

LABORATORY RECORD COMMUNICATION ENGINEERING LAB (EC 2094) AUTUMN 2020

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SCHOOL OF ELECTRONICS ENGINEERING, KIIT Deemed to be University

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Experiment Number	01
Date of Experiment	03/08/2020
Date of Submission	17/08/2020
Name of the student	Debagnik Kar
Roll Number	1804373
Section	ETC-06

Aim of The Experiment :-

Generation and detection of Amplitude Modulation and Demodulation, DSBSC and SSB-SC modulation.

Equipment / Software Required:-

• MATLAB R2018a

Theory:

Amplitude modulation: AM is a modulating technique which is generally used to transmit Radio signals over a carrier signal, where the amplitude of the modulated changes in response of the signal.

Mathematically,

If the carrier signal is,

$$c(t) = Ac.\cos(2\pi fc.t)$$

And the message signal is,

$$m(t) = Am.\cos(2\pi.fm.t)$$

Then the equation of the modulated signal is,

$$s(t) = [Ac + \cos(2.\pi.fm.t)].\cos(2.\pi.fc.t)$$

If we solve this equation further, we get,

$$s(t) = Ac \left[1 + \frac{Am}{Ac} \cdot \cos(2 \cdot \pi \cdot fm \cdot t) \right] \cdot \cos(2 \cdot \pi \cdot fc \cdot t)$$

$$s(t) = Ac[1 + \mu . \cos(2.\pi.fm.t)].\cos(2.\pi.fc.t), \quad Where, \; \mu = \frac{Am}{Ac}$$

Modulating Index(μ), also known as modulation depth, of a modulation scheme describes by how much the modulated variable of the carrier signal varies around its unmodulated level. It is defined differently in each modulation scheme.

DSBSC: The transmission of a signal, which contains a carrier along with two sidebands can be termed as Double Sideband Full Carrier system or simply DSBSC.

Mathematically,

If the carrier Signal is,

$$c(t) = Ac.\cos(2.\pi.fc.t)$$

And the message signal is,

$$m(t) = Am.\cos(2.\pi.fm.t)$$

Then the modulating signal will be,

$$s(t) = m(t).c(t)$$

or,
$$s(t) = Am.Ac.\cos(2.\pi.fm.t).\cos(2.\pi.fc.t)$$

SSBSC: The process of suppressing one of the sidebands along with the carrier and transmitting a single sideband is called as Single Sideband Suppressed Carrier system or simply SSBSC.

