

## Generation of Signals

Aim :- Write a MATLAB program to generate the continuous time signal and discrete time signal of unit step, unit impulse, ramp, periodic sinusoidal sinc function, plot all the signals.

Software Required :- Matlab Software.

### Theory :-

If the amplitude of the signal is defined at every instant of time then it is called continuous time signal. If the amplitude of the signal is defined at only at some instants of time then it is called discrete time signal.

If the signal repeats itself at regular intervals then it is called periodic signal. otherwise they are called aperiodic signals.

Examples :- Aperiodic Signals - Ramp, Impulse, unit Step, Sinc  
periodic Signals - Square, Sawtooth, Triangular

### unit step Signal

$$u(t) = \begin{cases} 1, & t \geq 0 \\ 0, & t < 0 \end{cases} \quad u[n] = \begin{cases} 1, & n \geq 0 \\ 0, & n < 0 \end{cases}$$

### unit impulse signal

$$\delta(t) = \begin{cases} 1, & t=0 \\ 0, & t \neq 0 \end{cases} \quad \delta[n] = \begin{cases} 1, & n=0 \\ 0, & n \neq 0 \end{cases}$$

Ramp signal

$$g(t) = \begin{cases} t, & t \geq 0 \\ 0, & t < 0 \end{cases} \quad g[n] = \begin{cases} n, & n \geq 0 \\ 0, & n < 0 \end{cases}$$

→ equivalently  $g(t) = t u(t)$        $g[n] = n u[n]$

Periodic sinusoids

sin and cos →  $A \cos(\omega t + \phi)$  &  $A \sin(\omega t + \phi)$

Sinc function

$$\text{sinc}(x) = \frac{\sin x}{x} \quad \text{if } x \neq 0$$

Procedure:-

- open MATLAB
- open new m-file
- Type the program
- Save in current directory
- compile and run the program
- For the output see Command window / figure window

Program:-

% Generation of signals

clc;

clear all;

close all;

% generation of unit step signal

```
t = linspace(-20, 20, 1000);
```

```
y = t >= 0;
```

```
figure(1);
```

```
Subplot(2,1,1);
```

```
plot(t, y)
```

```
grid
```

```
ylabel('c(t)');
```

```
xlabel('Time');
```

```
title('continuous time unit step');
```

```
axis([-20 20 -0.5 1.5]);
```

```
n = -10:1:10;
```

```
y = n >= 0;
```

```
Subplot(2,1,2);
```

```
stem(n, y)
```

```
grid
```

```
ylabel('c(n)');
```

```
xlabel('Time');
```

```
title('discrete time unit step');
```

```
axis([-10 10 -0.5 1.5])
```

% generation of unit impulse

```
t = (-5:0.01:5);
```

```
impulse = t == 0;
```

```
figure(2);
```

```
Subplot(2,1,1);
```

```
plot(t, impulse);
```

```
ylabel('delta(t)');
```

```

xlabel('Time');
title ('continuous time unit impulse signal');
n = -5:1:5;
y = (n == 0);
subplot(2,1,2);
stem(n,y);
ylabel ('delta(n)');
xlabel ('Time');
title ('discrete time unit impulse signal');

```

## 4. generation of Ramp Signal

```

t = linspace(-20,20,1000);
y = t.* (t >= 0);
figure(3);
subplot(2,1,1);
plot(t,y);
grid on;
ylabel('r(t)');
xlabel ('Time');
title ('continuous time Ramp signal');
n = -10: 1: 10;
y = n.* (n >= 0);
subplot(2,1,2);
stem(n,y);
grid on;
ylabel('r(n)');
title ('discrete time Ramp Signal');

```

## % generation of periodic sinusoidal

```

x=0:pi/100:2*pi;
y1=sin(x);
y2=cos(x);
figure(4);
plot(x,y1,'--',x,y2,'-');
grid on;
xlabel('Angle x \leq \pi');
ylabel('Amplitude');
legend('sin(x)', 'cos(x)');
title('continuous time periodic sinusoidal signals');
xticks(0:pi/2:2*pi);
xticklabels({'0', 'pi/2', 'pi', '3pi/2', '2pi'});
axis([0 2*pi -2 2]);
n=0:pi/10:2*pi;
y1=sin(n);
y2=cos(n);
figure(5);
subplot(2,1,1);
stem(n,y1,'g');
xlabel('Angle n \leq 2\pi');
ylabel('sin(n)');
title('Discrete time sine signal');
xticks(0:pi/2:2*pi);
xticklabels({'0', 'pi/2', 'pi', '3pi/2', '2pi'});
axis([0 2*pi -2 2]);
subplot(2,1,2);
stem(n,y2,'m');

```

```

ylabel('cos(n)');
title('Discrete time cosine signal');
xticks([0:pi/2:2*pi]);
xticklabels(['0', 'pi/2', 'pi', '3pi/2', '2pi']);
axis([-2*pi -2 2]);

```

### Sinc function generation

```
x = -2*pi:pi/1000:2*pi;
```

```
y = sinc(x);
```

```
figure(6)
```

```
subplot(2,1,1);
```

```
plot(x,y);
```

```
grid;
```

```
xticks(-2*pi:pi/2:2*pi);
```

```
xticklabels([-9pi, -3pi/2, -pi, -pi/2, 0, pi/2, 3pi/2, 9pi])
```

```
ylabel('sinc(x)');
```

```
xlabel('0 leq x leq 2pi');
```

```
title('continuous sinc signal');
```

```
axis([-2*pi 2*pi -0.5 1]);
```

```
n = (-2*pi:pi/10:2*pi);
```

```
y = sinc(n);
```

```
subplot(2,1,2);
```

```
stem(n,y,'y');
```

```
grid;
```

```
xticks(-2*pi:pi/2:2*pi);
```

```
ylabel('sinc(x)'),
```

```
title('Discrete sinc Signal');
```

```
axis([-2*pi 2*pi -0.5 1]);
```

Result generated the continuous time signals and discrete time signals - namely unit Step, unit Impulse, Ramp, periodic Sinusoidal & sinc function using MATLAB.

## Fouliier Series and Fourier Transform

@ Aim: Find the Fourier Transform of a square pulse using MATLAB. Plot its amplitude and phase spectrum.

Software Required: MatLab Software

### Theory:

Aperiodic signal can be expressed as a continuous sum (integral) of everlasting exponentials.

$$x(t) \iff X(\omega)$$

$$X(\omega) = \int_{-\infty}^{\infty} x(t) e^{-j\omega t} dt$$

We can plot the spectrum  $X(\omega)$  as a function of  $\omega$  since  $X(\omega)$  is complex, we have both amplitude & angle or phase spectra.

$$X(\omega) = |X(\omega)| e^{j\angle X(\omega)}$$

→ where  $|X(\omega)|$  is the amplitude of  $X(\omega)$ .

$\angle X(\omega)$  is the angle or phase of  $X(\omega)$

### Procedure:

- open MATLAB
- open new m-file
- Type the program
- Save in current directory.

- compile and Run the program
- For output see figure window

Program :-

```

clc;
x = 0:0.01:10';
g = Square(2*pi*x)';
y = abs(fft(x));
subplot(311);
plot(x,y)
axis([-2 12 -2 2]);
title('Square Pulse');
grid on;
subplot(312);
plot(fftshift(y));
title('Amplitude Spectrum');
grid on;
subplot(313);
z = angle(fftshift(y));
stem(z);
title('Phase Spectrum');
grid on;

```

Result :-

Fourier transform of the square pulse was generated and observed.

(b) Aim: Write a MATLAB program to find the trigonometric f. series coefficients of a rectangular periodic signal. Reconstruct the signal by combining the f. series coefficients with appropriate weightings.

Software Required: MATLAB Software.

### Theory:

→ Fourier developed a mathematical model to transform signals b/w time domain to frequency domain which is called Fourier Transform.

→ To represent any periodic signal  $x(t)$ , Fourier developed an expression called f. series

→ This is in terms of an infinite sum of sines and cosines or exponentials & F.S uses orthogonality conditions.

$$x(t) = \sum_{n=-\infty}^{\infty} a_n e^{jn\omega_0 t}$$

### Procedure:

- Switch on the computer & click on the MATLAB icon
- Write code
- Run the code
- For O/P see figure window
- Observe the plotted graph

## Program:-

```

clear all;
close all;
function f=exgb(n,t,T,k)
w0=(2*pi/T);
for i=1:n
    a(i)=2*sin(2*pi*i*T/T)/(pi*i);
end
disp('Trigonometric coeffs:');
a0=(2*t/T);
for j=1:length(k)
    f(j)=a0;
    for i=1:n
        f(j)=f(j)+a(i)*cos(i*w0*k(j));
    end
end
plot(k,f);

```

## Result:-

The trigonometric Fourier series coefficients of a rectangular periodic signal was calculated and observed.

(C) Aim :- write a MATLAB program to find the trigonometric and exponential fourier series coefficients of a periodic signal of rectangular. plot the discrete spectrum of the signal

Software Required: MATLAB Software

Theory:-

To represent any periodic signal  $x(t)$ , Fourier developed an expression called Fourier Series. This is in terms of an infinite sum of sines and cosines or exponentials.

$$x(t) = \sum_{n=-\infty}^{\infty} a_n e^{jn\omega_0 t}$$

$$\text{where } a_n = \frac{1}{T_0} \int_{t_0}^{t_0+T} x(t) e^{-jn\omega_0 t} dt$$

Trigonometric F.S

$$f(t) = a_0 + \sum_{n=1}^{\infty} (a_n \cos n\omega_0 t + b_n \sin n\omega_0 t)$$

$$\text{where } a_n = \frac{2}{T} \int_{t_0}^{t_0+T} f(t) \cos n\omega_0 t dt$$

$$a_0 = \frac{1}{T} \int_{t_0}^{t_0+T} f(t) dt$$

$$b_n = \frac{2}{T} \int_{t_0}^{t_0+T} f(t) \sin n\omega_0 t dt$$

EXponential F.S

$$f(t) = \sum_{n=-\infty}^{\infty} F_n e^{jn\omega_0 t}$$

$$\text{where } F_n = \frac{1}{T} \int_0^T f(t) e^{-j\omega_0 n t} dt$$

Procedure

- open MATLAB
- open new m-file
- type the program
- compile and run program
- observe o/p graphs

Program

```

clear all;
close all;
function f=ex2c(n,t,T,K,w)
w_0=(2*pi/T);
for i=1:n
    f(i)=(2*pi*t*i*sinc(2*pi*i*t/T)/T);
end
f_0=(2*pi*t/T);
j=8*pi*(-1)
for l=1:length(K)
    y(l)=f_l;
    for m=1:n

```

```

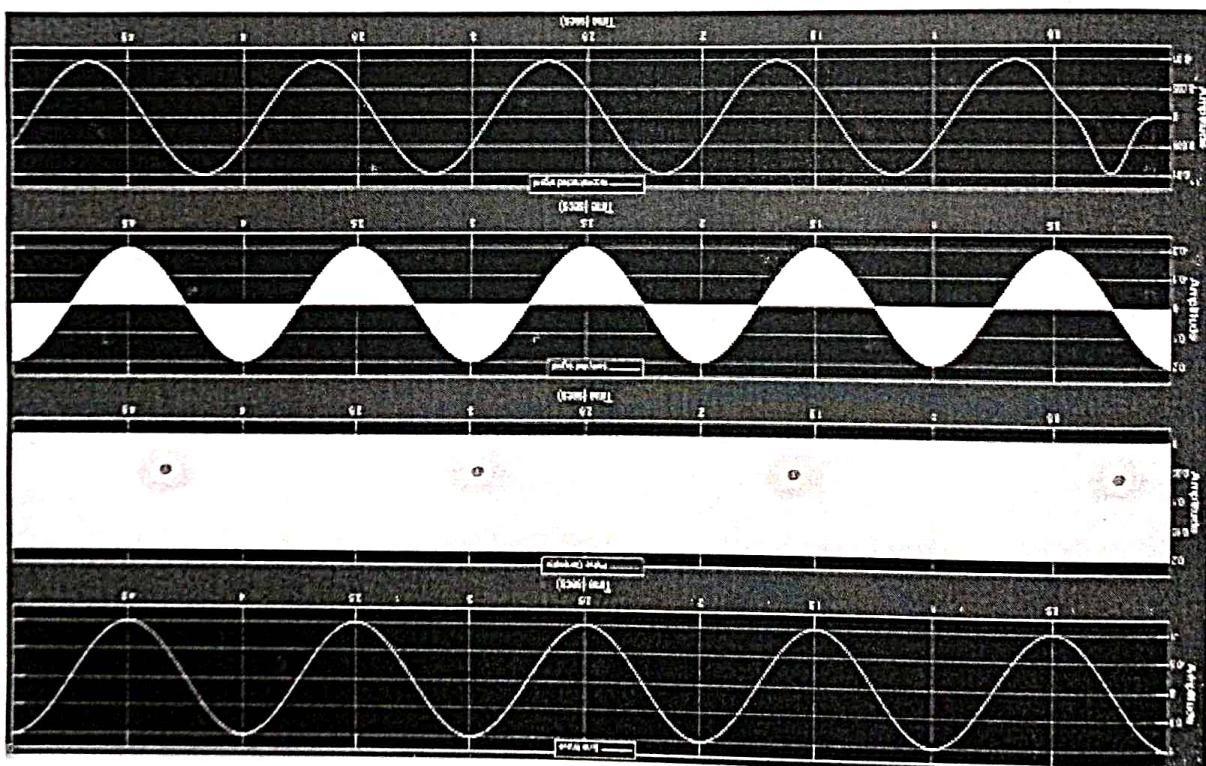
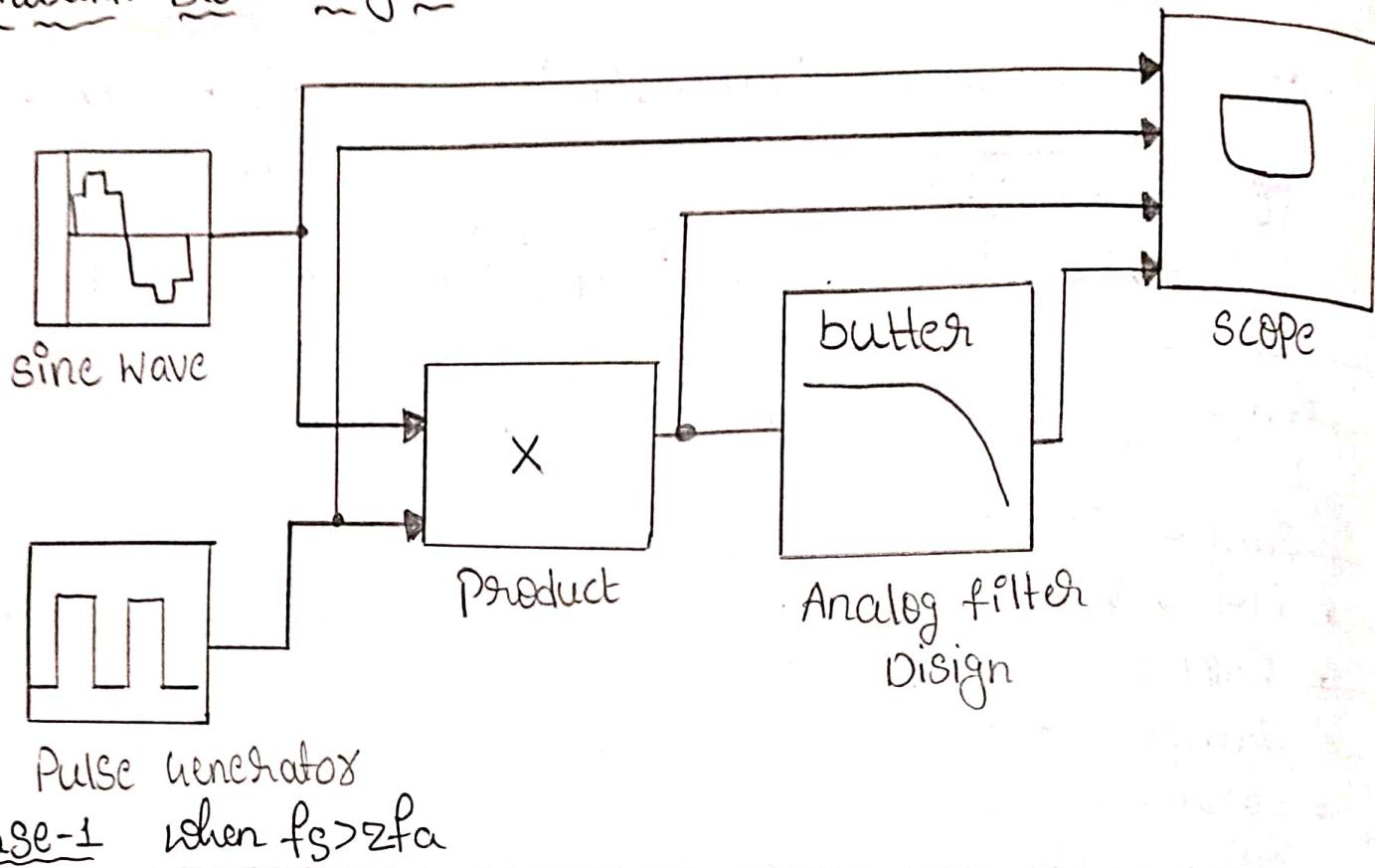
 $y(l) = y(l) + f(m) * \exp(-j * m * w_0 * k(l));$ 
end
for m = n+1 : Q * h
 $y(l) = y(l) + f_1(m-n) * \exp(j * (m-n) * w_0 * k(l));$ 
end
end
F = FFT(Y);
subplot(1, 1);
plot(k, Y);
title('Reconstructed Signal')
subplot(2, 1);
stem(abs(F));
title('Discrete Signal')

