

DIGITAL COMMUNICATIONS LAB

List of Experiments:

1. PCM Generation and Detection.
2. Differential Pulse Code modulation.
3. Delta modulation.
4. Time Division Multiplexing of 2band Limited Signals.
5. Frequency Shift Keying: Generation and Detection.
6. Phase Shift Keying: Generation and Detection.
7. Amplitude shift Keying: Generation and Detection.
8. Study of the spectral characteristics of PAM, QAM.
9. DPSK: Generation and Detection.
10. QPSK: Generation and Detection.

Note: (Any 6 Experiments)

Expt No: 1

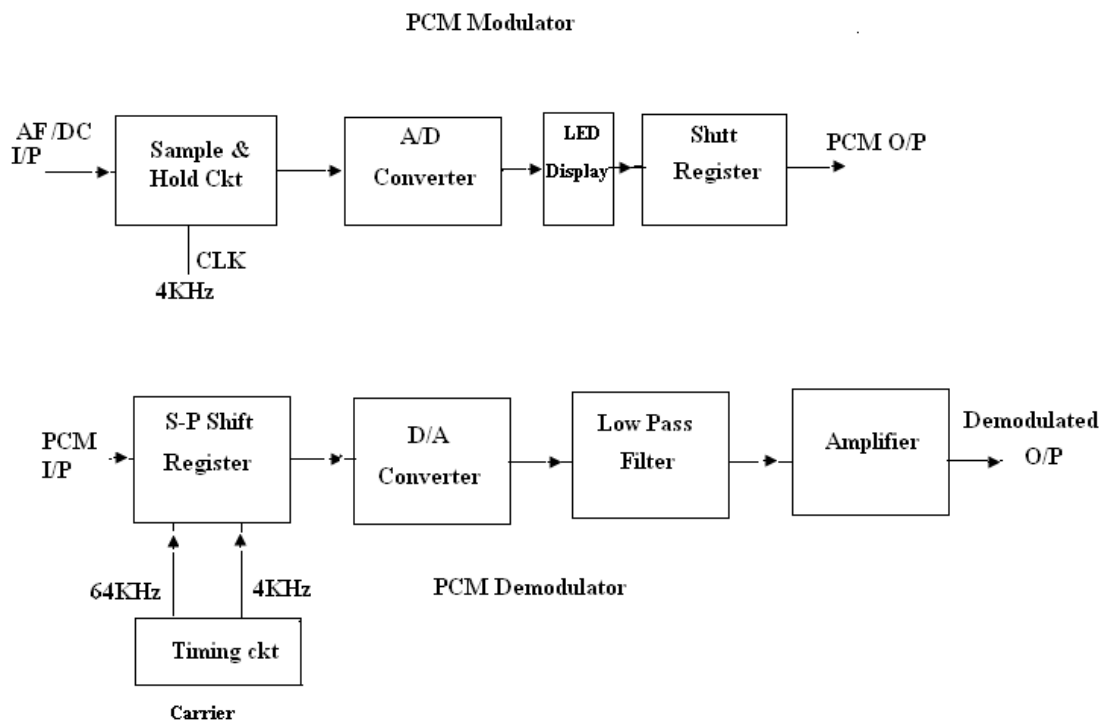
PULSE CODE MODULATION AND DEMODULATION

AIM: To Study & understand the operation of the Pulse code modulation & Demodulation.

APPARATUS:

1. PCM Modulator trainer
2. PCM Demodulator trainer
3. Storage Oscilloscope/ Dual Trace Oscilloscope
(Note: Storage oscilloscope is desired for satisfactory observation of PCM wave forms)
4. Digital multimeter.
5. 2 No.'s of co-axial cables (standard accessories with trainer)
6. patch chords
- 7.

CIRCUIT DIAGRAM:



THEORY:

Pulse modulation: A form of modulation in which a pulse train is used as the carrier. Information is conveyed by modulating some parameter of the pulses with a set of discrete instantaneous samples of the messages signal. The minimum sampling frequency is the minimum frequency at which the modulating waveform can be sampled to provide the set of discrete values without a significant loss of information.

PCM: In pulse code modulation (PCM) only certain discrete values are allowed for the modulating signals. The modulating signal sampled, as in other forms of pulse modulation. But any sample falling within a specified range of values is assigned a discrete value. Each value is assigned a pattern of pulses and the signal transmitted by means of this code. The electronic circuit that produces the coded pulse train from the modulating waveform is termed a coder or encoder. A suitable decoder must be used at the receiver in order to extract the original information from the transmitted pulse train.