Mathematically,

If the carrier Signal is,

$$c(t) = Ac.\cos(2.\pi.fc.t)$$

And the message signal is,

$$m(t) = Am.\cos(2.\pi.fm.t)$$

Then the modulating signal will be,

$$s(t) = \frac{Am.Ac}{2}.\cos[2.\pi.(fc + fm).t]$$
 for the upper Sideband

$$s(t) = \frac{Am.Ac}{2}.\cos[2.\pi.(fc - fm).t]$$
 For the lower Sideband

<<<File: GenerateAM.m Comment: This code will amplitude modulate and then demodulate a sine wave.>>>

```
%generating user defined AM Signals
%Written By Debagnik Kar 1804373
clc;
clear all;
close all;
t = linspace(0, 1, 1000) %Time of 1 secs divided by 1000 times
%Carrier wave
fc = input('Enter fc = ')
ac = input('Enter ac = ')
xc = cos(2*pi*fc*t)
%Message signal
fm = input('Enter fm = ');
am = input('enter am = ');
xm = cos(2*pi*fm*t);
%AMplitude modulation
y = [ac + am*xm].*xc;
%plot AM
subplot(4,1,1)
plot(t,xc);
xlabel("Time -->")
ylabel("Amplitude -->")
title("Carrier Wave")
subplot(4,1,2)
plot(t,xm)
xlabel("Time -->")
ylabel("Amplitude -->")
title("Message Wave")
subplot(4,1,3)
plot(t, y)
xlabel("Time -->")
ylabel("Amplitude -->")
title("Modulated Wave")
%if else statement
mu = am/ac;
if mu == 1
```

```
disp('Critical modulation');
elseif mu>1
   disp('Over modulated signal');
elseif mu<1</pre>
   disp('under modulated signal');
end
%Demodutating The wave
dm = y.^2;
[b,a] = butter(10,0.1);
xd = filter(b,a,dm);
subplot(4,1,4)
plot(t,xd);
xlabel("Time -->")
ylabel("amplitude")
title ("Demodulated Wave")
<<<File:DSBSC.m Comment: This code will generate a DSBCS Modulated signal>>>
%DSBCS Generation
%Written by Debagnik Kar 1804373
clc;
clear all;
close all;
t = linspace(0, 4, 1000);
fc = input('Enter the carrier frequency: ')
ac = input('Enter the carrier amplitude: ')
fm = input('Enter the message frequency: ')
am = input('Enter the message amplitude: ')
y = am*cos(2*pi*fm*t) %message signal
z = ac*cos(2*pi*fc*t) %carrier signal
w = ((am*ac)/2).*(cos(2*pi*(fc+fm)*t)+cos(2*pi*(fc-fm)*t))
%DSBSC Modulation
subplot(3,1,1)
plot(t,z)
xlabel("time -->")
ylabel("magnitude -->")
title("Carrier Signal")
subplot(3,1,2)
plot(t, y)
xlabel("time -->")
ylabel("magnitude -->")
title("Message Signal")
subplot(3,1,3)
plot(t,w)
xlabel("time -->")
ylabel("magnitude -->")
```

```
title("DSBSC Signal")
<<<File: SSBSC.m Comment: Generates an upper sideband, a lower side band SSBSC>>>
% Generation of SSB-SC Signal
% Written by Debagnik Kar
clear all
close all
clc
fc = input('Enter the frequency of Carrier: ')
ac = input('Enter the amplitude of Carrier: ')
fm = input('Enter the frequency of Message: ')
am = input('Enter the amplitude of Message: ')
t = linspace(0, 1, 1000)
m = am*cos(2*pi*fm*t) %Message signal
c = ac*cos(2*pi*fc*t) %carrier Signal
Susb = ((am*ac)/2).*cos(2*pi*(fc+fm)*t) %upper Sideband
Slsb = ((am*ac)/2).*cos(2*pi*(fc-fm)*t) %lower Sideband
subplot(4,1,1)
plot(t,c,'r')
xlabel("Time -->")
ylabel("Amplitude-->")
title("Carrier Wave")
subplot(4,1,2)
plot(t,m,'g')
xlabel("Time -->")
ylabel("Amplitude-->")
title("Message Wave")
subplot(4,1,3)
plot(t,Susb,'b')
xlabel("Time -->")
ylabel("Amplitude-->")
title("Upper Sideband SSB-SC Signal")
subplot(4,1,4)
plot(t,Slsb,'k')
xlabel("Time -->")
ylabel("Amplitude-->")
title ("Lower Sideband SSB-SC Signal")
```

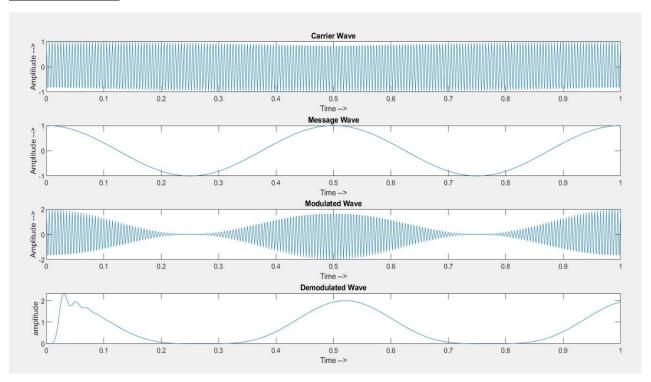


Fig 1.1: Amplitude modulating and Demodulating a Sine wave of 2Hz

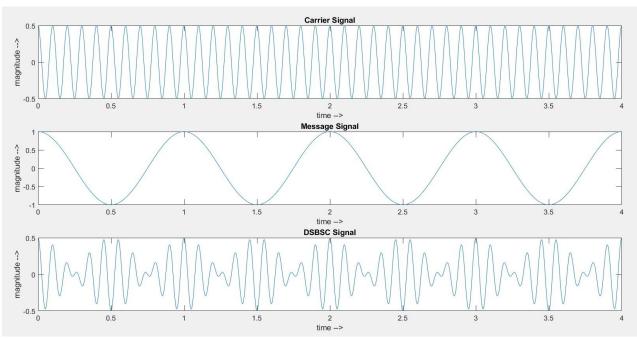


Fig 1.2: Generation of DSBSC Signal

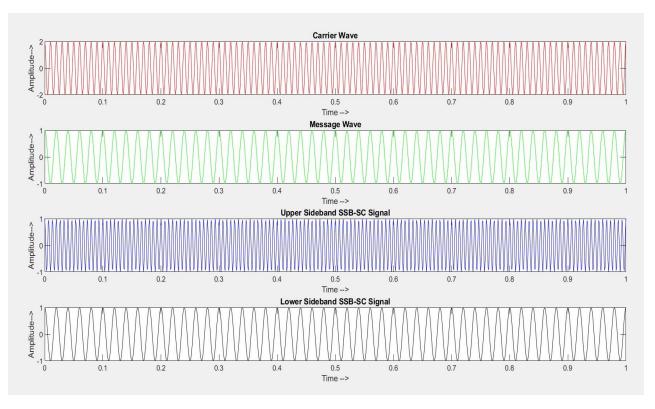


Fig 1.3: Generation of SSBSC Signal

Discussion or Inference of the experiment

This experiment taught me about different kind of amplitude modulation and demodulation techniques that is practiced in analog radio communication technology. It also helped me visualize the difference between the different techniques of transmission