```

Result:

the trigonometric and exponential Fourier series coefficients of a periodic signal of rectangular way calculated and observed.

## Simulink Block diagram



## Sampling and reconstruction of signal

Aim: To verify sampling theorem for different sampling frequencies using Simulink.

Apparatus Required: (a) Hardware Tools: computer system  
(b) Software Tool: MATLAB10d upgraded version

### Theory:

Sampling theory is the bridge b/w the continuous time and discrete-time worlds.

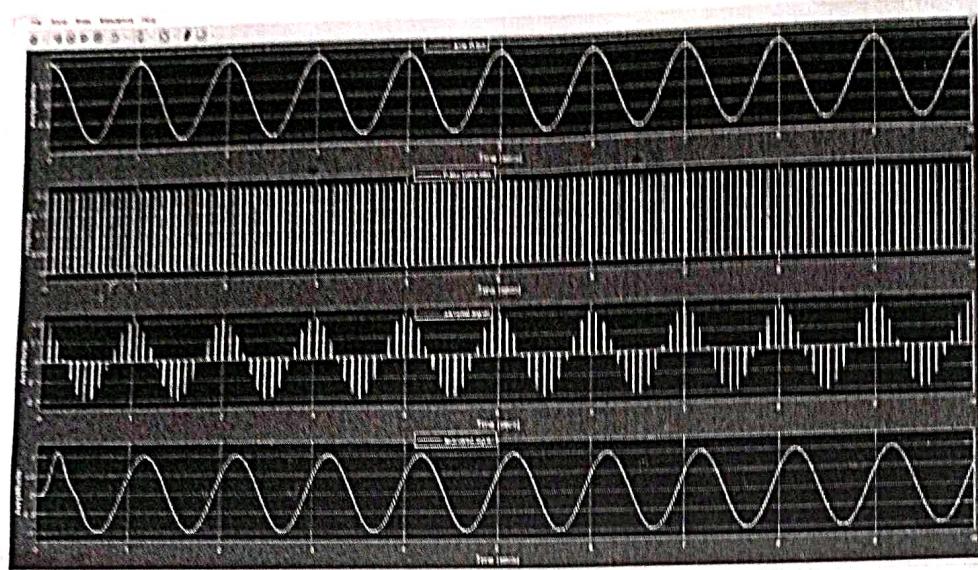
As a first step to convert analog signals into digital form, the samples of the analog signals are taken at regular intervals. The levels of these signals are then encoded and send to the receiver. At the receiver these samples are recovered and form that the original signal is reconstructed.

Sampling theorem states that original signal can be faithfully reconstructed only if the sampling frequency is at least double that of the highest frequency component in sampled signal.

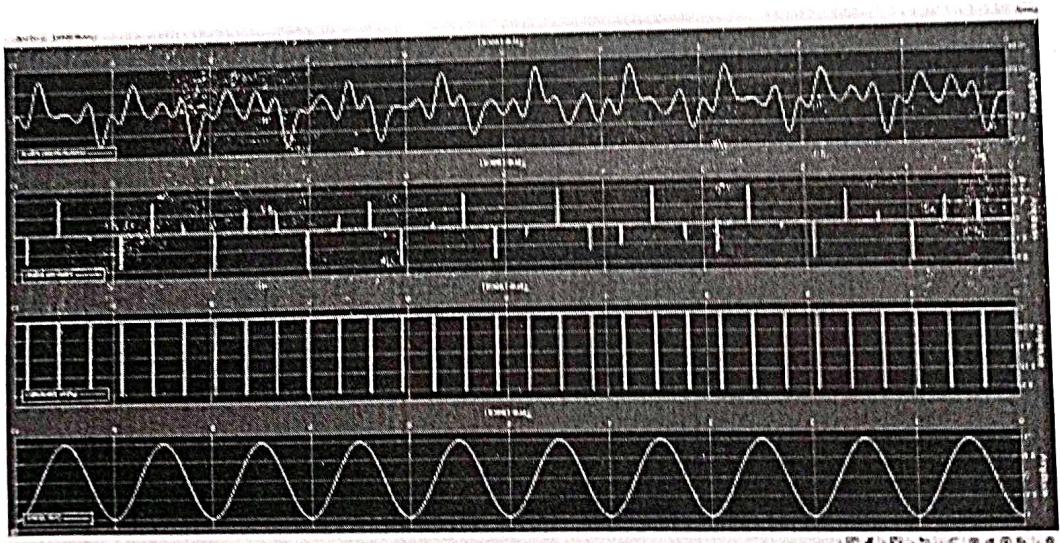
### Procedures:

1. Switch on the computer and click on the MATLAB icon
2. Go to Start at the bottom of the command window then select "Simulink" then go to library browser

case-2 when  $f_s = 2f_a$



case-3 when  $f_s < 2f_a$



drag it into creating file & once you open the MATLAB then click on the Simulink icon

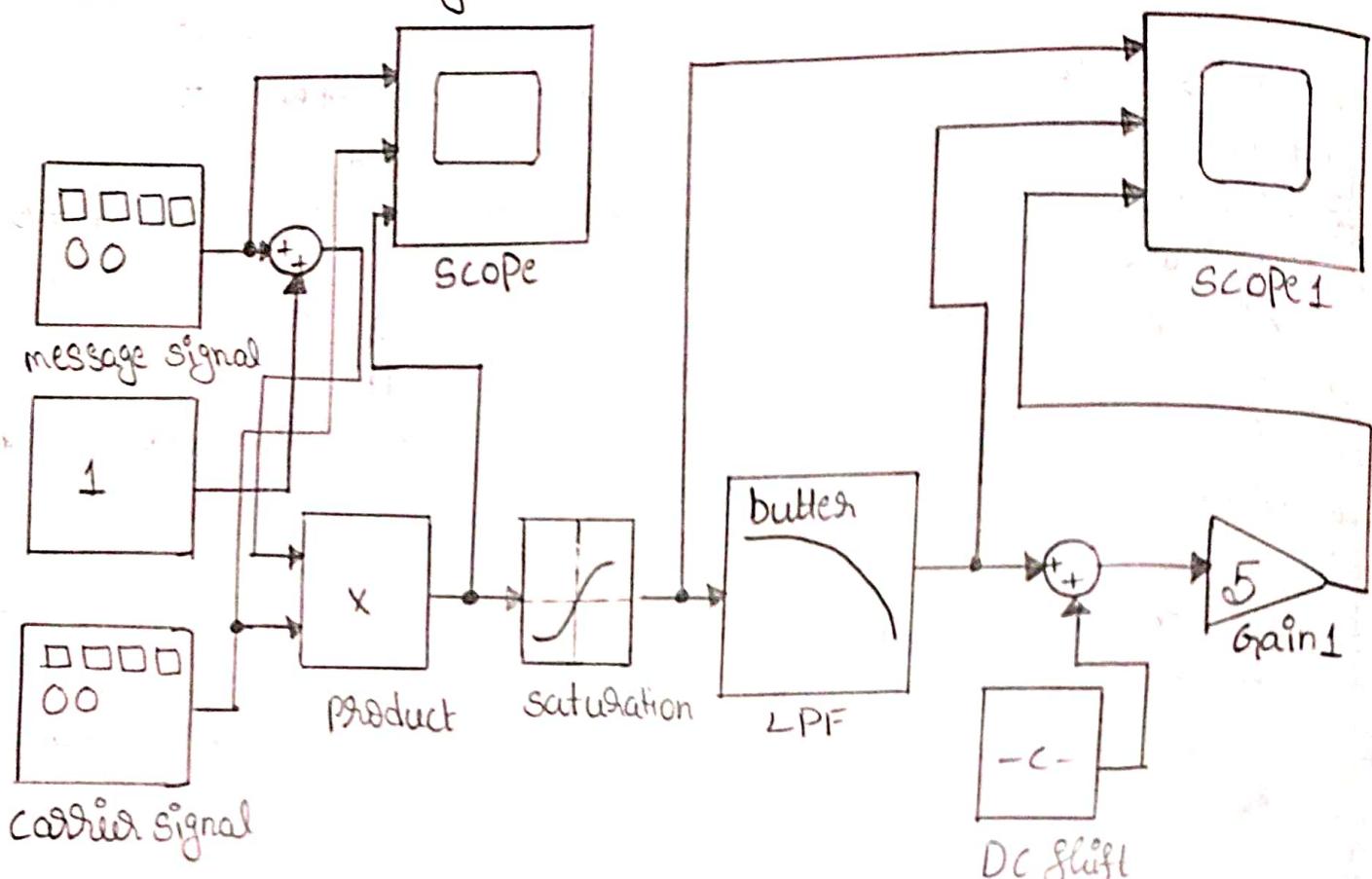
3. Go to file and select new and then Select model  
you will get a new window.

4. Arrange the functional blocks as shown in  
Simulink model

5. Assign required parameters to each functional block  
6. observe the outputs on scope.

Result: Sampling theorem is verified.

## Simulink Block diagram.



## Observation table

$m$	$V_{max}$	$V_{min}$	$\frac{V_{max} - V_{min}}{V_{max} + V_{min}}$
0.5	2.7	0.9	$\frac{2.7 - 0.9}{2.7 + 0.9} = 0.5$
1	4	0	$4/4 = 1$
1.5	5	-1	$5 - (-1) / 5 + (-1) = 1.5$

## Amplitude modulation & Demodulation

Aim: To Study the function of Amplitude modulation and demodulation (under modulation) using simulink

### Apparatus Required:

- (a) Hardware tools: Computer system
- (b) Software tool: MATLAB 7.0 and above version.

### Theory:

Amplitude modulation is defined as a process in which the amplitude of carrier wave  $c(t)$  is varied linearly with the instantaneous value of  $m(t)$ .

Let  $m(t)$  is the source message signal & F.T is  $M(f)$

Let  $c(t)$  is carrier signal

→ To move the freq response of  $m(t)$  to a new frequency band centered at  $f_c$  Hz, we use frequency shifting property of F.T

→ multiply  $m(t)$  by a sinusoid of frequency  $f_c$  such that

$$s(t) = m(t) \cos 2\pi f_c t$$

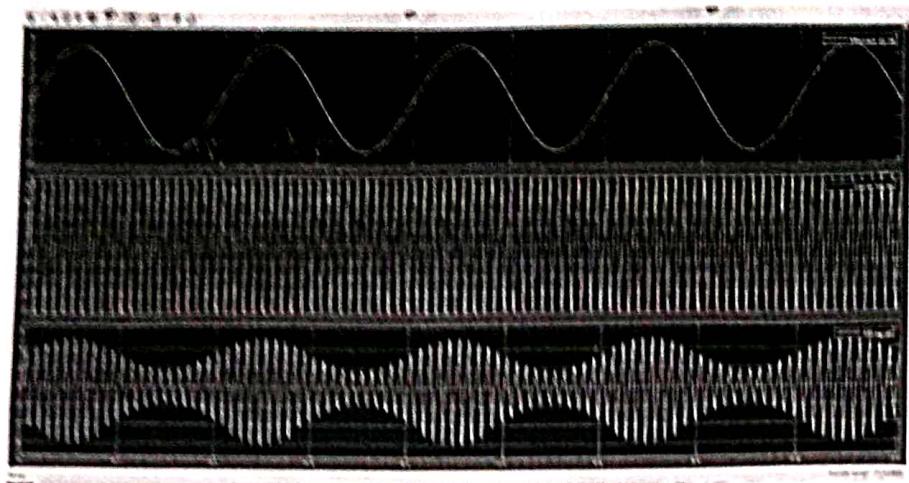
$$S(f) = \frac{1}{2} [m(f-f_c) + M(f+f_c)]$$

mathematical representation of AM (Time domain representation)

Let the message signal be sinusoidal and be represented as  $e_m = E_m \cos \omega_m t$

Similarly  $e_c = E_c \cos \omega_c t$

50% Modulation



50% Demodulation

where  $e_m \rightarrow$  instantaneous amplitude of modulating signal  
 $E_m \rightarrow$  Peak amplitude

$w_m \rightarrow 2\pi f_m$  and  $f_m \rightarrow$  freq. of modulating signal

$E_c \rightarrow$  peak amplitude of carrier signal

$f_c \rightarrow$  carrier freq.