PROCEDURE:

1. Study the theory of operation thoroughly.
2. Connect the trainer (Modulator) to the mains and switch on the power supply.
3. Observe the output of the AF generator using CRO, it should be a Sine wave of 200Hz frequency with 3Vpp amplitude.
4. Verify the output of the DC source with multimeter / scope, output should vary 0 to +5v.
5. Observe the output of the Clock generator using CRO, they should be 64KHz and 4KHz frequency of square wave with 5 Vp amplitude.

Note: These clock signals are internally connected the circuit so no external connections are required.

6. Connect the trainer (De Modulator) to the mains and switch on the power supply.
7. Observe the output of the clock generator using CRO, it should be 64 KHz square wave with 5 Vp-p amplitude.

PCM Operation (with DC input)

Modulation:

8. Set DC source to some value say 1 V with the help of multimeter and connect it to the A/D converter input and observe the output LED's.
9. Note down the digital code i.e output of the A/D converter and compare with the theoretical value

Theoretical value can be obtained by:

$$\frac{\text{A/D input voltage}}{1 \text{ LSB value}} = X_{(10)} = Y_{(2)}$$

Where

$$1 \text{ LSB value} = V_{\text{ref}} / 2^n$$

Since $V_{\text{ref}} = 5\text{V}$ and $n = 8$

$$1 \text{ LSB Value} = 0.01953$$

Example:

$$\begin{aligned} \text{A/D input voltage} &= 1 \text{ V} \\ &= 51.2_{(10)} \\ &= 00110011_{(2)} \end{aligned}$$

So digital output is 00110011

10. Keep CRO in dual mode. Connect one channel to 4KHz signal (which is connected to the shift register) and another channel to the PCM out put
11. Observe the PCM output with respect to the 4KHz signal and sketch the waveforms. Compare them with the given waveforms

Note: From this wave form you can observe that the LSB bit enters the output first.

Demodulation:

12. Connect PCM signal to the demodulators (S-P Shift register) from the PCM modulator (AET-68M) with help of coaxial cable (supplied with the trainer)
13. Connect clock signal (64 KHz) from the transmitter to the receiver using coaxial cable.
14. Connect transmitter clock to the timing circuit
15. Observe and note down the S-P shift register output data and compare it with the transmitted data (i.e output A/D converter at transmitter). You will notice that the output of the S-P shift register is following the A/D converter output

in the modulator. Observe D/A converter output (demodulated output) using multimeter /scope and compare it with the original signal and you can observe that there is no loss in information in process of conversion and transmission.

16. Similarly you can try for different values of modulating signal voltage.

Sample work sheet:

1. Modulating signal : 1V
2. A/D output (theoretical) : 00 11 00 11₍₂₎
3. A/D output (practical) : 00 11 00 11₍₂₎
4. S-P output : 00 11 00 11₍₂₎
5. D/A Converter output : 1V
(Demodulation output)

PCM Operation (with AC input)

Modulation:

17. Connect AC signal of 2V_{PP} amplitude to Sample & Hold circuit.
18. Keep the CRO in dual mode. Connect one channel to the AF signal and another channel to the sample & hold output. Observe and sketch the sample & hold output.
19. Connect the sample and hold output to the A/D converter and observe the PCM output using storage oscilloscope/ DTO
20. Observe PCM output by varying AF signal voltage.

