB code-**

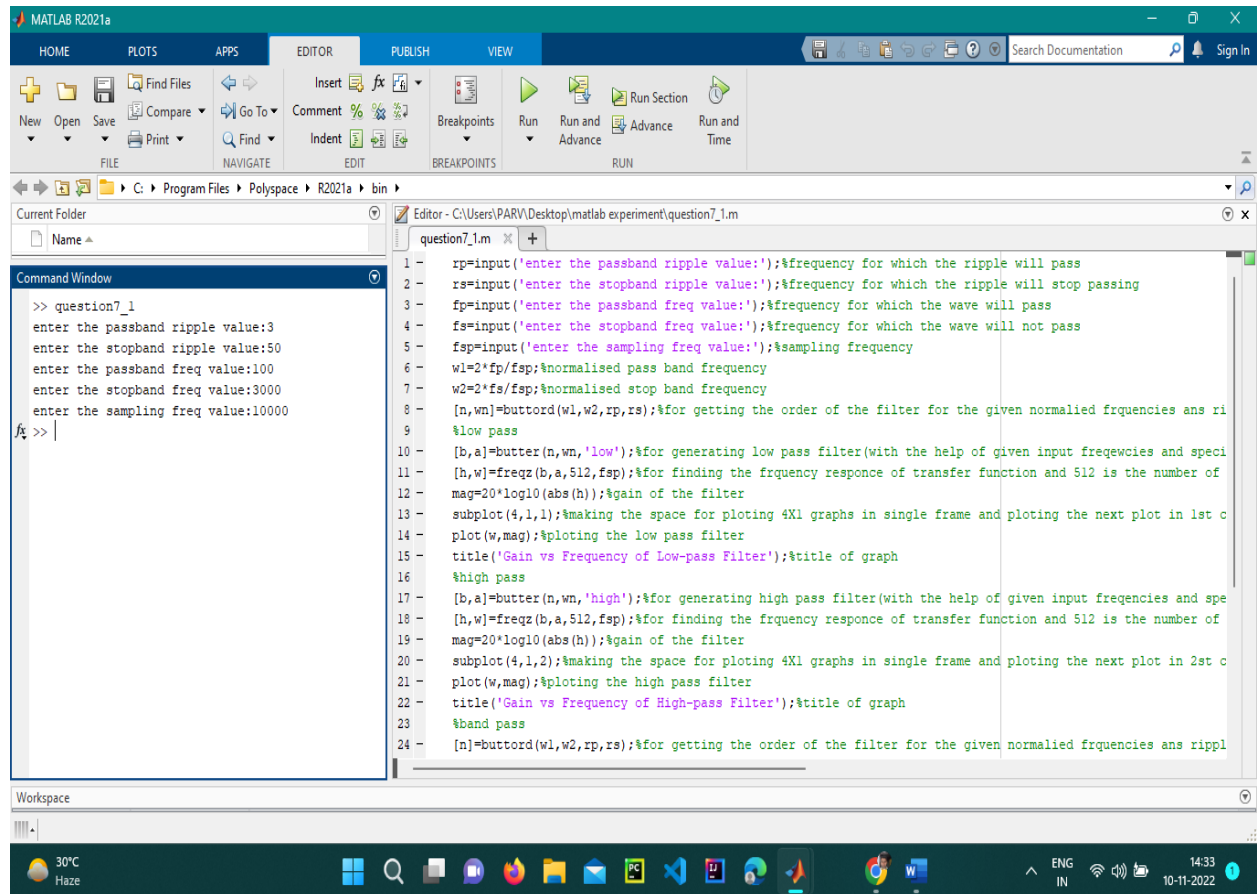
```
rp=input('enter passband ripple value');
rs=input('enter stopband ripple value');
fp=input('enter passband freq value');
fs=input('enter stopband freq value');
fsp=input('enter sampling freq value');
w1=2*fp/fsp;
w2=2*fs/fsp;
[n,wn]=buttord(w1,w2,rp,rs);
[b,a]=butter(n,wn,'low');
[h,w]=freqz(b,a,512,fsp);
mag=20*log(abs(h));
subplot(4,1,1);
plot(w,mag);
title('gain vs frequency of low pass filter');
%high pass
[b,a]=butter(n,wn,'high');
[h,w]=freqz(b,a,512,fsp);
mag=20*log(abs(h));
subplot(4,1,2);
plot(w,mag);
title('gain vs frequency of high-pass filter');
%band pass
[n]=buttord(w1,w2,rp,rs);
wn=[w1,w2];
[b,a]=butter(n,wn,'band pass');
[h,w]=freqz(b,a,512,fsp);
mag=20*log(abs(h));
subplot(4,1,3);
plot(w,mag);
title('gain vs frequency of band-pass filter');
%band stop
[n]=buttord(w1,w2,rp,rs);
[b,a]=butter(n,wn,'stop');
```