$\rightarrow$  Am wave is expressed by the following eqn

$$e_{Am} = A \cos \omega t$$

where  $A \rightarrow E_c + E_m \cos w_m t$

$$e_{Am} = [E_c + E_m \cos w_m t] \cos \omega c t$$

$$e_{Am} = E_c \left[ 1 + \frac{E_m}{E_c} \cos w_m t \right] \cos \omega c t$$

if  $\frac{E_m}{E_c} = m$  be the modulation index.

$$e_{Am} = E_c [1 + m \cos w_m t] \cos \omega c t$$

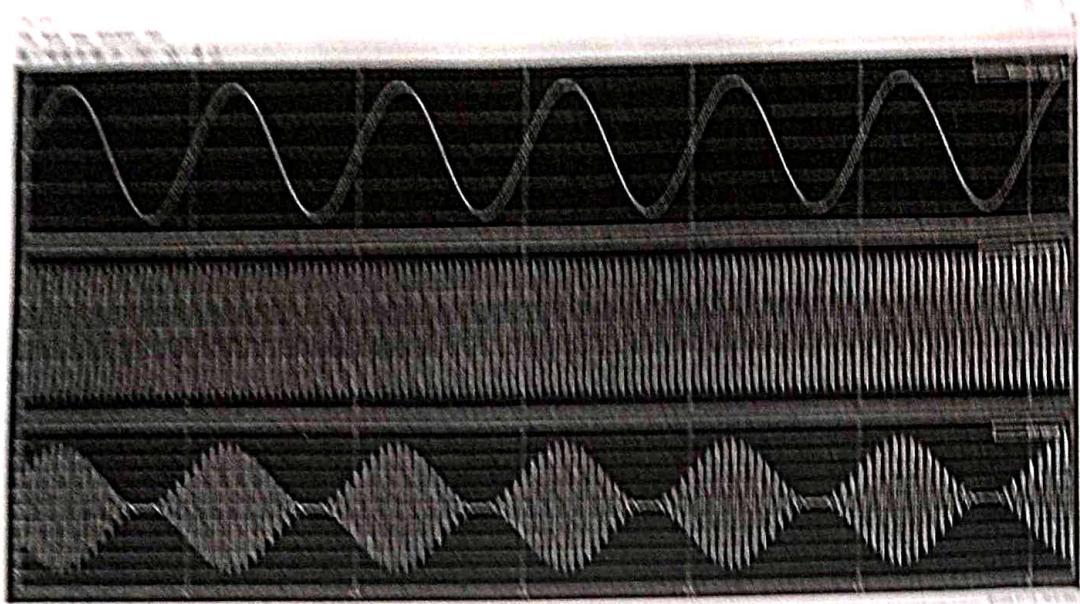
when  $E_m \leq E_c \rightarrow$  no distortion

$E_m > E_c \rightarrow$  distortion (over modulation)

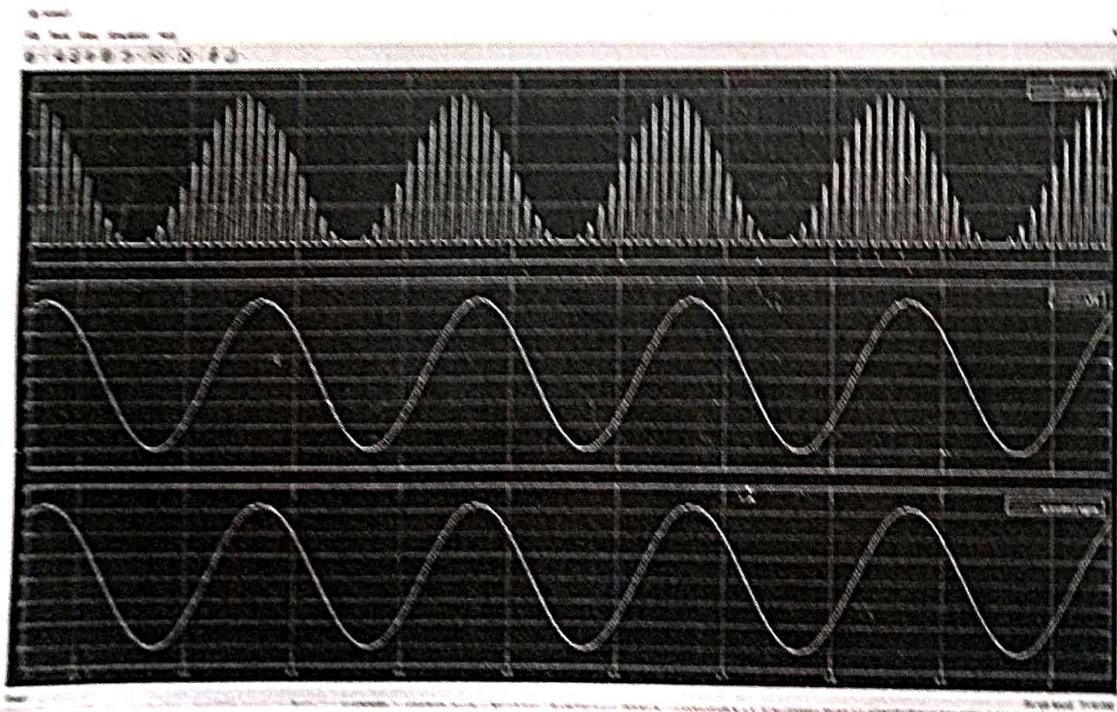
$$e_{Am} = E_c \cos \omega c t + \underbrace{m E_c}_{\text{carrier}} \cos(\omega_c + \omega_m) t + \underbrace{\frac{m E_c}{2} \cos(\omega_c - \omega_m) t}_{\text{LSB}}$$

$$B.W = 2f_m$$

100% modulation



100% Demodulation



CARRIER POWER ( $P_c$ )

$$P_c = \frac{E_c Q}{R}$$

$$P_{USB} = P_{LSB} = \frac{m^2}{4} P_c$$

TOTAL POWER

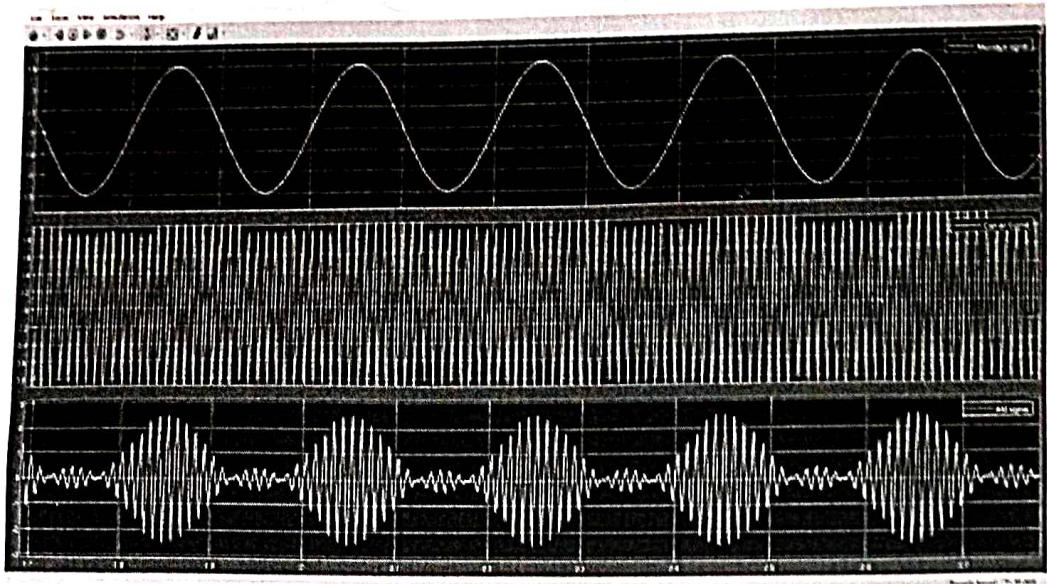
$$P_t = P_c \left[ 1 + \frac{m^2}{2} \right]$$

Procedure:-

1. switch on the computer and click on the MATLAB icon.
2. go to start at the bottom of the command window then select 'simulink' then go to library browser and drag it in to creating file
3. arrange the functional blocks as shown in simulink model.
4. Assign required parameters to each functional block.
5. observe the output on scope

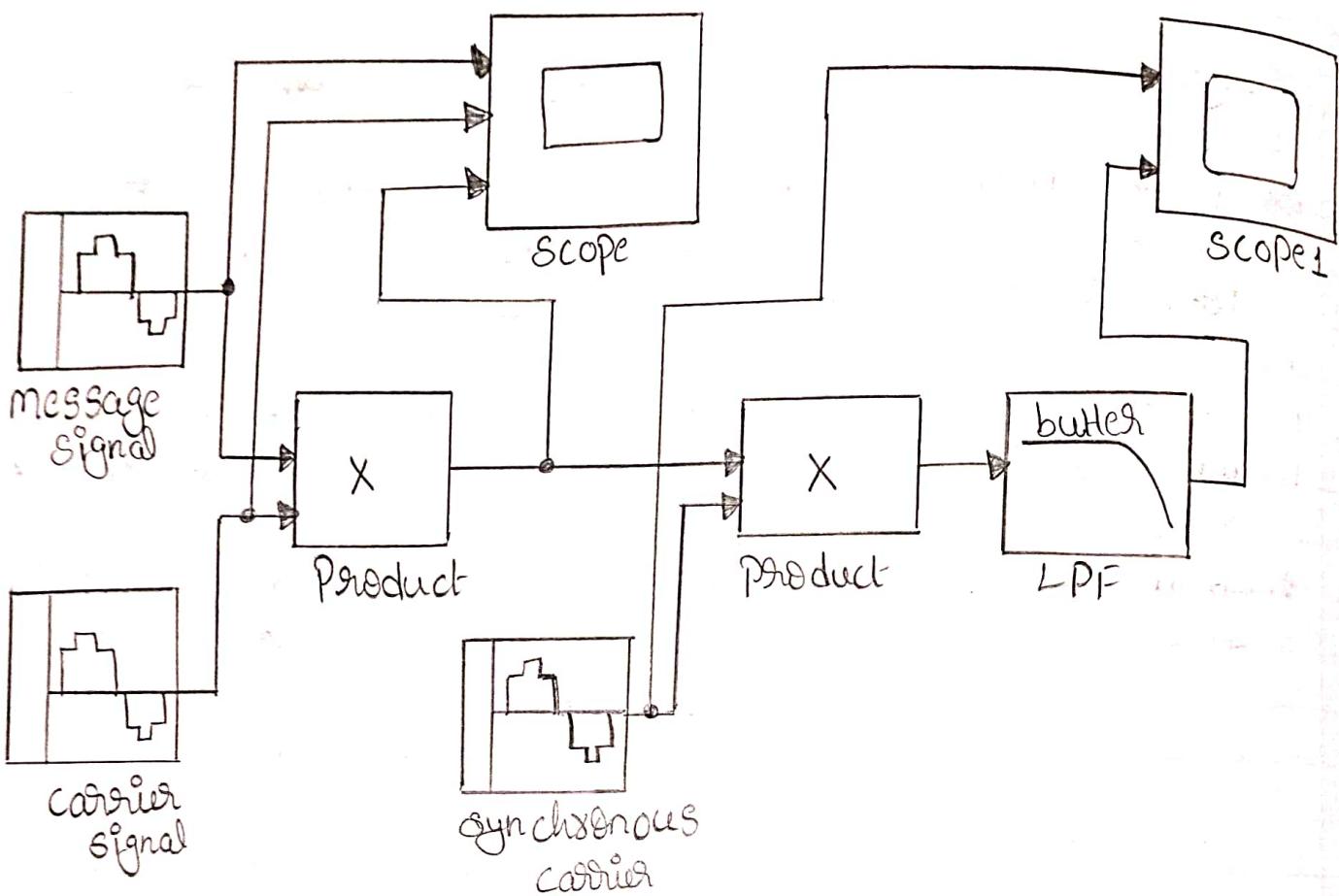
Result:- observed the waveforms of Amplitude modulation for 3 conditions - 50% modulation (under modulation)  
 100% modulation (exact modulation)  
 150% modulation (over modulation) and  
 observed the reconstructed message signal

150% modulation

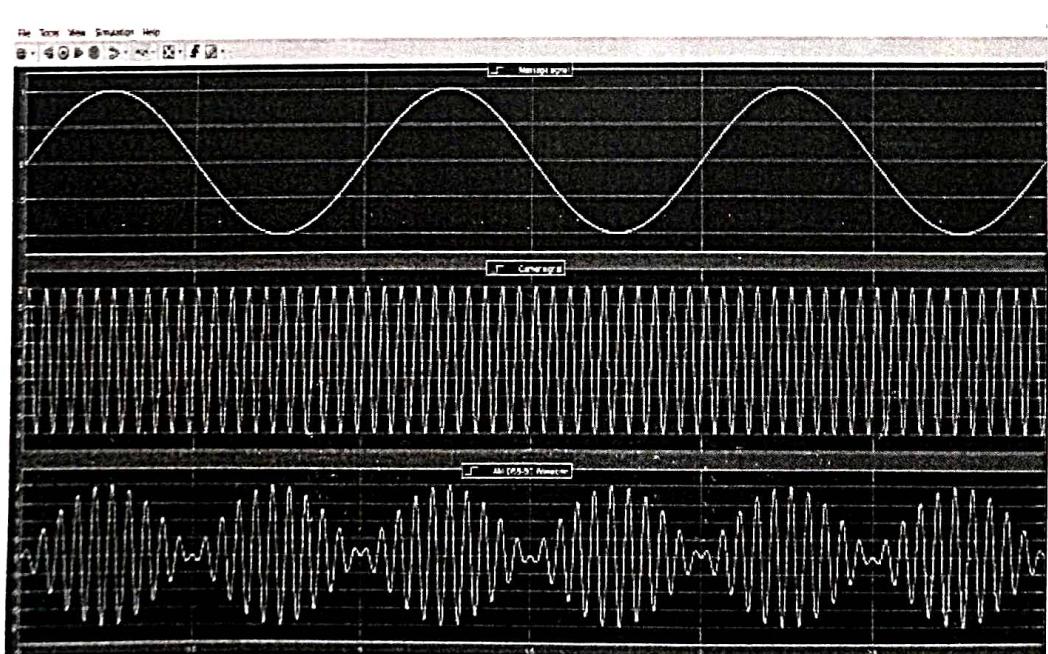


150% De-modulation

## Simulink block diagram



## DSB-SC Modulation



## DSB-SC Modulation & Demodulation

Aim: To Perform the DSB-SC Signal Generation and Detection using simulink.

### Apparatus Required:

- (a) Hardware Tools: computer system
- (b) Software Tool: MATLAB 7.0 and above version

### Theory:

Amplitude modulations characterized by an information bearing carrier amplitude  $A(t)$  that is a linear function of the baseband  $m(t)$ .

At the same time, the carrier frequency  $f_c$  and phase remain constant

If the carrier amplitude is made directly proportional to the modulating signal  $m(t)$ , then modulated signal is

$$m(t) \cos 2\pi f_c t$$

$$m(t) \cos 2\pi f_c t \Rightarrow \frac{1}{2} [m(f-f_c) + m(f+f_c)]$$

→ the modulation process does not introduce a sinusoid at  $f_c$  for this reason it is called double-sideband suppressed carrier (DSB-SC) modulation.