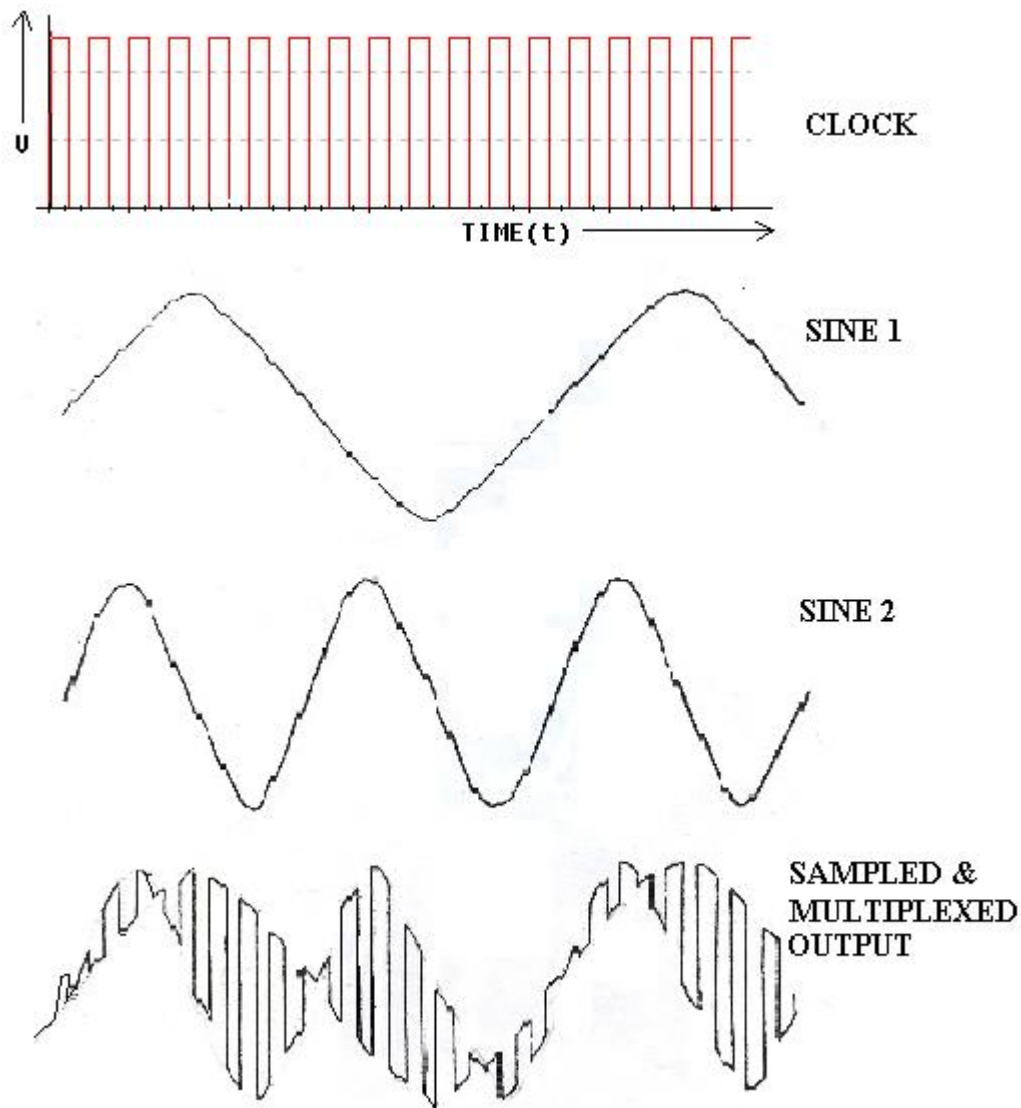
Demodulation

21. Connect PCM signal to the demodulator input (S-P shift register) from the PCM modulator with the help of coaxial cable (supplied with the trainer)
22. Connect clock signal (64KHz) from the transmitter to the receiver using coaxial cable
23. Connect transmitter clock to the timing circuit.
24. Keep CRO in dual mode. Connect CH 1 input to the sample and hold output and CH 2 input of the D/A converter output .
25. Observe and sketch the D/A output.
26. Connect D/A output to the LPF input.
27. Observe output of the LPF/Amplifier and compare it with the original modulating signal .
28. From above observation you can verify that there is no loss in information (modulating signal) in conversion and transmission process.