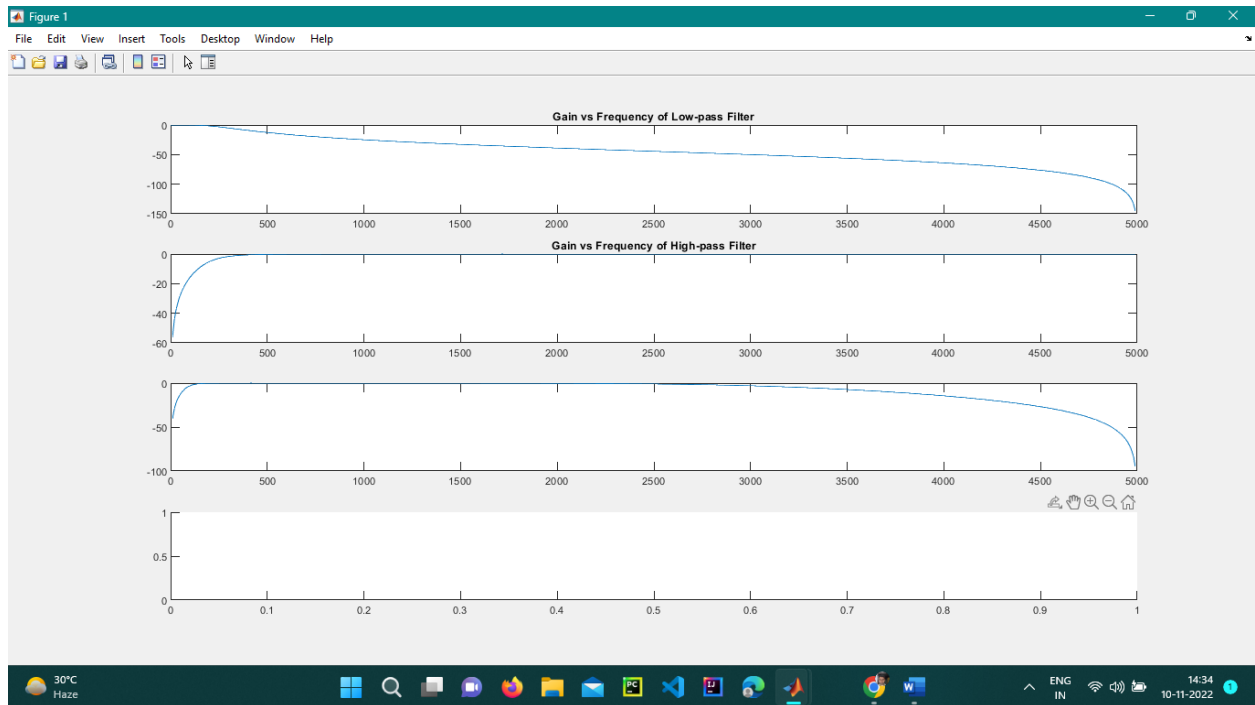
```

[h,w]=freqz(b,a,512,fs);
mag=20*log(abs(h));
subplot(4,1,4);
plot(w,mag);

title('gain vs frequency of band-stop filter');

```





## Question 8

Write a MATLAB code to design FIR low pass, high pass, band pass and band stop filters and verify its characteristics for given cut-off frequency:

Low pass- 1500 Hz

High pass-1500 Hz

Band pass- 1000 Hz to 2800Hz

Band stop- 1000 Hz to 2800Hz

**ANSWER-**

**MATLAB code-**

```

fp=input('enter the passing freq:');
fstop=input('enter the stopping freq:');
fsp=input('enter the sampling freq value:');
n=input('enter the order of the filter:');
w1=2*fp/fsp;
w2=2*fstop/fsp;
%low pass filter
A=fir1(n,w1,'Low');
%freqz(A,1,512,fsp);
[h,w]=freqz(A,1,512,fsp);
mag=20*log10(abs(h));
subplot(4,1,1);
plot(w,mag);
title('Gain vs Frequency of Low-pass Filter');
%high pass filter
A=fir1(n,w2,'high');
%freqz(A,1,512,fsp);
[h,w]=freqz(A,1,512,fsp);
mag=20*log10(abs(h));
subplot(4,1,2);
plot(w,mag);
title('Gain vs Frequency of High-pass Filter');
%band pass filter
wn=[w1,w2];
A=fir1(n,wn,'bandpass');
%freqz(A,1,512,fsp);
[h,w]=freqz(A,1,512,fsp);
mag=20*log10(abs(h));
subplot(4,1,3);
plot(w,mag);
title('Gain vs Frequency of Band-pass Filter');
%band stop filter
A=fir1(n,wn,'stop');
%freqz(A,1,512,fsp);
[h,w]=freqz(A,1,512,fsp);
mag=20*log10(abs(h));
subplot(4,1,4);
plot(w,mag);

```



title('Gain vs Frequency of Band-stop Filter');%title of graph