Conclusion:-

The simulation of experiment is done successfully using MATLAB Software.

Experiment Number	02
Date of Experiment	10/08/2020
Date of Submission	17/08/2020
Name of the student	Debagnik Kar
Roll Number	1804373
Section	ETC-06

Aim of The Experiment :-

Study of Frequency modulation and Demodulation Techniques.

Equipment / Software Required:-

MATLAB R2018a

Theory

Frequency Modulation (FM) is a form of modulation in which changes in the carrier wave frequency correspond directly to changes in the baseband signal.

Mathematically,

If the Carrier signal is

$$c(t) = \cos(2.\pi.fc.t)$$

And the message signal is

$$m(t) = \sin(2.\pi. fm. t)$$

Then the modulating signal will be,

$$s(t) = \cos[2.\pi. fc. t - \{\mu. \sin(2.\pi. fm. t)\}]$$

$$Where \mu = \frac{Am}{Ac}$$

<<<File:FMModDemod.m Comment: This code generates a FM signal and Demodulates it>>

```
%Generating a FM Signal and Demodulating it
%Written by Debagnik Kar 1804373
clc;
clear all;
close all;
fc=input('Enter the carrier signal: ');
fm=input('Enter the message signal: ');
mu=input('Modulation index ');
t=linspace(0,1,1000);
c=cos(2*pi*fc*t);%carrier signal
m=sin(2*pi*fm*t);%message signal
subplot(4,1,1);
plot(t,c,'r'); %plotting the carrier signal
ylabel('amplitude');
xlabel('time');
title('Carrier signal');
subplot(4,1,2);
plot(t,m,'g'); %plotting the message signal
ylabel('amplitude');
xlabel('time');
title('Message signal');
y=cos(2*pi*fc*t-(mu*cos(2*pi*fm*t))); % FM Generation
subplot(4,1,3);
plot(t,y,'b'); % Plotting the FM Generation
ylabel('amplitude');
xlabel('time');
title('Frequency Modulated signal');
%FM Demodulation
dem = diff(y);
dem = [0, dem];
rect dem = abs(dem)
b,a] = butter(10,0.06);
rec = filter(b,a,rect dem);
subplot(4,1,4)
plot(t,rec,'k')
xlabel('Time')
ylabel('Amplitude')
title('Demodulated Signal')
```

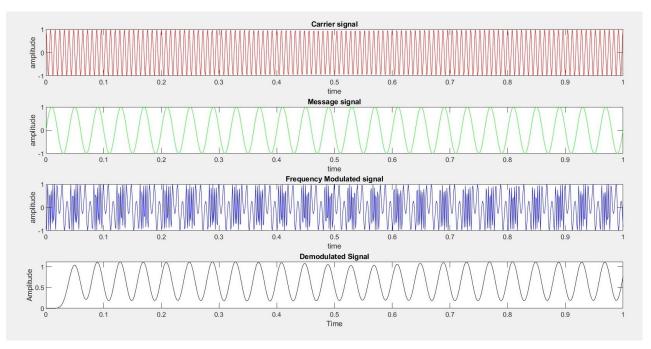


Fig 2.1: Generation of FM Signal and demodulating it.

Discussion or Inference of the experiment

This experiment taught me techniques of radio transmission that are currently being practiced. It also taught me the reasons that FM is better than AM in radio signal transmissions. I also learnt to demodulate a FM signals.

Conclusion:-

Simulation of experiment is done successfully

Experiment Number	03
Date of Experiment	24/08/2020
Date of Submission	29/08/2020
Name of the student	Debagnik Kar
Roll Number	1804373
Section	ETC - 06

Aim of The Experiment:-

Generation and detection of PAM, PWM and PPM techniques

Equipment / Software Required:-

MATLAB R2018a

Theory

<u>Pulse Amplitude Modulation (PAM):</u> is an analog modulating scheme in which the amplitude of the pulse carrier waves varies proportionally to the instantaneous amplitude of the message signal. There are two types of PAM 1. Natural PAM 2. Flat top PAM.