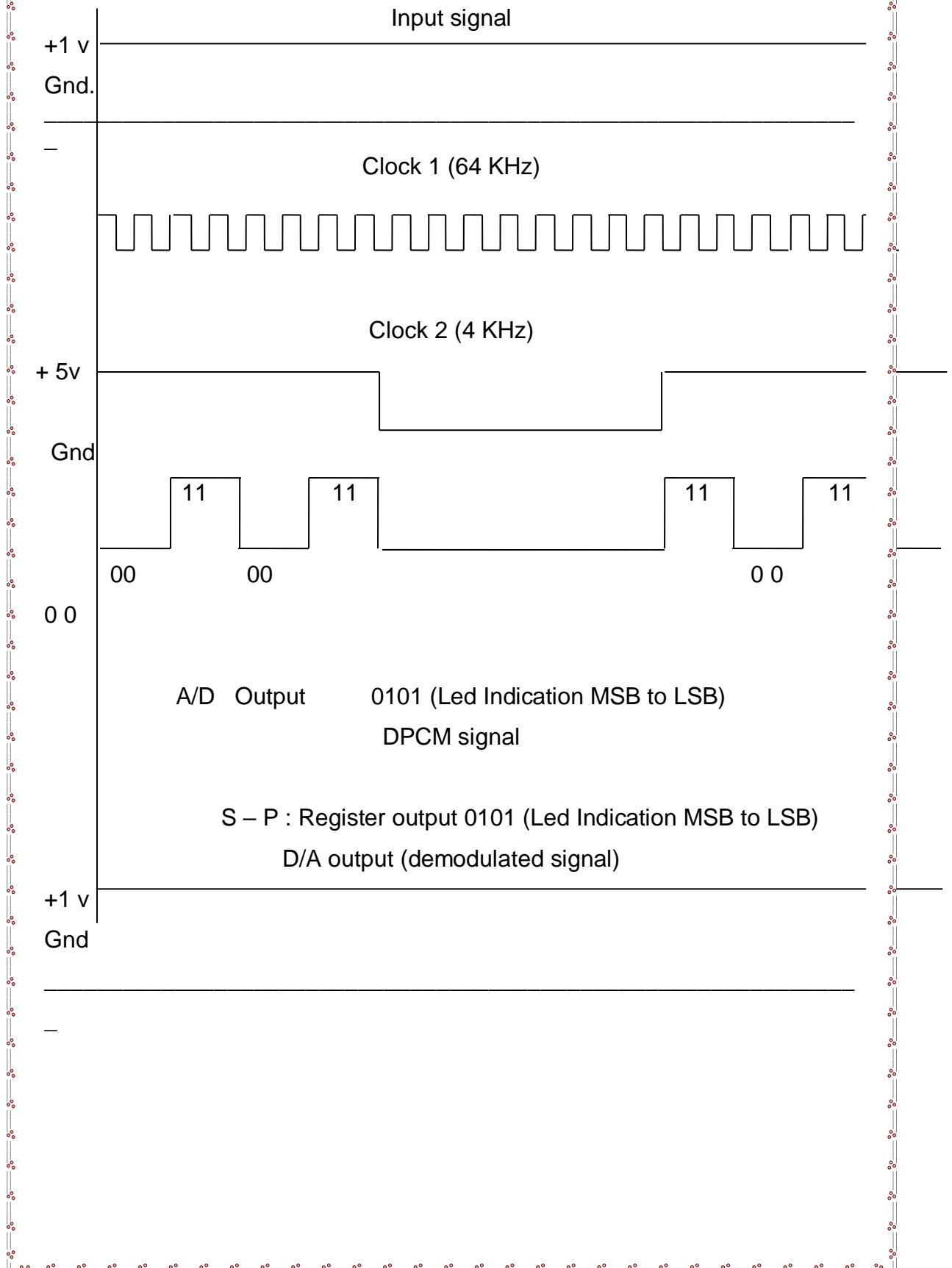
29. Disconnect clock from transmitter and connect to local oscillator (i.e. clock generator output from Demodulator) with remaining setup as it is. Observe D/A output and compare it with the previous result. This signal is little bit distorted in shape. This is because lack of synchronization between clock at transmitter and clock at receiver.

Note: You can take modulating signals from external sources. Maximum amplitude should not exceed 4V incase of DC and 3V_{PP} in case of AC (AF) signals.

EXPECTED WAVEFORMS:



PCM wave forms / timing diagram of DC input



RESULT:

QUESTIONS:

1. Differentiate PCM over Analog modulation?
2. What is bit synchronization & frame synchronization?
3. Explain block diagram of PCM?
4. What is the different error control coding technique?
5. What is resolution in ADC?
6. For arbitrary fixed reference voltage write the table of 4-bit ADC?
7. The accuracy of any digital reproduction of an analog signal depends on what?
8. If sample requires at least 12 levels of precision (+0 to +5 and -0 to -5). How many bits should be sent for each sample? use one bit form sign.
9. What is the formula for bit rate in PCM?
10. If we want to digitize human voice (4 KHz B.W), what is the bit rate assuming 8 Bits/sample?
11. What is the sampling rate for PCM if the frequency ranges from 1000Hz to 4000Hz?
12. If the interval between two samples in a digital signal is 125 micro seconds. What is the sampling rate?

DIFFERENTIAL PULSE CODE MODULATION AND DEMODULATION

AIM: To Study & understand the operation of the DPCM

APPARATUS:

1. DPCM Modulator trainer
2. DPCM Demodulator trainer
3. Storage Oscilloscope
4. Digital Multimeter.
5. 2 No's of co- axial cables (standard accessories with trainer)
6. Patch chords