→ To recover the original signal  $m(t)$  from the modulated signal it is necessary to retranslate the spectrum to its original position.

DSB-SC Demodulation

The process of recovering the signal from the modulated signal (retranslating the spectrum to its original position) is referred to as demodulation.

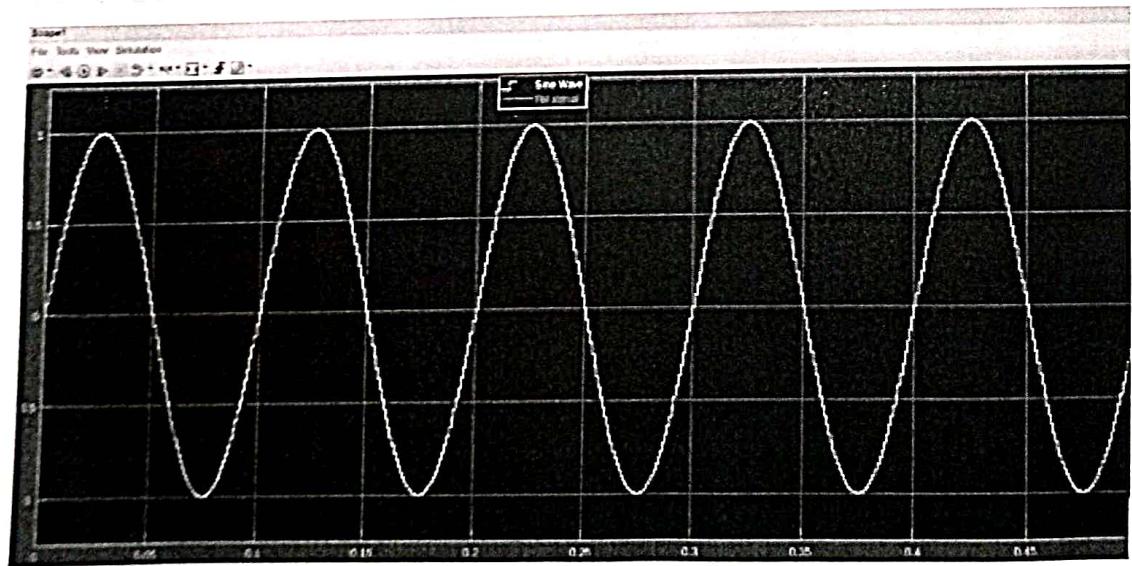
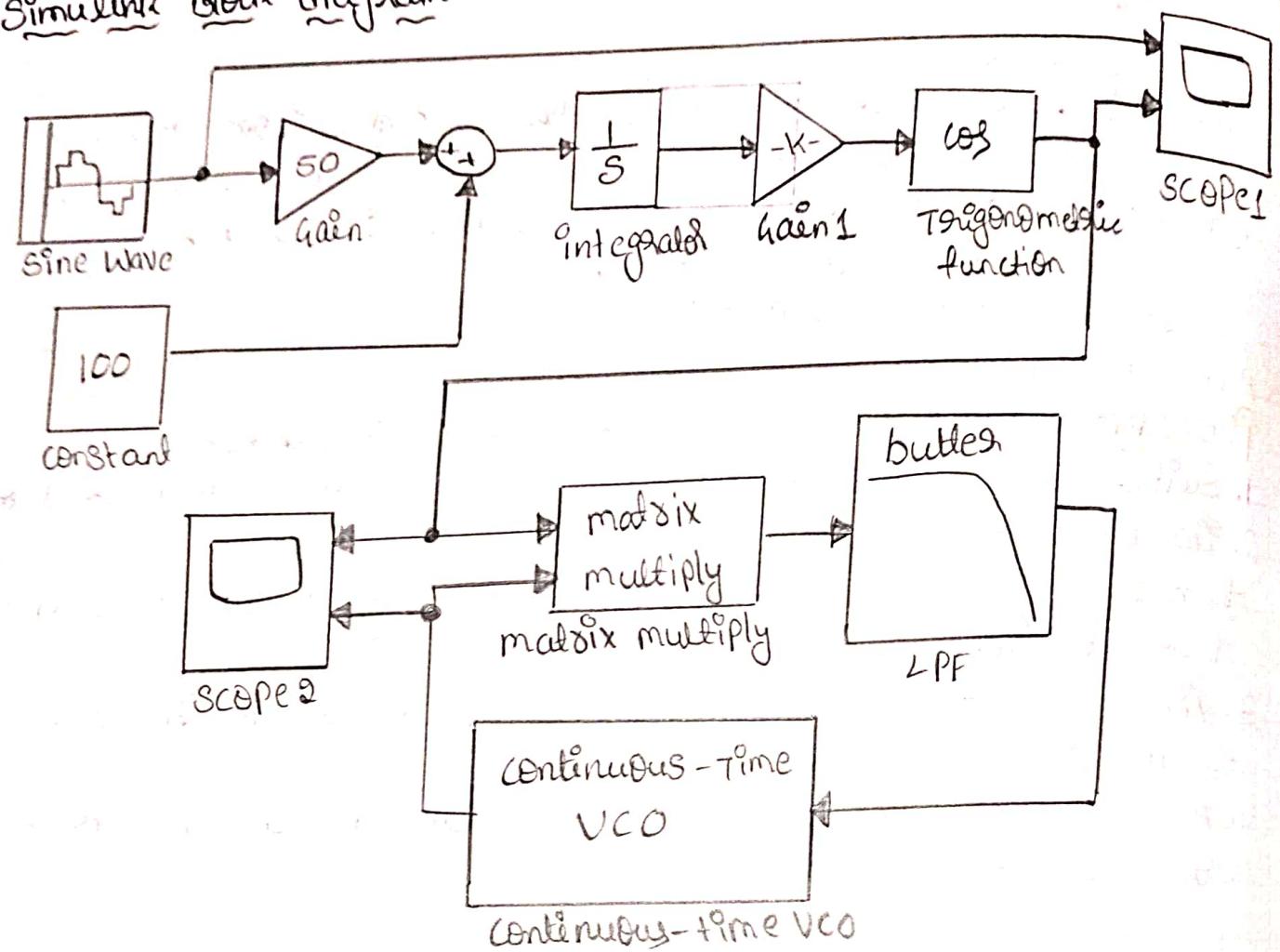
$$f(t) = \frac{1}{2} m(t)$$

### Procedure:-

1. Switch on the computer and click on the MATLAB icon.
2. Go to Start at the bottom of the command window, then select "simulink" then go to library browser and drag it into creating file.
3. Arrange the functional blocks as shown in simulink model
4. Assign required parameters to each functional block
5. Observe the outputs on scope.

Result:-

## Simulink Block diagram



## Frequency modulation and Demodulation

Aim :- To perform the frequency modulation signal generation and detection using simulink.

### Apparatus Required :-

- (a) Hardware Tools : computer system
- (b) Software Tool : MATLAB 7.0 or above version

### Theory :-

Frequency modulation is the process in which information (message signal) is transmitted over a carrier wave by varying its instantaneous frequency. The difference between instantaneous frequency and central frequency of carrier will be directly proportional to the instantaneous value of the amplitude of message signal. Times wired in a table made can be used for generating FM waves.

$$\text{FM equation} \rightarrow V = A \sin [w_c t + (\frac{\Delta f}{f_m}) \sin w_m t]$$

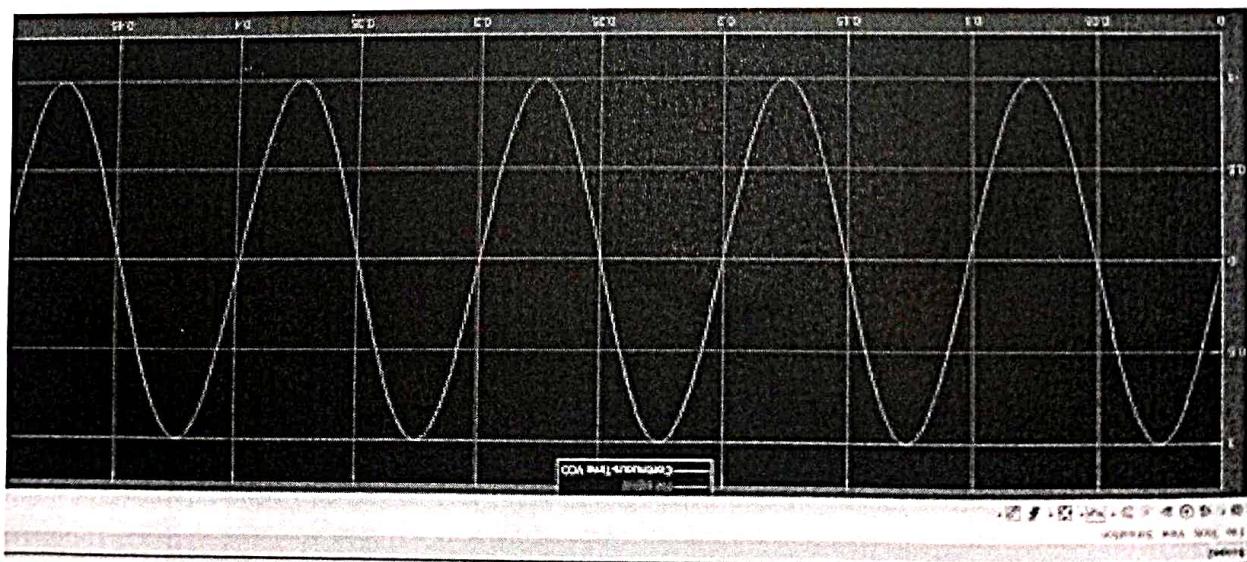
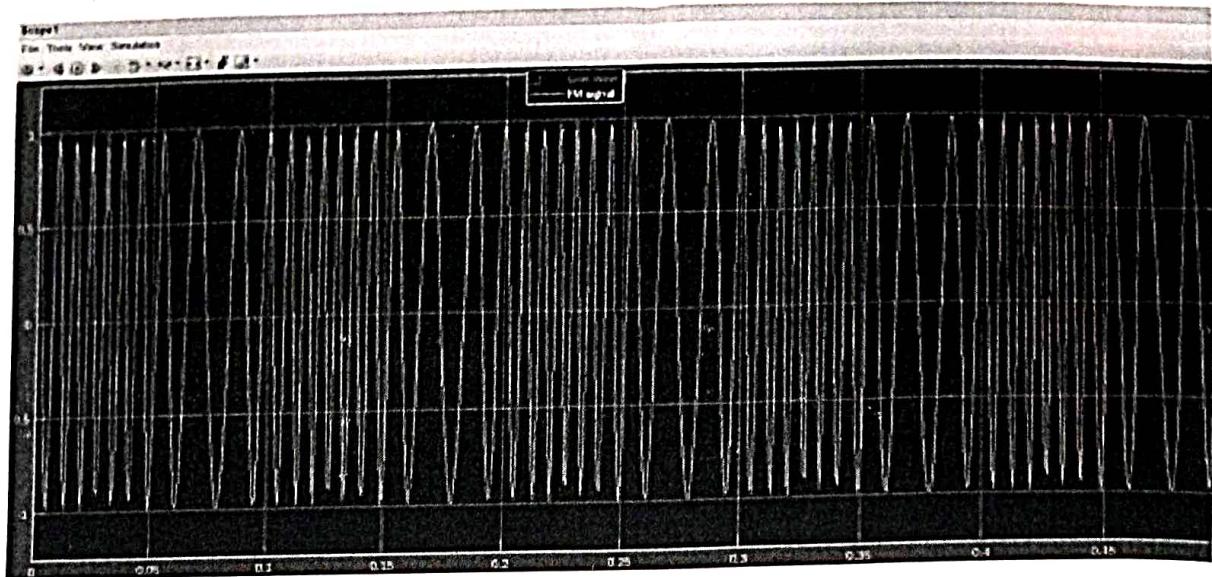
$$V = A \sin [w_c t + m_f \sin w_m t]$$

where  $\rightarrow A$  - Amplitude of the FM signal

$\Delta f$  - frequency deviation

$m_f$  = modulation index of FM

$$m_f = \frac{\Delta f}{f_m}$$

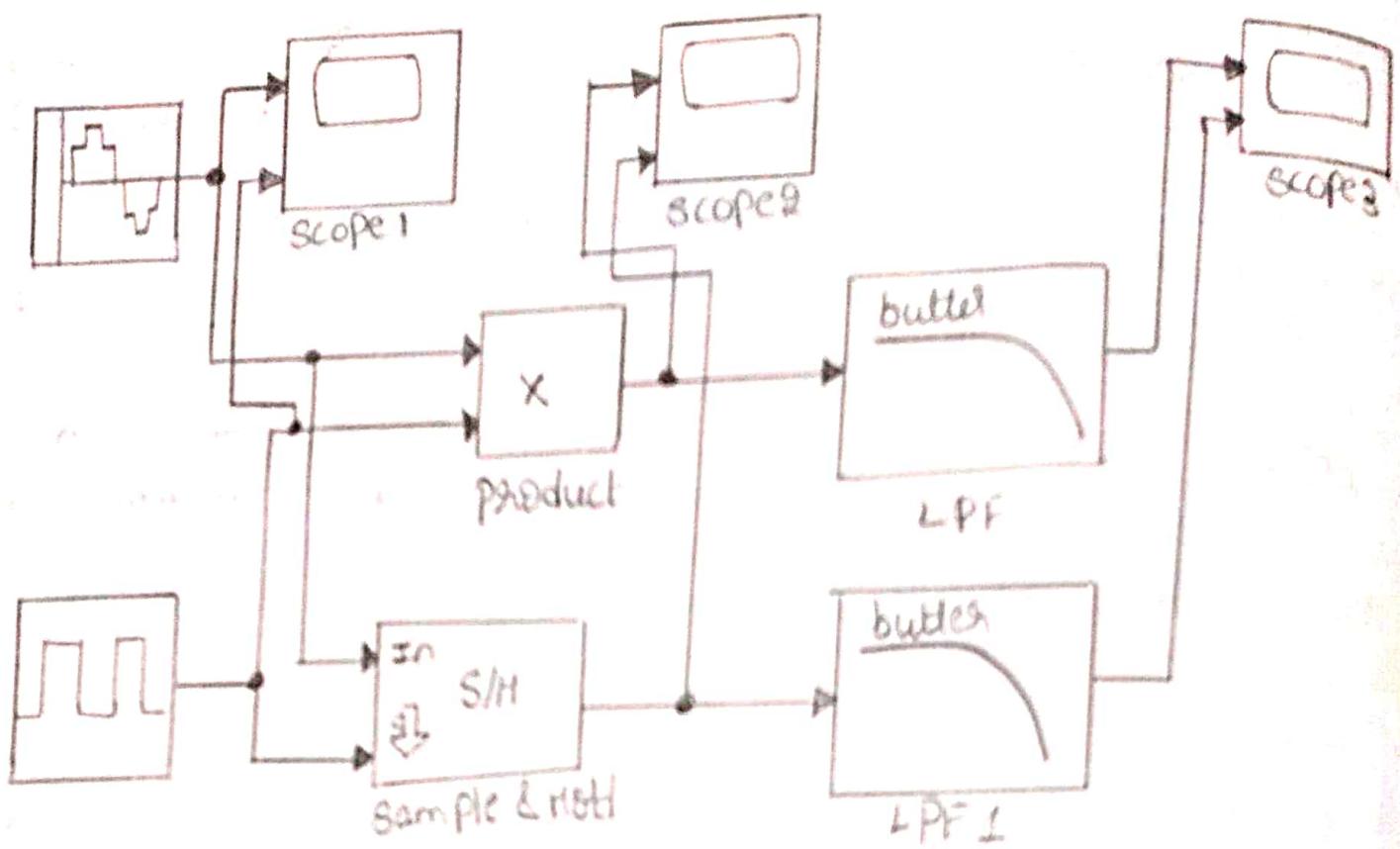


Procedure:

1. Switch on the computer and click on the MATLAB icon.
2. Go to start at the bottom of the command window, then go to library browser and drag it into creating file.
3. Arrange the functional block as in simulink model.
4. Assign required parameters to each functional block.
5. observe the outputs on scope.