- 1. <u>Natural PAM:</u> a signalled at Nyquist rate is reconstructed, by passing it though an efficient Low pass Frequency with exact cut-off frequency.
- 2. <u>Flat top PAM:</u> Although the natural PAM 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 cannot be changed with respect to the analog signal, to be sampled. The tops of amplitude remain flat. This process simplifies the circuit design.

<u>Pulse Width Modulation (PWM):</u> is an analog modulating scheme in which the duration or width or time of the pulse carrier varies proportional to the instantaneous amplitude of the message signal.

The width of the pulse varies in this method, but the amplitude of the signal remains constant. Amplitude limiters are used to make the amplitude of the signal constant. These circuits clip off the amplitude, to a desired level and hence the noise is limited. It is mainly used in the semiconductor industries to transmit signals between microcontrollers or actuators.

<u>Pulse Position Modulation:</u> s an analog modulating scheme in which the amplitude and width of the pulses are kept constant, while the position of each pulse, with reference to the position of a reference pulse varies according to the instantaneous sampled value of the message signal.

The transmitter has to send synchronizing pulses (or simply sync pulses) to keep the transmitter and receiver in synchronism. These sync pulses help maintain the position of the pulses. The following figures explain the Pulse Position Modulation.

PAM	PWM	PPM
Amplitude is varied	Width is varied	Position is varied
Bandwidth depends on the	Bandwidth depends on the	Bandwidth depends on the
width of the pulse	rise time of the pulse	rise time of the pulse
Instantaneous transmitter	Instantaneous transmitter	Instantaneous transmitter
power varies with	power varies with the	power remains constant with
	amplitude and the width of	the width of the pulses
	the pulse	
System complexity is high	System complexity is low	System complexity is low
Noise interference is high	Noise interference is low	Noise interference is low
It is similar to amplitude	It is similar to amplitude	It is similar to phase
modulation	modulation	modulation

Table 3.1: Comparison of different types of modulations

Demodulation: For demodulating all the signals. A Butterworth filter is used and the specifications of it are given below:

- PAM:
- PWM:

<<< File: PulseAmpMod.m Comment: Generated a PAM signal and then Demodulates it. Also asks the user if he/she wants to continue to generate a PPM Signal from the PWM signal.>>>

```
%PULSE AMPLITUDE MODULATION PAM
%Written by Debagnik Kar 1804373
clc
close all
clear all
t = 1:0.0001:2
f = input('Enter the value of frequency: ')
x=sawtooth(2*pi*f*t)
ts=0.02
%PULSE GENERATION
for k=1:length(t)
  if mod(t(1,k),ts) == 0
      pulse(1, k) = 1 % PULSE
  else
      pulse(1, k) = 0;
  end
end
natural pam = x.*pulse %Natural Pulse amplitude modulation
subplot(5,1,1)
plot(t,x,'r')
xlabel('Time -->')
ylabel('Amplitude -->')
title('Message')
subplot(5,1,2)
plot(t,pulse,'b')
xlabel('Time -->')
ylabel('Amplitude -->')
title('Pulse Signal')
subplot(5,1,3)
plot(t, natural pam, 'q')
xlabel('Time -->')
ylabel('Amplitude -->')
title('Natural PAM')
%FLATTOP PAM
FlatTop pam =zeros(1,length(t));
k=1;
while k <length(t)</pre>
if natural pam (1, k) \sim 0
  FlatTop pam(1,k:k+49)=natural pam(1,k)*ones(1,50); % pulse
duration is 50*0.001
  k=k+49;
```

```
else FlatTop pam(1, k) = 0;
  k=k+1;
end
end
subplot(5,1,4)
plot(t,FlatTop pam,'k')
xlabel('Time -->')
ylabel('Amplitude -->')
title('Flat top PAM')
%Demodulation
[b,a] = butter(7,0.004)
demod=filter(b,a,FlatTop_pam)
subplot(5,1,5)
plot(t,demod,'c')
xlabel('Time -->')
ylabel('Amplitude -->')
title('Demodulated PAM')
<<<File: PulseWidthMod.m Comment: Generation and demodulation of PWM>>>
%Generation of PWM
%Written by Debagnik Kar 1804373
clc
clear all
close all
fc = input('Carrier frequency in Hz: ') %user input prompt
fm = input('Message Frequency in Hz: ') %user input prompt
t = 0:0.001:1 \% TIME
carrier = sawtooth(2*pi*fc*t)
subplot 411
plot(t,carrier)
title('Carrier wave')
xlabel('Time')
ylabel('Amplitude')
% Modulating waveform
message = cos(2*pi*fm*t)
subplot 412
plot(t, message)
title('Message wave')
xlabel('Time')
ylabel('Amplitude')
% PWM waveform generation
```

```
p=find(carrier>=message)
    pwm(p) = -1
q=find(carrier<message)</pre>
     pwm(q) = 1
subplot 413
plot(t,pwm)
axis([0 1 0 2])
title('PWM waveform')
xlabel('Time')
ylabel('Amplitude')
[b, a] = butter(6, 0.009)
demod = filter(b,a,pwm)
subplot 414
plot(t,demod,'r')
xlabel('Time -->')
ylabel('Amplitude -->')
title('Demodulated Wave')
prompt = input('Do you want to continue to PPM? (1 - yes/0 -
no): ')
if prompt == 1
    PulsePosMod(t,pwm)
elseif prompt == 0
    disp('Okay!')
    exit()
else
    disp('Wrong choice, Exiting')
    goto (46)
end
<<<File: PulsePosMod.m Comment: Extension of PulseWidthMod.m , It generates a PPM
Signal from the PWM from the above signal>>>
%PPM Extension Funtion to PWM
%Written By Debagnik Kar
function PulsePosMod(t,pwm,f)
clc
figure(2)
dip=diff(pwm)
dip = [0, dip]
ppm=zeros(1,length(dip))
k=1
%Positive edge
while k<length(dip)</pre>
    if dip(1,k) == 2 \% take -2 for negative edge triggering
        ppm(1,k:k+9)=ones(1,10) % I took to block
        k=k+10
    else
```

```
k=k+1
    end
end
subplot 311
plot(t,cos(2*pi*f*t),'r')
xlabel('Time -->')
ylabel('Amplitude -->')
title('Message signal')
subplot 312
plot(t,pwm,'c')
axis([0 1 0 2])
xlabel('Time -->')
ylabel('Amplitude -->')
title('PWM Signal')
subplot 313
plot(t,ppm,'g')
axis([0 1 0 2])
xlabel('Time -->')
ylabel('Amplitude -->')
title('PPM Signal')
end
```