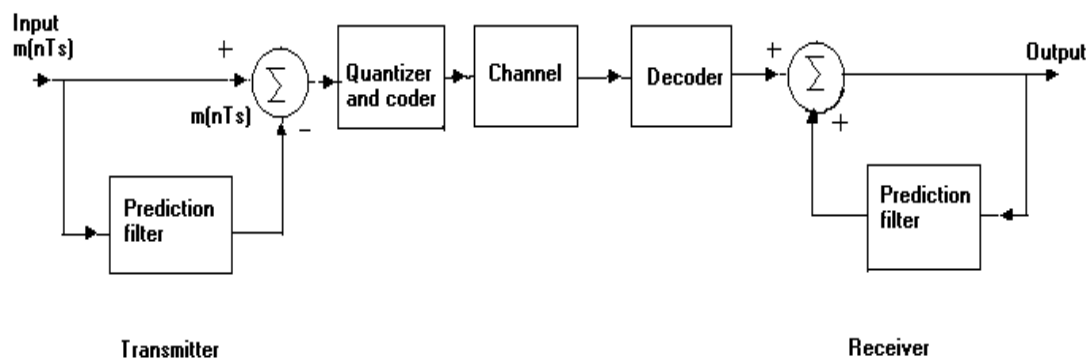
THEORY:

Differential PCM is quite similar to ordinary PCM. However, each word in this system indicates the difference in amplitude, positive or negative, between this sample and the previous sample. Thus the relative value of each sample is indicated rather than, the absolute value as in normal PCM.

This unique system consists of

BLOCK DIAGRAM:

Block diagram of DPCM



I. DPCM Modulator

1. Regulated power supply
2. Audio Frequency signal generator
3. Prediction Filter
4. Sample & Hold circuit
5. A/D Converter
6. Parallel –Serial Shift register
7. Clock generator / Timing circuit
8. DC source

II. DPCM Demodulator

1. Regulated Power Supply
2. Serial-Parallel Shift registers.
3. D/A converter.
4. Clock generator
5. Timing circuit
6. Prediction filter
7. Passive low pass filter

PROCEDURE:

1. Study the theory of operation thoroughly.
2. Connect the trainer (Modulator) to the mains and switch on the power supply.
3. Observe the output of the AF generator using CRO, it should be Sine wave of 400 Hz frequency with 3V pp amplitude.
4. Verify the output of the DC source with multimeter/scope; output should vary 0 to +290mV.
5. Observe the output of the Clock generator using CRO, they should be 64 KHz and 8 KHz frequency of square with 5 Vp-p amplitude.
6. Connect the trainer (De Modulator) to the mains and switch on the power supply.
7. Observe the output of the Clock generator using CRO; it should be 64 KHz square wave with amplitude of 5 pp.

DPCM Operation (with DC input):

Modulation:

1. Keep CRO in dual mode. Connect one channel to 8 KHz signal (one which is connected to the Shift register) and another channel to the DPCM output.
2. Observe the DPCM output with respect to the 8 KHz signal and sketch the Waveforms.

Note: From this waveform you can observe that the LSB bit enters the output First.

Demodulation

3. Connect DPCM signal to the demodulator (S-P register) from the DPCM modulator with the help of coaxial cable (supplied with the trainer).
4. Connect clock signal (64 KHz) from the transmitter to the receiver using coaxial cable.
5. Connect transmitter clock to the timing circuit.
6. Observe and note down the S-P shift register output data and compare it with the transmitted data (i.e. output A/D converter at transmitter) notice that the output of the S-P shift register is following the A/D converter output in the modulator.
7. Observe D/A converter output (demodulated output) using multimeter/scope and compare it with the original signal and can observe that there is no loss in information in process of conversion and transmission.

DPCM Operation (with AC input):

Modulation:

8. Connect AC signal of $3V_{PP}$ amplitude to positive terminal of the summer circuit.

Note: The output of the prediction filter is connected to the negative terminal of the summer circuit and can observe the waveforms at the test points provided on the board.

9. The output of the summer is internally connected to the sample and hold circuit
10. Keep CRO in dual mode. Connect one channel to the AF signal and another channel to the Sample and Hold output. Observe and sketch the sample & hold output

11. Connect the Sample and Hold output to the A/D converter and observe the DPCM output using oscilloscope.
12. Observe DPCM output by varying AF signal voltage.