Result:

# Simulink Block diagram



## PAM, PWM, PPM and PCM

Aim: To perform PAM, PWM, PPM and PCM using Simulink

### Apparatus Required:

(a) Hardware Tools: Computer System

(b) Software Tool: MATLAB 7.0 & above version

### (1) Pulse Amplitude modulation

Theory: PAM is the simplest pulse modulation scheme. In pulse amplitude modulation system, the amplitude of a carrier pulse train is varied in accordance with the instantaneous level of the modulating signal.

→ pulses can be of a rectangular form or some other appropriate shape

→ In natural PAM, a signal sampled at the Nyquist rate is reconstructed by passing it through an efficient LPF with exact cutoff frequency.

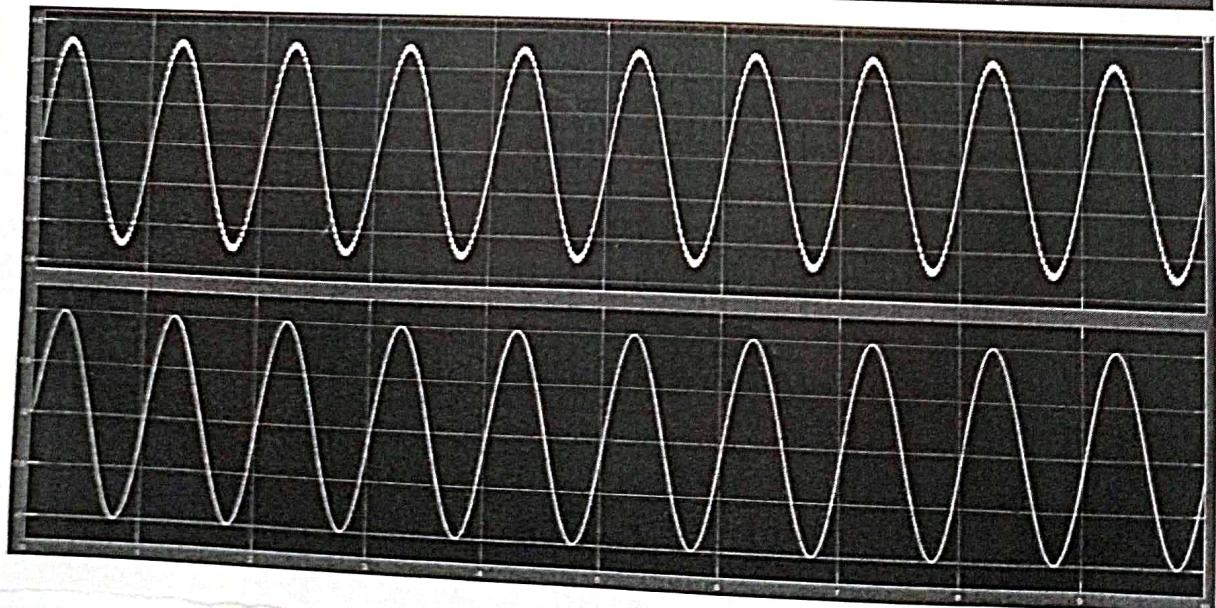
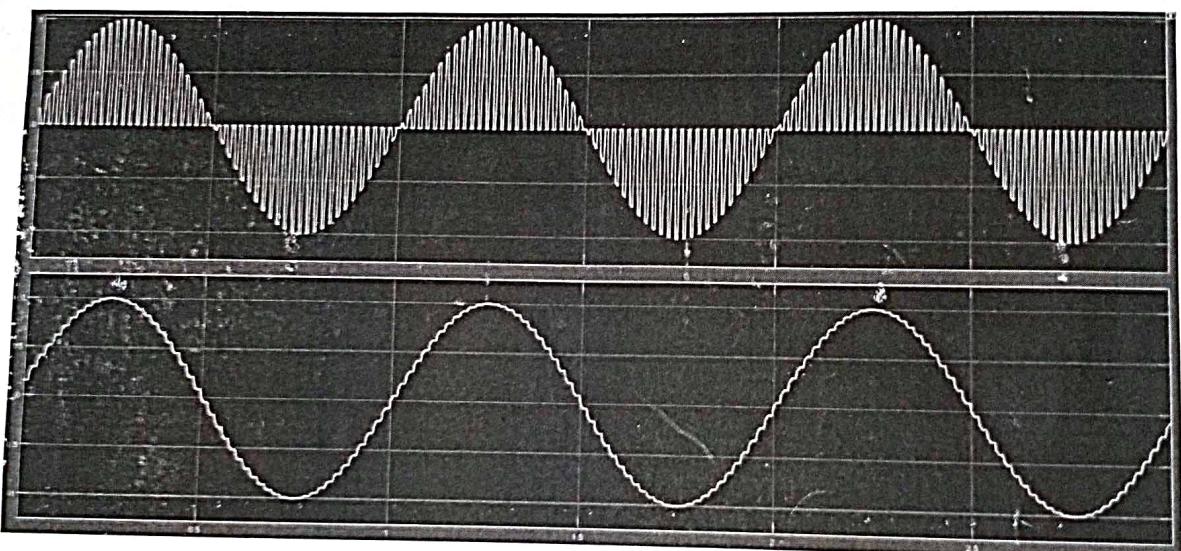
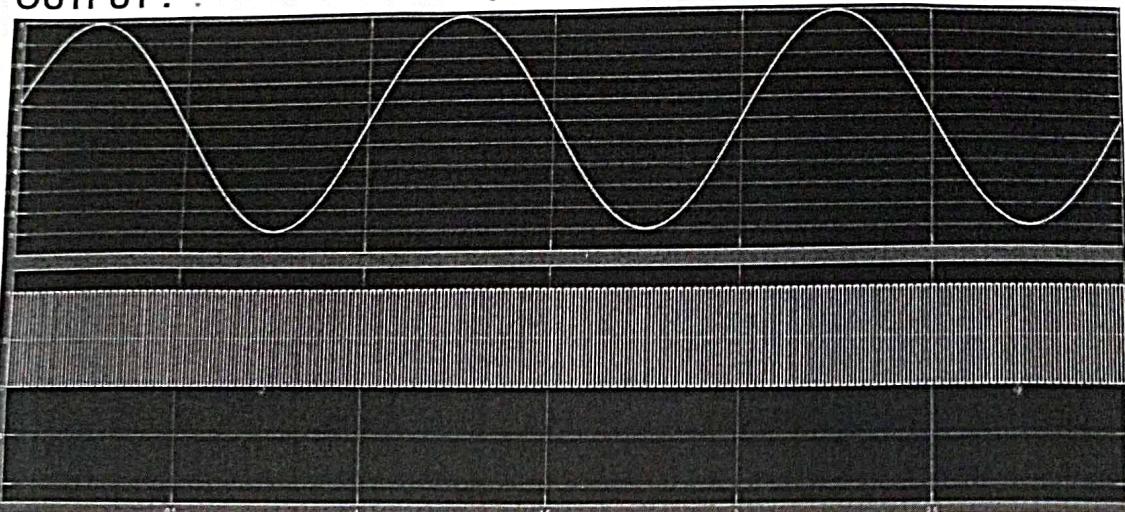
→ though the PAMI signal is passed through an LPF it cannot recover the signal without distortion

→ hence to avoid this noise, flat-top Sampling is done

→ Flat-top Sampling is the process in which sampled signal can be represented in pulses for which the amplitude of the signal can't be changed w.r.t. to the analog signal, to be sampled

→ the tops of amplitude remain flat. This process simplifies the circuit design.

**OUTPUT :**

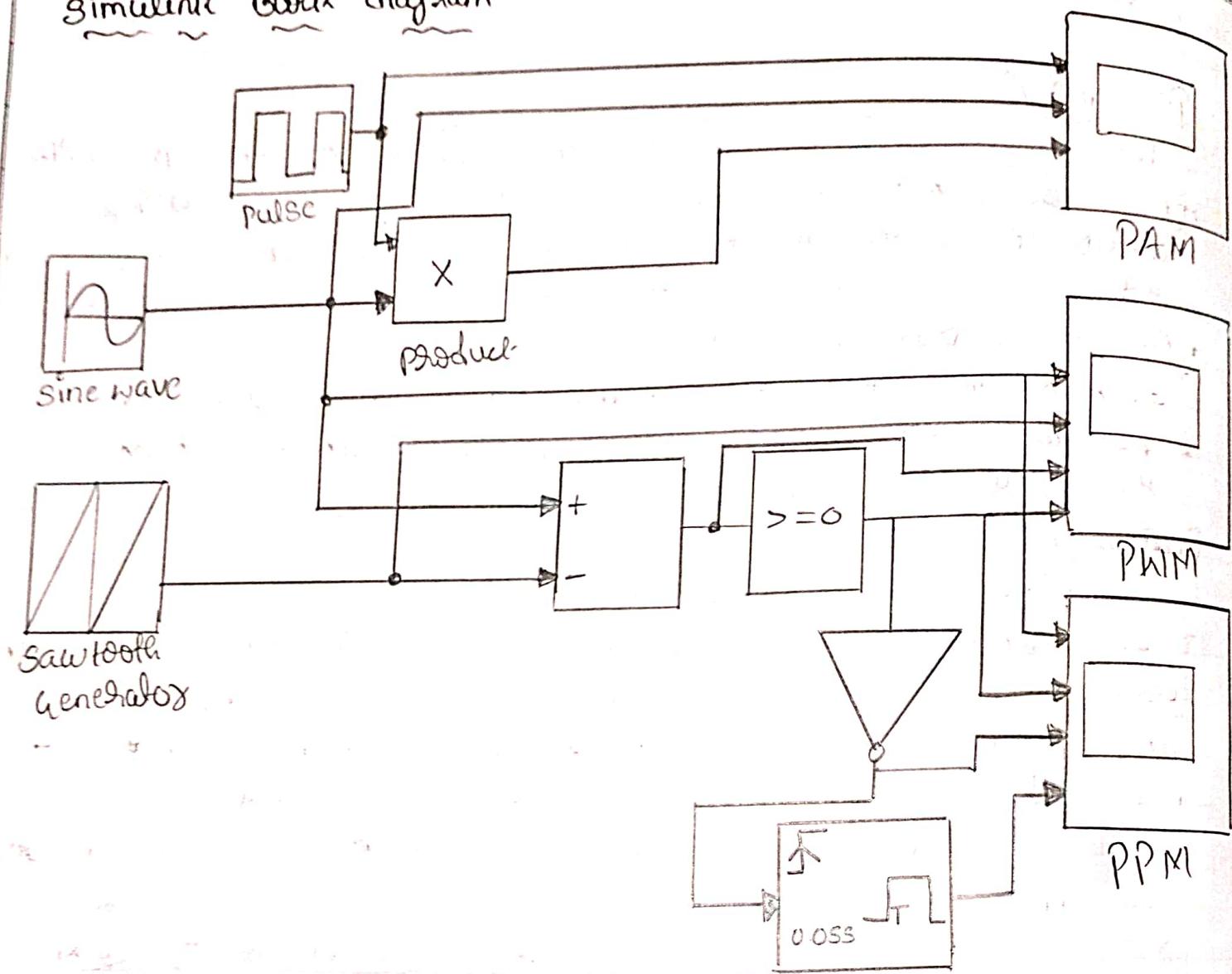


Procedure:

1. switch on the computer and click on the MATLAB icon
2. go to start at the bottom of the command window then select "simulink" then go to library browser and drag it into creating file
3. Arrange the functional blocks as shown in simulink model
4. Assign required parameters to each functional block
5. observe the output on scope

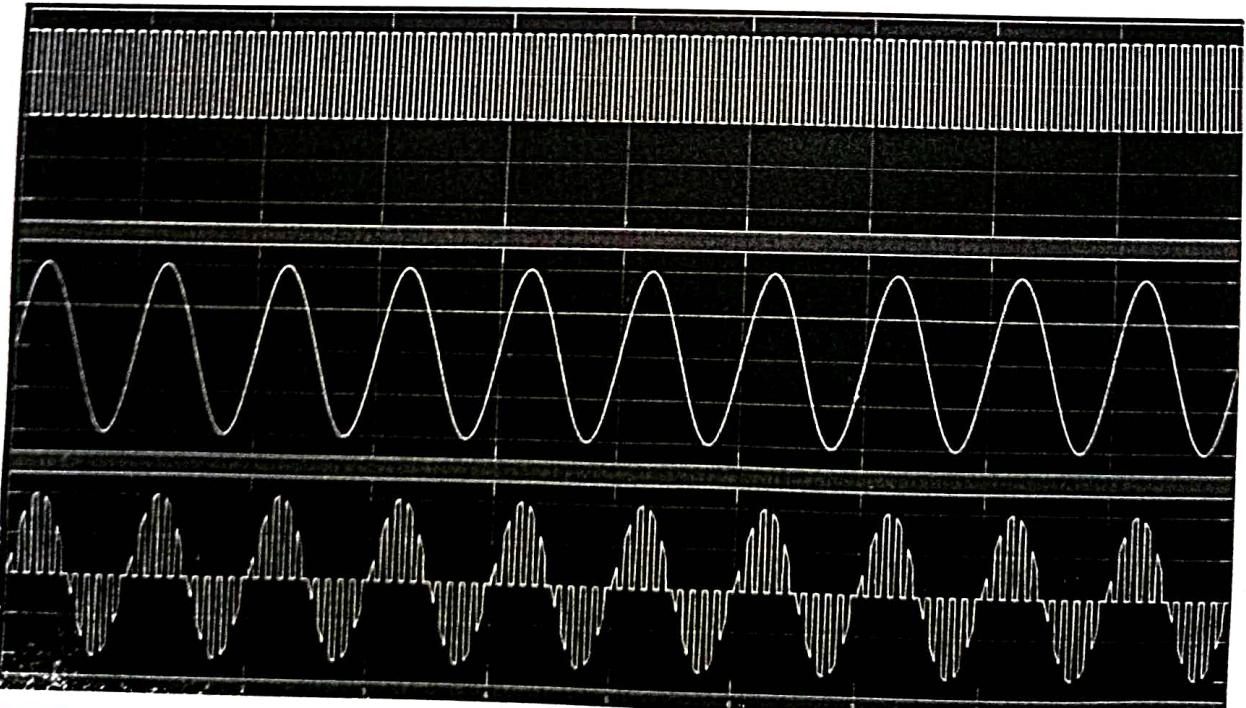
Result:

# Simulink Block diagram



**OUTPUT :**

**PAM DIAGRAM :**



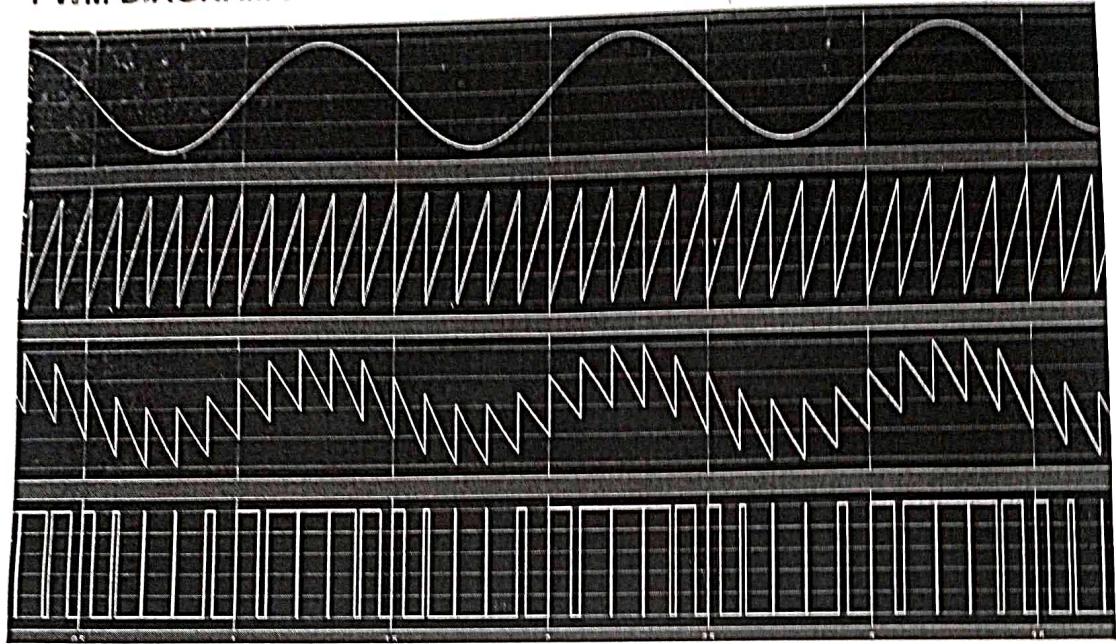
## ⑧ Pulse width and pulse position modulation

- In pulse-duration modulation (PDM), the samples of the message signal are used to vary the duration of the individual pulses.
- This form of modulation is also referred to as pulse-width modulation or pulse-length modulation.
- The modulating signal may vary the time of occurrence of the leading edge, the trailing edge or both edges of the pulse.
- PDM is wasteful of power, in that long pulses expend considerable power during the pulse while bearing the pulse no additional information.
- If this unused power is subtracted from PDM, so that only time transitions are essentially preserved we obtain a more efficient type of pulse modulation known as pulse-position modulation.
- In PPM, the position of a pulse relative to its unmodulated time of occurrence is varied in accordance with the message signal.

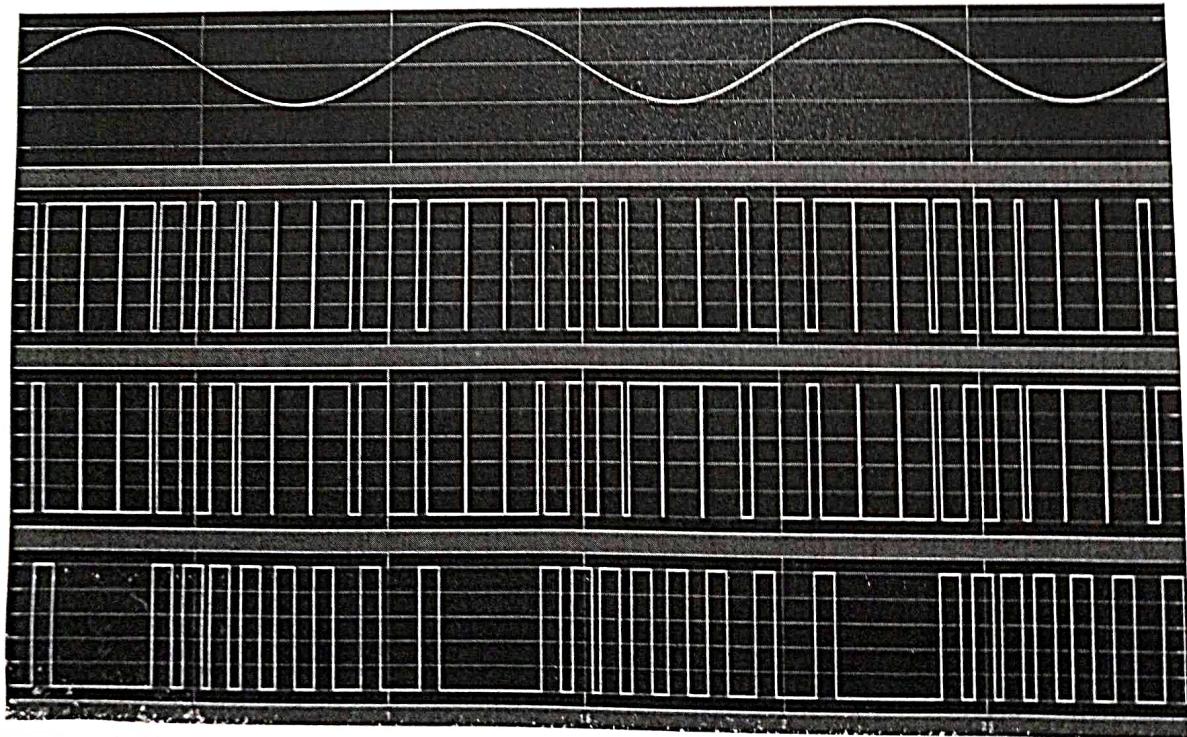
### Procedure:

1. Switch on the computer and click on the MATLAB icon.
2. Go to start at the bottom of the command window then select "simulink" then go to library browser and drag it into creating file
3. Arrange the functional blocks as shown in simulink model.

**PWM DIAGRAM :**



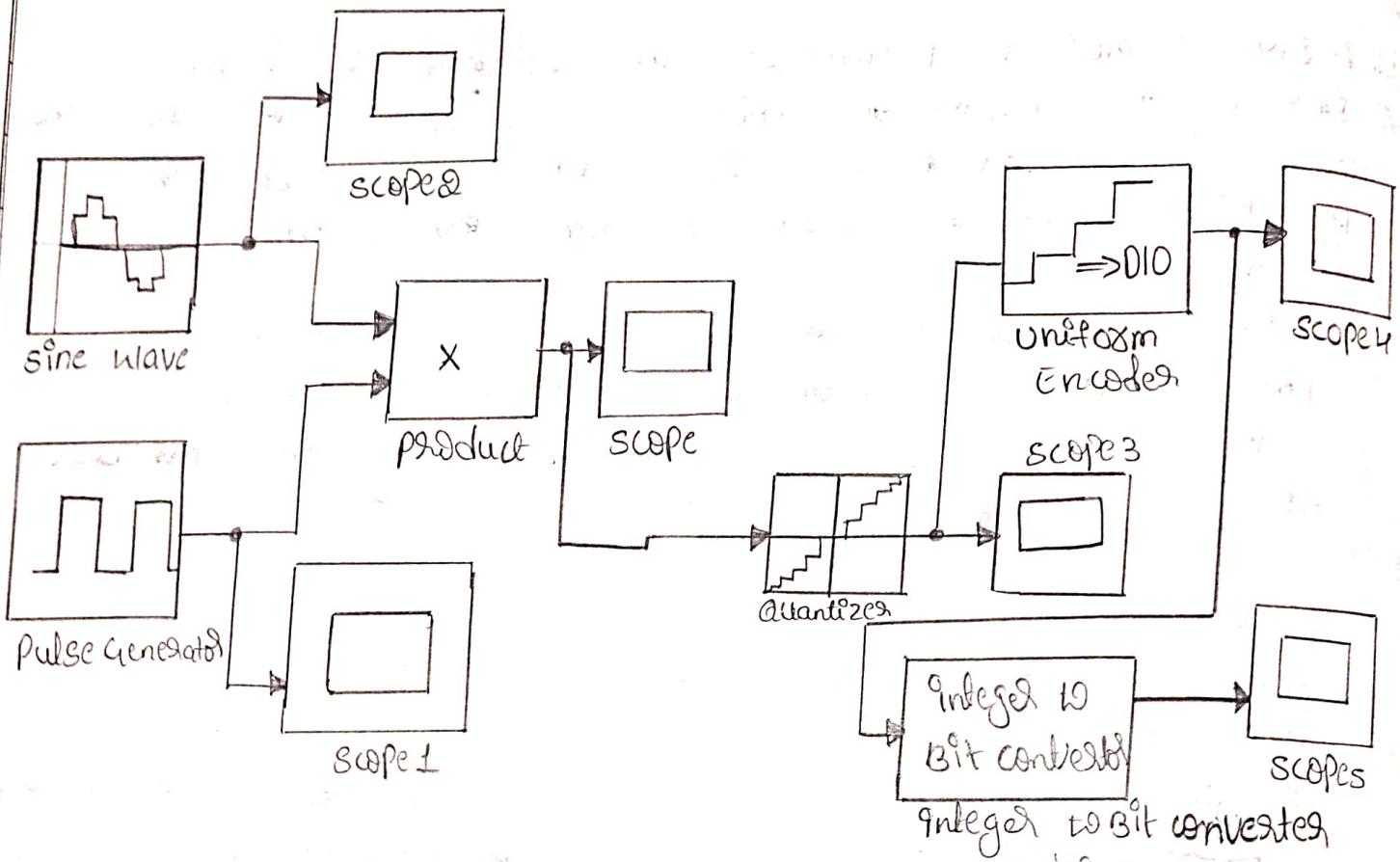
**PPM DIAGRAM :**



4. Assign required parameters to each functional block
5. Observe the outputs on scope.

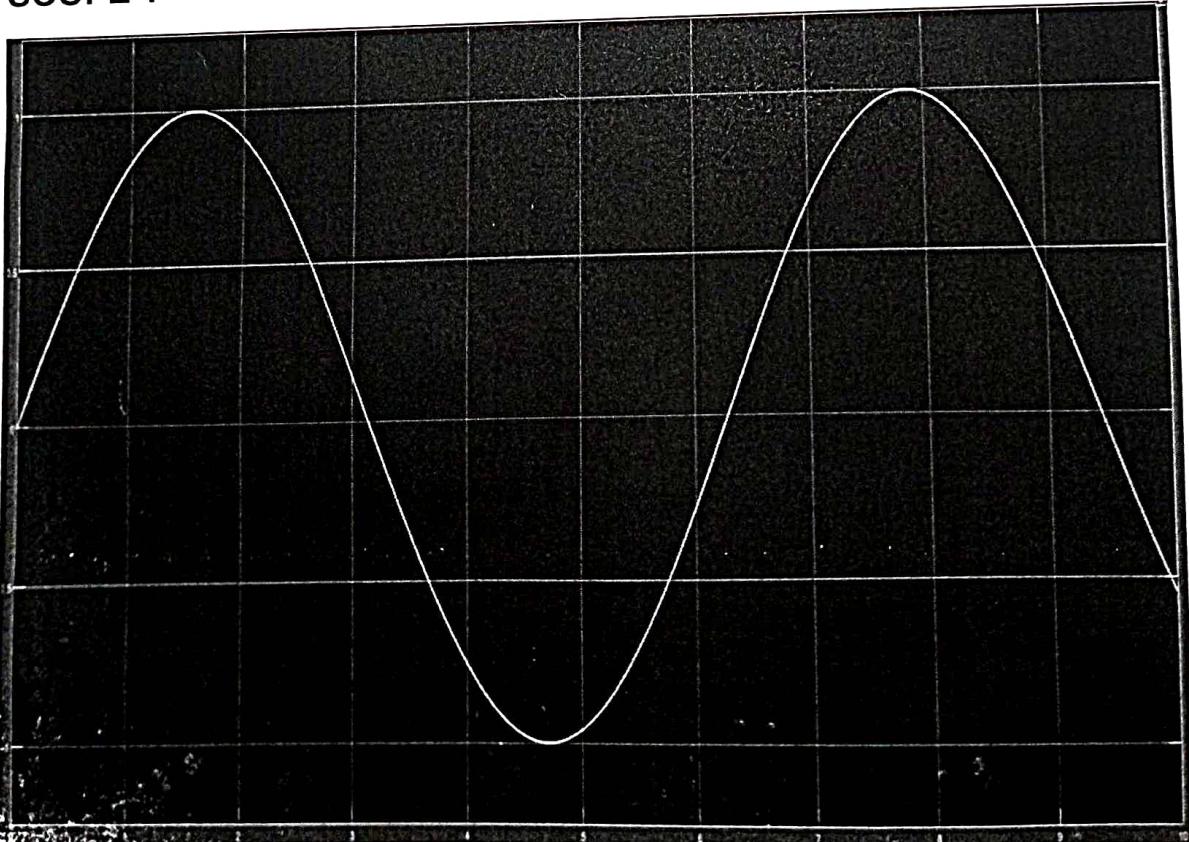
Result:

## Simulink block diagram



**OUTPUT :**

**SCOPE 1**



### ③ Pulse code modulation

PCM basically is a tool for converting an analog signal into a digital signal (A/D conversion)

→ An analog signal is characterized by an amplitude that can take on any value over a continuous range. This means that it can take on an infinite no<sup>n</sup> of values

→ Digital signal amplitude can take on only a finite no<sup>n</sup> of values

→ An analog signal can be converted into a digital signal by means of sampling and quantizing, that is rounding off its value to one of the ~~def~~ permissible numbers of quantized levels

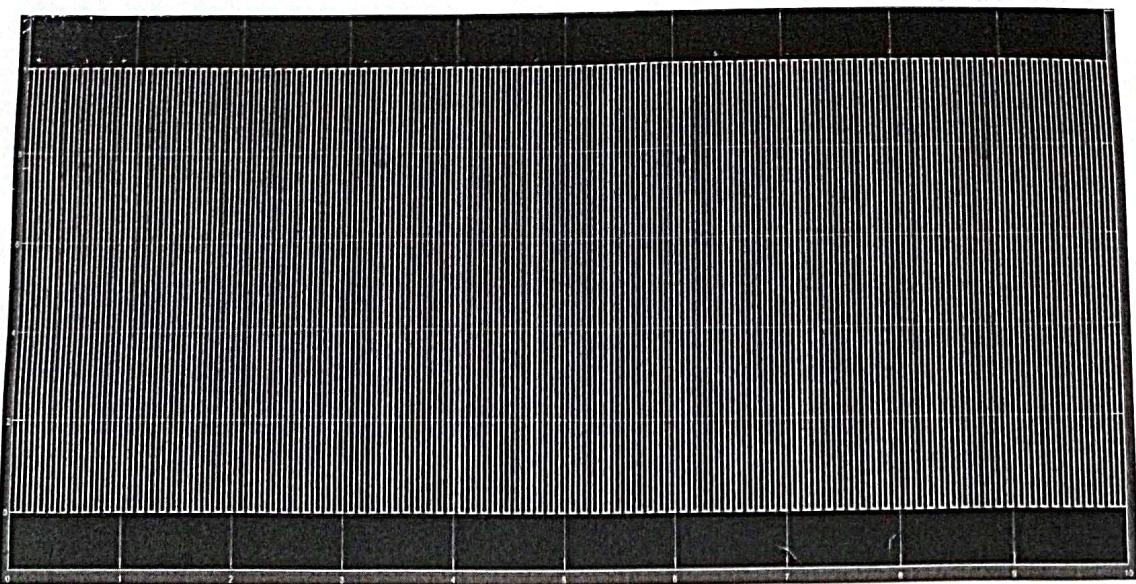
→ The amplitude of the analog signal  $m(t)$  lie in the range  $(-m_p, m_p)$ , which is partitioned into  $L$  subintervals, each of magnitude  $\Delta V = 2m_p/L$ .