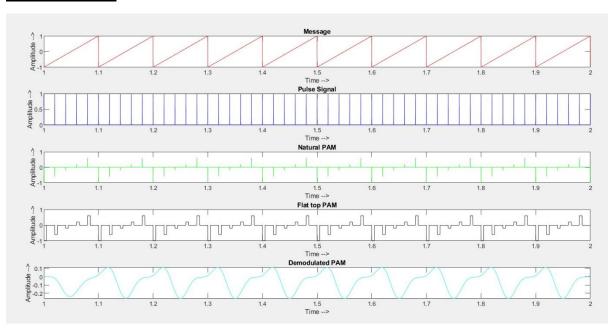


Fig 3.1: Modulation and Demodulation of Natural PAM and Flat-top PAM.

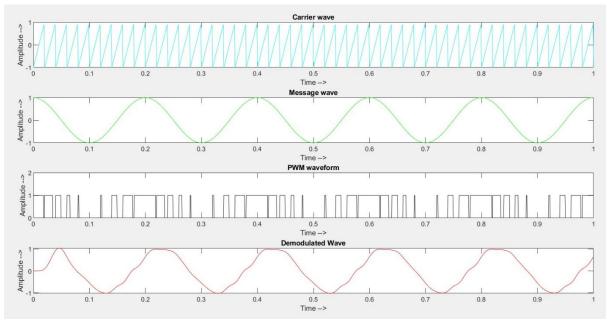


Fig 3.2: Modulation and Demodulation of PWM Signal.

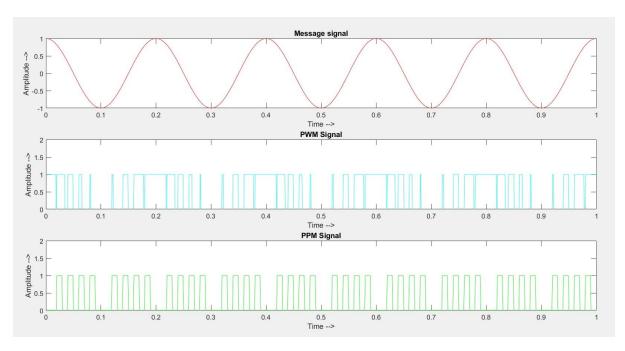


Fig 3.3: Modulation of PPM.

Discussion or Inference of the experiment

This Experiment taught me different pulse modulation techniques such as PAM, PWM, PPM. I also realized advantages of Pulse Position Modulation, over Pulse Amplitude Modulation or Pulse Width Modulation.