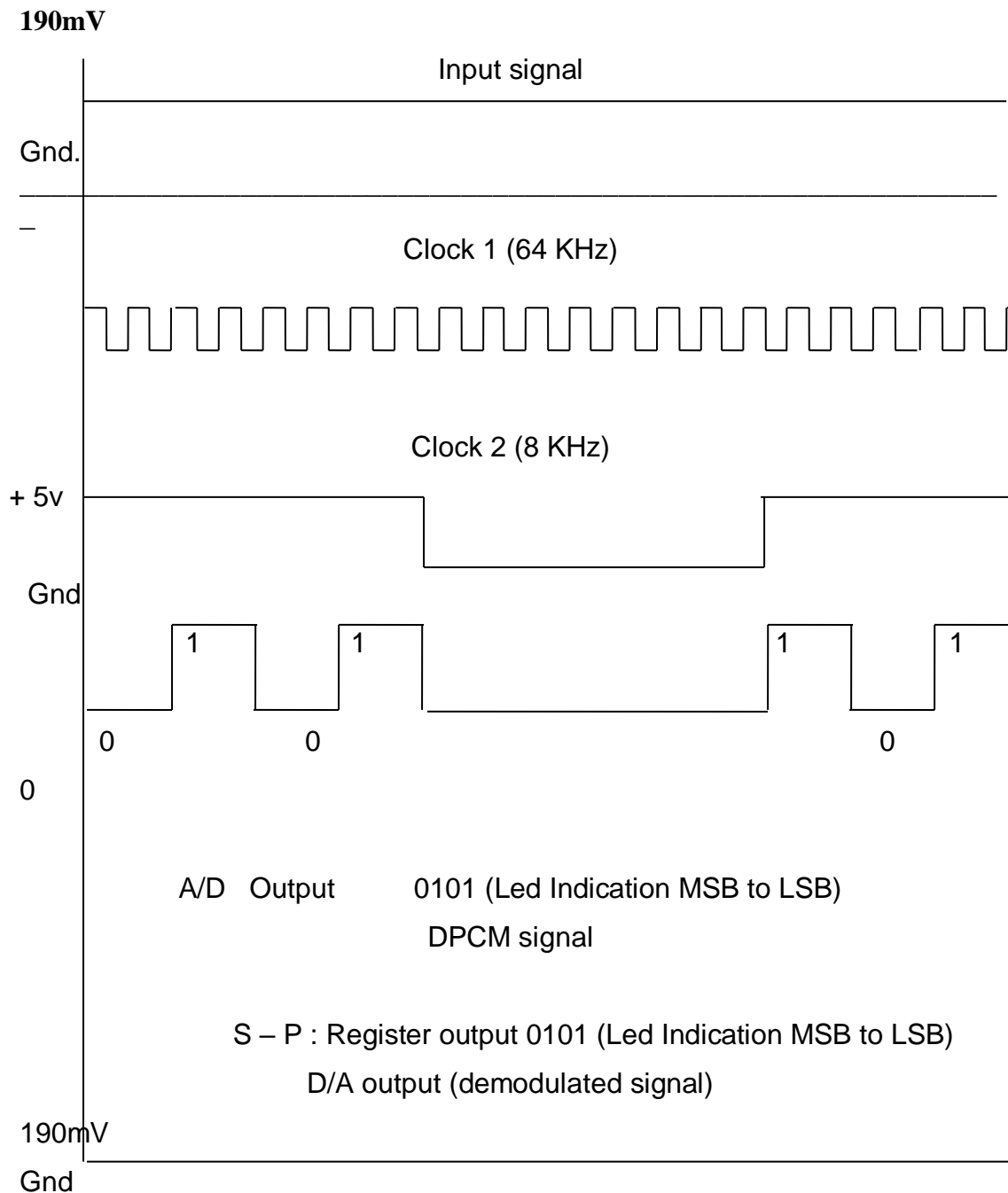
Demodulation:

13. Connect DPCM signal to the demodulator input (S-P shift register) from the DPCM modulator with the help of coaxial cable (supplied with trainer).
14. Connect clock signal (64 KHz) from the transmitter to the receiver using coaxial cable.
15. Connect transmitter clock to the timing circuit.
16. Keep CRO in dual mode. Connect one channel to the sample & hold output and another channel to the D/A converter output.
17. Observe and sketch the D/A output
18. Connect D/A output to the LPF input and observe the output of the LPF.
19. Observe the wave form at the output of the summer circuit.
20. Disconnect clock from transmitter and connect to the local oscillator (i.e., clock generator output from Demodulator) with remaining setup as it is. Observe D/A output and compare it with the previous result. This signal is little bit distorted in shape. This is because lack of synchronization between clock at transmitter and clock at receiver.

EXPECTED WAVEFORMS:

Draw the wave forms for the given DC input (190mV) corresponding binary data wave form, and for AC input draw sample and hold waveform then D/A converter o/p and then reconstructed AC signal

DPCM way form / timing diagram DC input



RESULT:

Expt No: 3

DELTA MODULATION AND DEMODULATION

AIM: To transmit an analog message signal in its digital form and again reconstruct back the original analog message signal at receiver by using Delta modulator.