→ Next, each sample amplitude is approximated by the midpoint value of the subinterval in which the sample falls. Each sample is now approximated to one of the  $L$  numbers

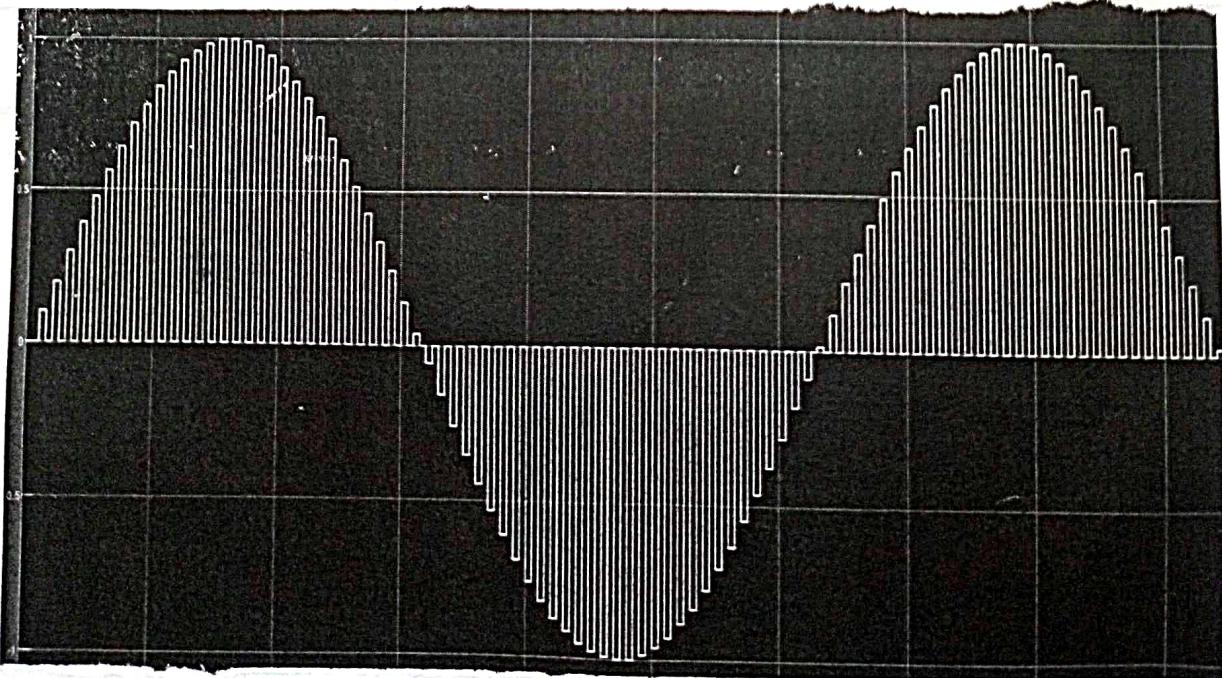
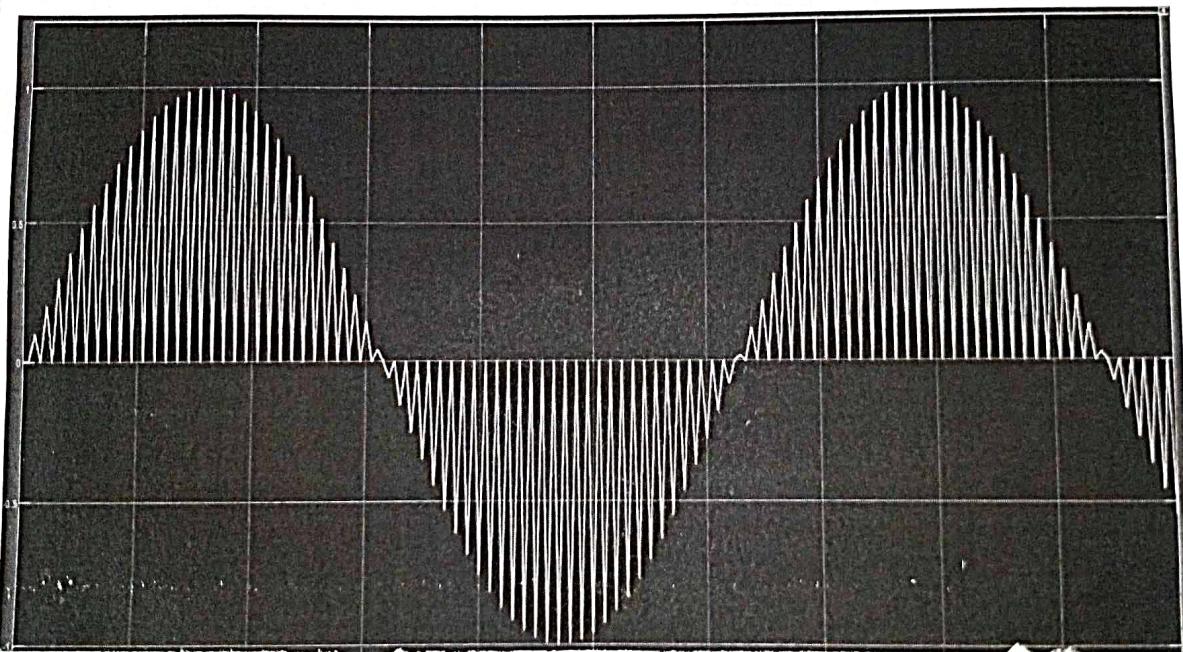
### Procedure:-

1. Switch on the computer and click on the MATLAB icon
2. Go to start at the bottom of the command window then select 'simulink' then go to library browser and drag it into clearing file
3. Arrange the functional blocks as shown in simulink model
4. Assign required parameters to each functional block
5. Observe the outputs on scope

**SCOPE 2**



**SCOPE 3**



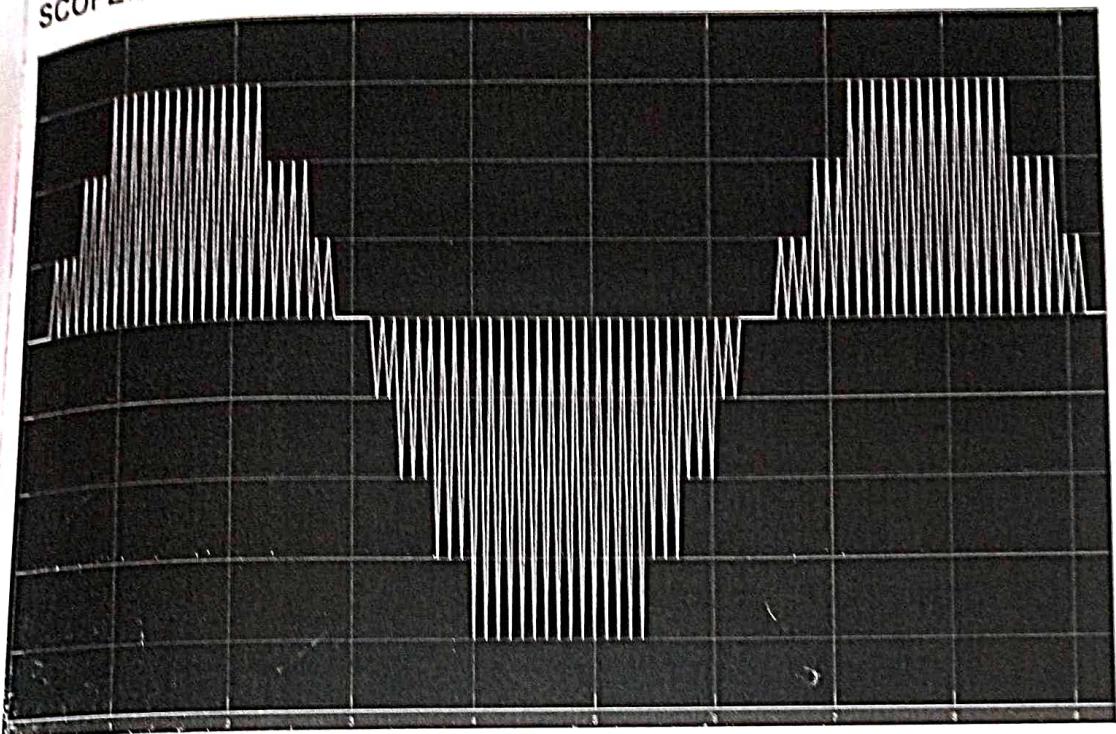
Date \_\_\_\_\_

Expt. No. \_\_\_\_\_

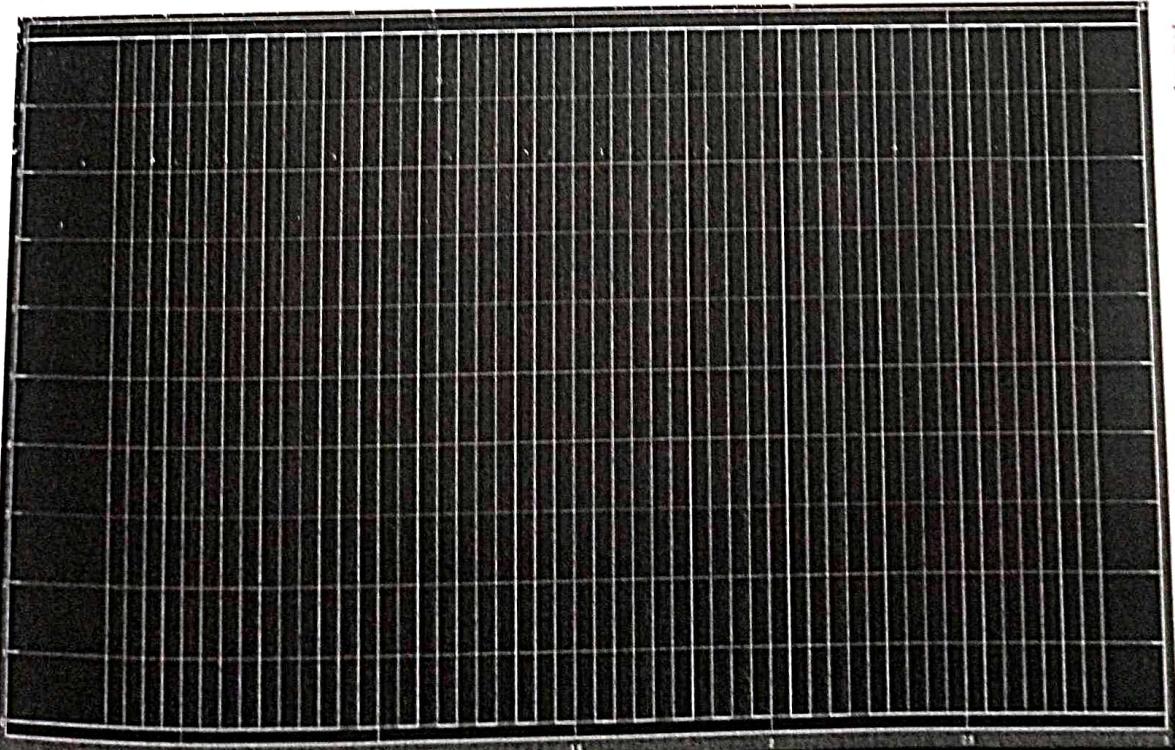
Page No. \_\_\_\_\_

Result &

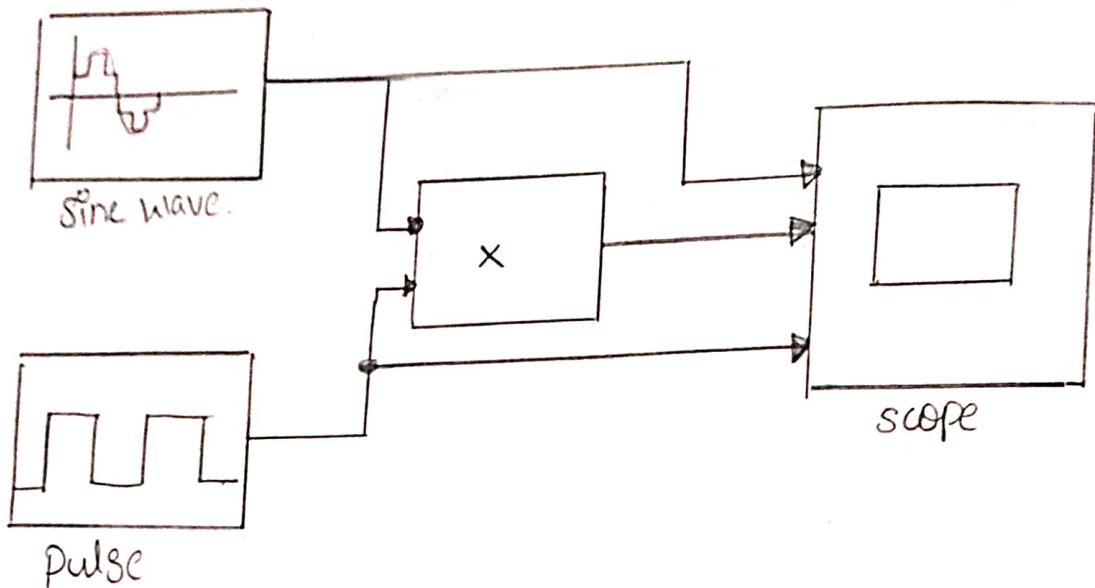
SCOPE 5



SCOPE 6



Block Diagram  
Amplitude shift Keying modulation



Amplitude shift Keying Demodulation using MATLAB code

```

clc;
clear all;
close all;
Tb=1; fc=10; % Carrier Signal.
t=0:Tb:1;
C=sqrt(100*(B/Tb))*sin(2*pi*fc*t);
% generation message signal.
N=8;
m=rand(1,N);
t1=0; t2=Tb
for i=N:N
    t=[t1:.01:t2];
    if m(i)==1;
        m_s=ones(1,length(t));
    else
        m(i)=0;
        m_s=ones(1,length(t));
    end
end
  
```

## (a) Amplitude Shift Keying modulation

Aim: To study Amplitude Shift Keying (modulation & demodulation)

Apparatus Required:-

(a) Hardware Tools: Computer System

(b) Software Tool: MATLAB 7.0 and above version.