By understanding, Pulse position modulation has low noise interference when compared to PAM because amplitude and width of the pulses are made constant during modulation. Noise removal and separation is also very easy in pulse position modulation. I also learned about differentiator and inverter circuit, and its use during generation of PPM signal, through PWM signal.

Conclusion:-

The simulation of PAM, PWM and PAM was done successfully using MATLAB Software.

Experiment Number	04
Date of Experiment	31/08/2020
Date of Submission	15/08/2020
Name of the student	Debagnik Kar
Roll Number	1804373
Section	ETC-06

Aim of The Experiment:

Study of Pulse Code Modulation (PCM) and demodulation. Multiplexing of signal using Time Division Multiplexing (TDM) technique.

Equipment / Software Required:-

MATLAB R2018a

Theory

• <u>TDM</u>: Time division multiplexing is a technique of multiplexing, where the users are allowed the total available bandwidth on time sharing basis. Here the time domain is divided into several recurrent slots of fixed length, and each signal is allotted a time slot on a round-robin basis.

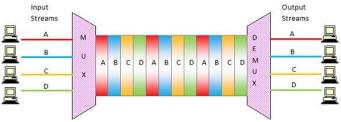


Fig 4.1: Visualizing TDM

• <u>PCM</u>: Pulse-code modulation (PCM) is a method used to digitally represent sampled analog signals. It is the standard form of digital audio in computers, compact discs, digital telephony and other digital audio applications. In a PCM stream, the amplitude of the analog signal is sampled regularly at uniform intervals, and each sample is quantized to the nearest value within a range of digital steps.

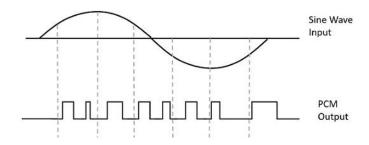


Fig 4.2: Visualizing PCM

<<<File: TDM.m Comment: Generation and demultiplexing of TDM signals>>>

```
%Generation of Time domain Multiplexing(TDM)
%Written by Debagnik Kar 1804373
clc
clear all
close all
f1=700
f2=50
s=0.005 %sampling rate
x=pulstran(0:1/f1:1,0:1/f2:1,'rectpuls',s)
y=(1/2)*pulstran(0:1/f1:1,0.01:1/f2:1,'rectpuls',s)
t=0:1/f1:1
subplot 411
plot(t,x)
title('Train of pulse [1]')
ylabel('Amplitude')
xlabel('Time')
subplot 412
plot (t, y, 'r')
title('Train of pulse [2]')
ylabel('Amplitude')
xlabel('Time')
%message signal
y1=20*sin(2*pi*2*t)
y2=20*sin(2*pi*4*t)
subplot 413
plot(t, y1, 'g')
hold on
plot(t,y2,'r')
ylabel('Amplitude')
xlabel('Time')
title('message signal')
Pam1=x.*y1
Pam2=y.*y2
```

```
y3=Pam1+Pam2
subplot(4,1,4)
plot(t, y3, 'k')
title('Tdm signals')
ylabel('Amplitude')
xlabel('Time')
figure (2)
demux=y3.*x
[b,a] = butter(7,0.02)
s1=filter(b,a,demux)
subplot 313
plot(t,s1,'g')
hold on
title('Demuxed and demodulated signal')
demux2=y3.*y
[b,a] = butter(7,0.02);
s2=filter(b,a,demux2)
plot(t,s2,'r');
hold off
ylabel('Amplitude')
xlabel('Time')
subplot 311
plot(t,y1,'g')
hold on
plot(t, y2, 'r')
ylabel('Amplitude')
xlabel('Time')
title('message signal')
subplot 312
plot(t, y3)
ylabel('Amplitude')
xlabel('Time')
title('TDM signal')
<<<File: PCM.m Comment: Generation and demodulation of PCM>>>
%Generation of PCM Signals and Demodulating it
%written by Denagnik 1804373
clc
close all
clear all
n=input('Enter n values for n-bit Pcm system: ')
n1=input('Enter number of samples in a period : ')
%Sampling Operation
```

```
x=0:2*pi/n1:4*pi
s=8*sin(x)
subplot 411
plot(s,'r')
axis([0 50 -10 10])
grid on
title('Analog Signal');
ylabel('Amplitude')
xlabel('Time')
subplot 412
stem(s,'y')
axis([0 50 -10 10])
grid on
title('Sampled Signal')
ylabel('Amplitude')
xlabel('Time')
%Quantization Process
vmax=8
vmin=-vmax
del=(vmax-vmin)/(2^n)
part=vmin:del:vmax
code=vmin-(del/2):del:vmax+(del/2)
[ind, q] = quantiz(s, part, code)
11=length(ind)
12=length(q)
for i=1:11
    if(ind(i) \sim = 0)
        ind(i) = ind(i) -1;
    end
    i=i+1
end
for i=1:12
    if(q(i) == vmin - (del/2))
        q(i) = vmin + (del/2);
    end
end
subplot 413
stem(q, 'b')
grid on
title('Quantized Signal')
ylabel('Amplitude')
xlabel('Time')
axis([0 50 -10 10])
%Encoding process
```

```
code=de2bi(ind)
k=1
for i=1:11
  for j=1:n
    coded(k) = code(i,j);
    j=j+1;
    k=k+1;
  end
  i=i+1;
end
subplot 414
grid on
stairs(coded, 'q')
axis([0 180 -0.5 1.5])
grid on
title('Encoded Signal');
ylabel('Amplitude')
xlabel('Time')
%Demodulation of pcm signals
index=bi2de(code)
q=del*index+vmin+(del/2);
figure(2)
subplot 311
plot(s,'k')
axis([0 50 -10 10])
grid on
title('Analog Signal');
ylabel('Amplitude')
xlabel('Time')
subplot 312
grid on
stairs(coded, 'c')
axis([0 180 -0.5 1.5])
grid on
title('Encoded Signal');
ylabel('Amplitude')
xlabel('Time')
subplot 313
plot(q,'m')
axis([0 50 -10 10])
grid on
title('Demodulated Signal')
ylabel('Amplitude')
xlabel('Time')
```