APPARATUS:

Equipment required:

1. DM Modulator trainer
2. DM Demodulator trainer
3. Storage Oscilloscope
(Note: Storage oscilloscope is desired for satisfactory observation of DM wave forms)
4. Digital multimeter.
5. 2 No's co-axial cables (standard accessories with trainer)

THEORY:

Introduction:

Pulse modulation: A form modulation in which a pulse train is used as the carrier. Information is conveyed by modulating some parameter of the pulses with a set of discrete instantaneous samples of the message signal. The minimum sampling frequency is the minimum frequency at which the modulating waveform can be sampled to provide the set of discrete values without a significant loss of information. There are different forms of pulse modulations like pulse amplitude modulation (PAM), pulse width modulation (PWM), pulse position modulation (PPM). In PAM the amplitude of the pulses is modulated by the corresponding samples of the modulating wave. In PPM the samples are used to vary the time of occurrence of some parameter of the pulses. In PWM the time of occurrence of the leading edge or trailing edge is varied from its un-modulated position. All these types are un-coded modulation.

PCM: in pulse code modulation (PCM) only certain discrete values are allowed for the modulating signals. The modulating signal is sampled, as in other forms of pulse modulation. But any sample falling within a specified range of values is assigned a discrete value. Each value is assigned a pattern of pulses and the signal transmitted by means of this code. The electronic circuit that produces the coded pulse train from the modulating waveform is termed a coder or encoder. A suitable decoder must be used

at the receiver in order to extract the original information from the transmitted pulse train.

DPCM: Differential PCM differs from the conventional PCM in the respect that here only the relative amplitude of various samples and not the absolute magnitude is indicated. That is each word in this system indicates difference in amplitude between this sample and the previous sample. The logic behind this approach is that there are very little variations from sample to sample and the transmission of difference would require only a fewer bits and hence a smaller bandwidth.

Encoding and decoding process in differential PCM tend to become very complicated and it is because of these reasons that this system has not found wide acceptance.

DM: Delta modulation is almost similar to differential PCM. In this, only one bit is transmitted per sample just to indicate whether the present sample is larger or smaller than the previous one. The encoding, decoding and quantizing process become extremely simple but this system cannot handle rapidly varying samples. This increases quantizing noise. It has also not found wide acceptance.

Delta Modulation & Demodulation is a self sustained and well organized kit for the demonstration of delta modulation & demodulation. Basic circuits those are well known to every electronic student have been used in this trainer system. And this is absolute practical implementation of delta modulation theory which we come across in class room. This unique system consists of:

DM Modulator:

- Regulated power supply.
- Audio Frequency signal generator.
- Buffer/Signal shaping network.
- Voltage comparator.
- 4 Bit UP/DOWN Counter.
- Clock generator/ Timing circuit.
- 4 Bit D/A converter.
- DC source.

DM Demodulator:

- Regulated power supply.
- 4 Bit UP/DOWN Counter.
- 4 Bit D/A converter.
- Clock generator.
- Passive low pass filter.
- Audio amplifier.

EXPERIMENTAL PROCEDURE:

DM Modulator:

1. Study the theory of operation.
2. Connect the trainer (DM Modulator) to the mains and switch on the power supply.
3. Observe the output of the AF generator using CRO, it should be a Sine wave of 100 Hz frequency with 3Vpp amplitude.
4. Verify the output of the DC source with multimeter/scope; output should vary 0 to +4v.
5. Observe the output of the Clock generator using CRO, they should be 4 KHz frequency of square wave with 5 Vpp amplitude.

Note: This clock signal is *internally connected to the up/down counter* so no external connection is required

DM With DC Voltage as modulating signal:

1. Connect dc signal from the DC source to the inverting input of the comparator and set some voltage say 3V.
2. Observe and plot the signals at D/A converter output (i.e., non-inverting input of the comparator), DM signal using CRO and compare them with the wave forms given in figure1:2.

Connect DM signal (from Modulator) to the DM input of the demodulator

1. Connect clock (4 KHz) from modulator to the clock input of the demodulator. Connect clock input of the UP/DOWN counter to the clock from transmitter with the help of springs provided.
2. Observe digital output (LED Indication) of the UP/DOWN counter and compare it with the output of the UP/DOWN. By this you can notice that the both the outputs are same.
3. Observe and plot the output of the D/A converter and compare it with the wave forms given in Figure 1:2.
4. Measure the demodulated signal (i.e. output of the D/A converter with the help of multimeter and compare it with the original signal.
5. From above observation you can notice that the both the voltages are equal and there is no loss in process of modulation, transmission and demodulation.
6. Similarly you can verify the DM operation for different values of modulating signal.