### Theory :-

Amplitude Shift Keying (ASK) is a digital modulation scheme where the binary data is transmitted using a carrier signal with two different amplitude levels.

for binary 0 and 1, the carrier switches b/w these two levels. In its simplest form, a carrier is sent during one symbol and no carrier is sent during the other. This kind of modulation scheme is called On-Off Keying.

### BASIC

$$b(t) = \begin{cases} \sqrt{E_b}, & \text{for binary symbol 1} \\ 0, & \text{for binary symbol 0} \end{cases}$$

$$s(t) = \begin{cases} \sqrt{\frac{Q}{E_b}} \cos(2\pi f_c t), & \text{for symbol 1} \\ 0, & \text{for symbol 0} \end{cases}$$

### Procedure :-

→ Open MATLAB

→ Open new file in Simulink

→ Connect blocks for modulation.

→ Write code for Demodulation.

→ Observe wave forms.

```

message(i,:) = m_s;
% product of carrier and message
ask_sig(i,:) = C.*m_s;
t1 = t1 + (Tb+.01);
t2 = t2 + (Tb+.01);
% plot the message and ASK signal.
subplot(5,1,2); axis([0 N-2 2 3]); plot(t,message(i,:));
title('message signal'); xlabel('t-->'); ylabel('m(t)'); grid on
hold on
subplot(5,1,4); plot(t,ask_sig(i,:));
title('message signal'); xlabel('t-->'); ylabel('S(t)'); grid on
hold on
and
hold off
subplot(5,1,3); plot(t,C);
title('carrier signal'); xlabel('t-->'); ylabel('C(t)'); grid on
subplot(5,1,1); stem(m);
title('binary data bits'); xlabel('n-->'); ylabel('b(n)'); grid on
% ASK Demodulation
t1=0; t2=Tb
for i=1:N
    t = [t1 : Tb/100 : t2]
    x=Sum(C.*ask_sig(i,:)); % correlator.
    if x>0 % decision device
        demod(i)=1;
    else
        demod(i)=0;
    end
    t1=t1+(Tb+.01);
    t2=t2+(Tb+.01);
end
subplot(5,1,5); stem(demod);
title('ASK demodulated signal'); xlabel('n-->'); ylabel('b(n)'); grid on

```

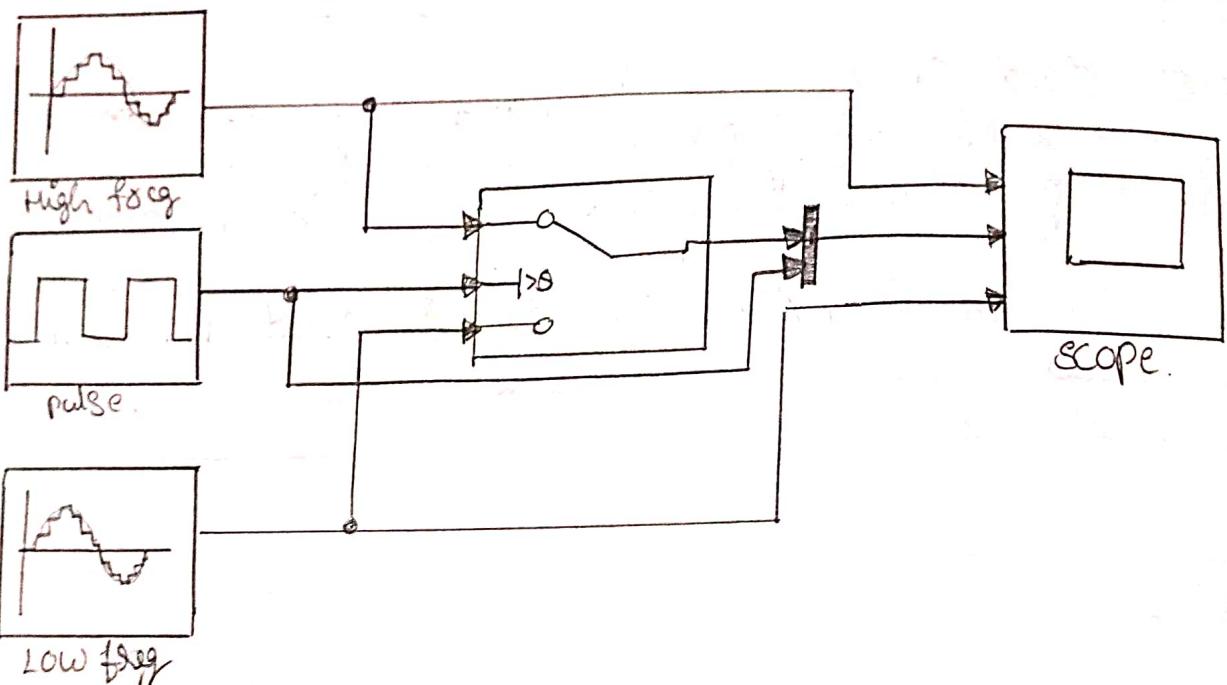
### (b) Amplitude Shift Keying Demodulation

→ BASK signal is generated by using a product modulator with two inputs. One input, the on/off signal i.e. is the modulating signal the sinusoidal carrier wave  $c(t)$ .

→ Detection of the BASK wave is concerned, the simplest is way to use an envelope detector. exploiting the nonconstant envelope property of BASK signal.

Result:-

Block diagram :-  
modulation



code for Demodulation:-

```

clc;
clear all;
close all;

% Generate carrier Signal
Tb=1; fc1=2; fc2=5;
t=0:(Tb/100):Tb;
c1=sqr((2/Tb))*sin(2*pi*fc1*t);
c2=sqr((2/Tb))*sin(2*pi*fc2*t);

% generate message Signal
N=8
m=rand(1,N);
t1=0; t2=Tb
for i=1:N
    t=[t1:(Tb/100):t2];
    if m(i)>0.5
        m(i)=1;
    else
        m(i)=0;
    end
end

```

## Frequency shift keying

Aim: To study frequency shift keying modulation and demodulation.

### Apparatus required:-

(a) Hardware Tools: computer system

(b) Software Tools: MATLAB 7.0 and above version

### Theory:-

Binary frequency-shift keying (BFSK): In which the carrier amplitude and carrier phase are both maintained constant while the carrier frequency is keyed bw the two possible values used to represent symbol 0 and 1.

$$s_{it} = \begin{cases} \sqrt{\frac{Q E_b}{T_b}} \cos(2\pi f_1 t), & \text{for symbol 1 corresponding } i=1 \\ \sqrt{\frac{Q E_b}{T_b}} \cos(2\pi f_2 t), & \text{for symbol 0 corresponding } i=0 \end{cases}$$

→ where  $E_b$  is the transmitted signal energy per bit when the frequencies  $f_1$  and  $f_2$  are chosen in such a way that they differ from each other by an amount equal to the reciprocal of the bit duration the BFSK signal is referred to as Sunde's BFSK after its originality.

→ This modulated signal is a continuous phase signal in the sense that phase continuity is always maintained including the inter-bit switching times.

Teacher's Signature \_\_\_\_\_

```

m_B = ones(1, length(t));
In_Vin_B = ones(1, length(t));
end
message(i,:) = In_Vin_B;
% multiply
fSK_Sig1(i,:) = C1.*m_B;
fSK_Sig2(i,:) = C2.*m_B;
PSK_Sig = fSK_Sig1 + fSK_Sig2;
% plotting the message signal and the modulated signal
subplot(3,2,2); axis([0 N-2 0]); plot(t, message(i,:),'r');
title('message signal'); xlabel('t-->'); ylabel('m(t)'); grid on;
subplot(3,2,3); plot(t, PSK_Sig(i,:));
title('FSK signal'); xlabel('t-->'); ylabel('s(t)'); grid on;
t1 = t1 + (Tb+0.01); tQ = tQ + (Tb+0.01);
end
hold off
% plotting binary data bits and carrier signal
subplot(3,2,1); stem(m);
title('Binary data'); xlabel('n-->'); ylabel('b1(n)');
subplot(3,2,2); plot(t, c1);
title('carrier signal-1'); xlabel('t-->'); ylabel('c1(t)'); grid on;
subplot(3,2,3); plot(t, cQ);
title('carrier signal-2'); xlabel('t-->'); ylabel('cQ(t)');
grid on;
% FSK Demodulation
t1=0; tQ=Tb
for q=1:N
    t=[t1:(Tb/100):tQ]
    % Correlator
    X1 = sum(c1.*fSK_Sig1(i,:));
    XQ = sum(cQ.*fSK_Sig2(i,:));

```

Procedure:-

open MATLAB

open new m-file in MATLAB

write code for Demodulation

open Simulink software

connect blocks for modulation circuit

observe wave forms

Result:-

$x = x_1 - x_8$ ;

% decision device

if  $x > 0$

demod(i) = 1;

else

demod(i) = 0;

end

$t_1 = t_1 + (\tau_b + 0.01)$ ;

$t_8 = t_8 + (\tau_b + 0.01)$ ;

end

% plotting the demodulated data bits

subplot(3,8,6); stem(demod);

title('demodulated data'); xlabel('n-->'); ylabel('b(n)');

grid on;

## Phase shift keying

Aim: To generate and demodulate phase shift keyed (PSK) signal using MATLAB

### Apparatus Required:

- (a) Hardware Tools: computer system
- (b) Software Tool: MATLAB 7.0 and above version

### Theory:

#### (a) Binary Phase-Shift Keying (BPSK)

In BPSK, the pair of signals is used to represent symbols 1 and 0, respectively, are defined by

$$s_i(t) = \begin{cases} \sqrt{\frac{QEb}{Tb}} \cos(\omega_n f_c t), & \text{for symbol } i \text{ corresponding to } i=1 \\ \sqrt{\frac{QEb}{Tb}} \cos(\omega_n f_c t + \pi), & \text{for symbol } i \text{ corresponding to } i=0 \end{cases}$$

$$= -\sqrt{\frac{QEb}{Tb}} \cos(\omega_n f_c t)$$

→  $T_b$  denoting the bit duration and

→  $E_b$  denoting the transmitted signal energy per bit

## (b) Quadrature-Shift Keying (QPSK)

→ An important goal of digital communication is the efficient utilization of channel bandwidth.

→ This goal is attained by a bandwidth-conserving modulation scheme known as QPSK.

$$s_i(t) = \begin{cases} \sqrt{\frac{Q_E}{T}} \cos(2\pi f_c t + (Q_i - 1)\frac{\pi}{4}), & 0 \leq t \leq T \\ 0, & \text{elsewhere} \end{cases}$$

### Procedure:

open MATLAB

open new file

Type program

Save in current directory

compile and run the program

for the output see command window / figure window

### Code for BPSK

% BPSK modulation

cic;

clear all;

close all;

generate carrier signal

Tb=1;

$t = 0 : Tb / 100 : Tb$ ;

$f_c = 8$

$$c = \sin(\omega t + \phi) + \sin(\omega t + \phi + \pi/2)$$

% generate message signal

$N = 8$

$m = rand(1, N);$

$t_1 = 0; t_2 = Tb$

for  $i = 1 : N$

$t = [t_1 : 0.1 : t_2]$

if  $m(i) > 0.5$

$m(i) = 1;$

$m_s = ones(1, length(t));$

else

$m(i) = 0$

$m_s = -1 * ones(1, length(t));$

end

message( $i, : ) = m_s;$

% Product of carrier and message signal

bPSK\_sig( $i, : ) = c * m_s;$

% plot the message and bPSK modulated signal

subplot(5, 1, 2); axis([0 N-8 8]); plot(t, message( $i, : ), 'r');$

title('message signal (POLAR form)'); xlabel('t-->'); ylabel('m(t)');

grid on; hold on;

subplot(5, 1, 4); plot(t, bPSK\_sig( $i, : ));$

Title('bPSK signal'); xlabel('t-->'); ylabel('S(t)');

grid on; hold on;

$t_1 = t_1 + 1.01; t_2 = t_2 + 1.01;$

end

hold off

% plot the i/p binary data and carrier signal

subplot (5,1,1); stem(m);

title ('binary data bits'); xlabel ('n-->'); ylabel ('b(n)');

grid on;

subplot (5,1,3); plot(t,c);

title ('carrier signal'); xlabel ('t-->'); ylabel ('c(t)');

grid on;

% BPSK Demodulation

$t_1 = 0$ ;  $t_2 = Tb$

for  $i = 1:N$

$t = [t \cdot 0.01 \cdot t_2]$

% correlator

$x = \text{sum}(c \cdot b_{\text{PSK}} \text{ sig}(i, :))$ ;

% decision device

If  $x > 0$

demod(i) = 1;

else

demod(i) = 0;

end

$t_1 = t_1 + 1.01$ ;

$t_2 = t_2 + 1.01$ ;

end

% plot the demodulated data bits

subplot (5,1,5); stem(demod);

title ('Demodulated'); xlabel ('n-->'); ylabel ('b(n)');

grid on;

## code for QPSK

% QPSK modulation

clc;

clear all;

close all;

% generate QPSK carrier signal

$$T_b = 1; t = 0 : (T_b/100) : T_b; f_c = 1;$$

$$c_1 = \cos((\theta/T_b) * \cos(\theta * p_1 * f_c * t));$$

$$c_2 = \sin((\theta/T_b) * \sin(\theta * p_1 * f_c * t));$$

% generate message signal

$$N = 8; m = rand(1, N);$$

$$t_1 = 0 : T_b = T_b$$

$$\text{for } i = 1 : N - 1$$

$$t = [t_1 : (T_b/100) : t_2]$$

if  $m(i) > 0.5$

$$m(i) = 1;$$

$$m_s = \text{ones}(1, \text{length}(t));$$

else

$$m(i) = 0;$$

$$m_s = -1 * \text{ones}(1, \text{length}(t));$$

end

% odd bits modulated signal

$$\text{odd\_sig}(i, :) = c_1 * m_s;$$

if  $m(i+1) > 0.5$  is

$$m(i+1) = 1;$$

$$m_s = \text{ones}(1, \text{length}(t));$$

else

$$m(i+1) = 0;$$

% decision device

if  $(x_1 > 0 \text{ } \& \& \text{ } x_2 > 0)$

demod(i) = 1;

demod(i+1) = 1;

else if  $(x_1 > 0 \text{ } \& \& \text{ } x_2 < 0)$

demod(i) = 1;

demod(i+1) = 0

else if  $(x_1 < 0 \text{ } \& \& \text{ } x_2 < 0)$

demod(i) = 0;

demod(i+1) = 1;

end

t1 = t1 + (Tb + 0.01); t2 = t2 + (Tb + 0.1);

end

subplot(3,2,5); stem(demod);

title('QPSK demodulated bits'); xlabel('n-->');

ylabel('b(n)'); grid on;

Result :-