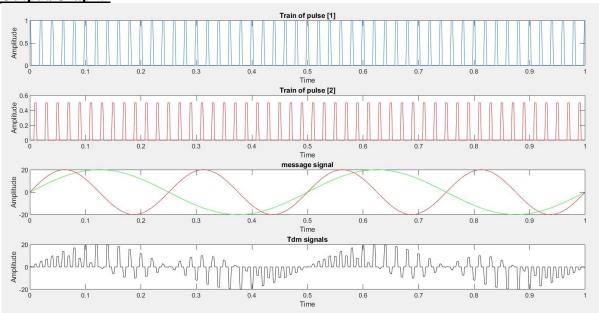


Fig 4.3: Time Domain Multiplexing

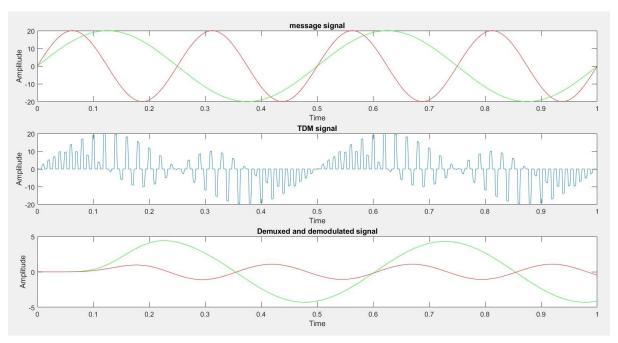


Fig 4.4: TDM Signal Demultiplexing

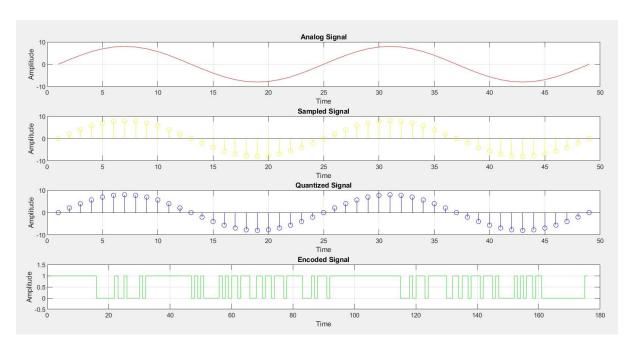


Fig 4.5: Generation of PCM Signals

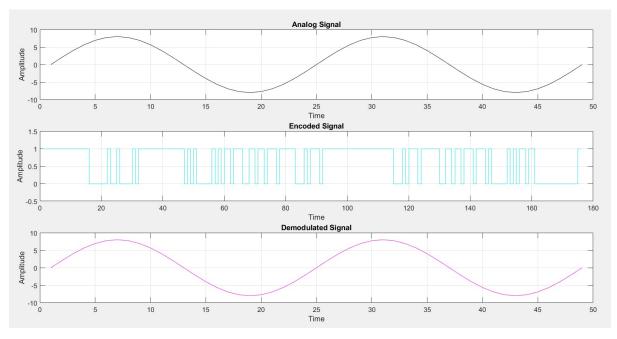


Fig 4.6: Demodulating PCM Signals

Discussion or Inference of the experiment

This experiment taught me the implementation of TDM and PCM, as we learned in our theory.

The experiment helped me understand the signal quantization process, and it's implementation in MATLAB software.

Also, I understood.

- 1. Time-division multiplexing systems are more flexible than frequency division multiplexing.
- 2. Time division multiplexing circuitry is not complex.
- 3. Problem of cross talk is not severe in TDM.

- 4. Full available channel bandwidth can be utilized for each channel in TDM.
- 5. The PCM (pulse code modulation) convenient for long-distance communication.
- 6. PCM has a higher transmitter efficiency.
- 7. PCM has a higher noise immunity

Conclusion:-

In this experiment, Time Division Multiplexing(TDM) and Pulse Code Modulation(PCM) were visualized, simulated and Demultiplexed/Demodulated (respectively) in the MATLAB software with the help of our prerequisite knowledge.

Experiment Number	05
Date of Experiment	14/09/2020
Date of Submission	15/09/2020
Name of the student	Debagnik Kar
Roll Number	1804373
Section	ETC-06

Aim of The Experiment:

Generation and detection of Delta modulation Technique

Equipment / Software Required:-

MATLAB R2018a

Theory

The type of modulation, where the sampling rate is much higher and in which the step-size after quantization is of a smaller value, such a modulation is termed as delta modulation.

Some Features of Delta Modulations:-

- An over-sampled input is taken to make full use of the signal correlation.
- The quantization design is simple.
- The input sequence is much higher than the Nyquist rate.
- The quality is moderate.
- The design of the modulator and the demodulator is simple.
- The stair-case approximation of output waveform.
- The step-size is very small, i.e., Δ delta.
- The bit rate can be decided by the user.
- This involves simpler implementation.

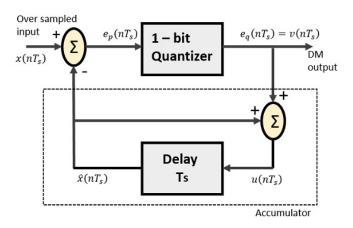


Fig 5.1: A Block diagram of Delta Modulator

```
%generation of Delta Modulated Signals
%Written by Debagnik Kar 1804373
clc
close all
clear all
am=5
            %Amplitude
fm=2
            %Frequency of the signal
fs=100
            %No. Of samples
t=0:1/fs:2 %Time
x=am*sawtooth(2*pi*fm*t) %Message signal (Sawtooth)
subplot(3,1,1)
plot(t,x,'r')
xlabel('time')
ylabel('Amplitude')
title('Message Signal')
d=(2*pi*am*fm)/fs
for n=1:length(x)
  if n==1
    e(n) = x(n);
    eq(n) = d*sign(e(n));
    xq(n) = eq(n);
  else
    e(n) = x(n) - xq(n-1);
    eq(n) = d*sign(e(n));
    xq(n) = eq(n) + xq(n-1);
  end
subplot(3,1,2)
stairs(t,xq,'b')
hold on
plot(t,x,'y')
hold off
xlabel('time')
ylabel('Amplitude')
title('Quantizad signal')
for i=1:length(x)
    if e(i) > 0
        dm(i)=1
    else
        dm(i) = 0
    end
end
subplot(3,1,3);
stairs(t,dm,'g')
axis([0 2 -1 2])
xlabel('time')
ylabel('amplitude')
title('Delta modulated signal')
```

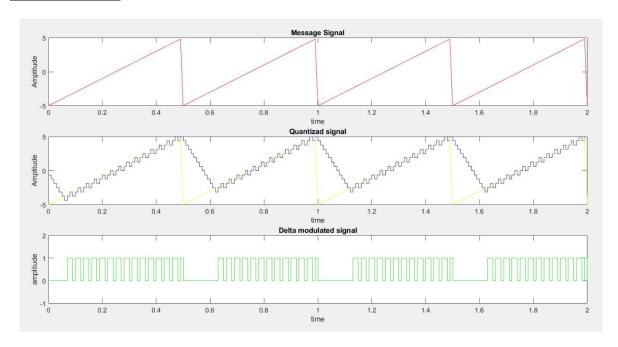


Fig 5.2: Delta modulation

Discussion or Inference of the experiment

In this experiment I came to know that Delta Modulation is simplified PCM. Here I also learnt the concept that instead of sending real value of every sample at each time, variances of present and previous samples are sent.

Conclusion:-

After successfully simulating the experiment through MATLAB I was able to generate the Delta modulated signal graph which is at par with the desired theory result.