DM With AF signals as modulating signal:

7. Connect AF signal from AF generator to the inverting input of the comparator and set output amplitude at 3Vpp.
8. Observe and plot the signals at D/A converter output (i.e., non-inverting input of the comparator), DM signal using CRO and compare them with the wave forms given in figure1:3.
9. Connect DM signal to the DM input of the demodulator.
10. Connect clock (4 KHz) from modulator to the clock input of the demodulator. Connect clock input of the UP/DOWN counter to the clock from transmitter with the help of springs provided.
11. Observe and plot the output of the D/A converter and compare it with the wave forms given in Figure 1:3.
12. Observe and sketch the D/A output.
13. Connect D/A output to the LPF input.
14. Observe the output of the LPF/Amplifier and compare it with the original modulating signal.

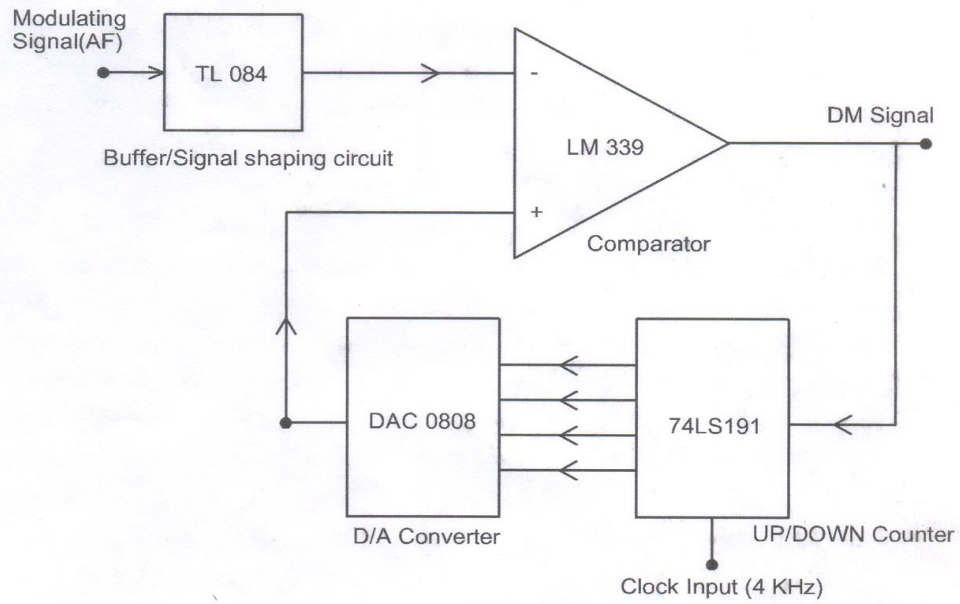
From above observation you can verify that there is no loss in information (modulating signal) in conversion and transmission process.

15. Disconnect clock from transmitter and connect to local oscillator (i.e., clock generator output from with remaining setup as it is. Observe demodulated signal output and compare it with the previous result. This signal is little bit distorted in shape. This is because lack of synchronization between clock at transmitter and clock at receiver.

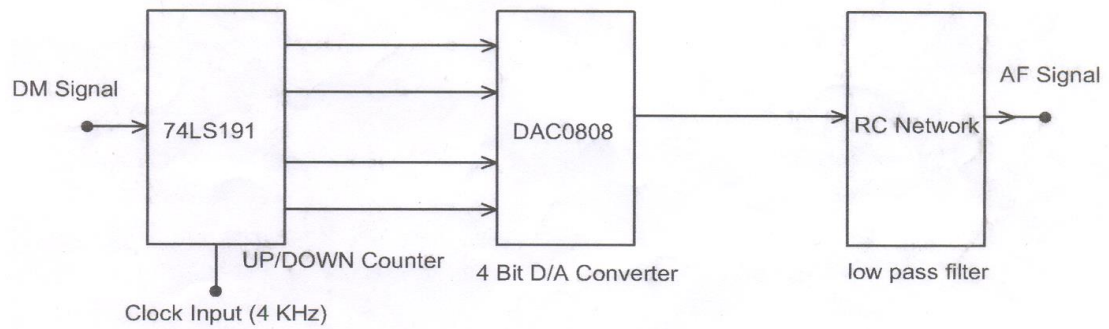
Note: you can take modulating signals from external sources. Maximum amplitude should not exceed 4 V in case of DC and 3 Vpp in case of AC (AF) signals.

CIRCUIT DIAGRAM

Figure 1:1 DM System Modulator



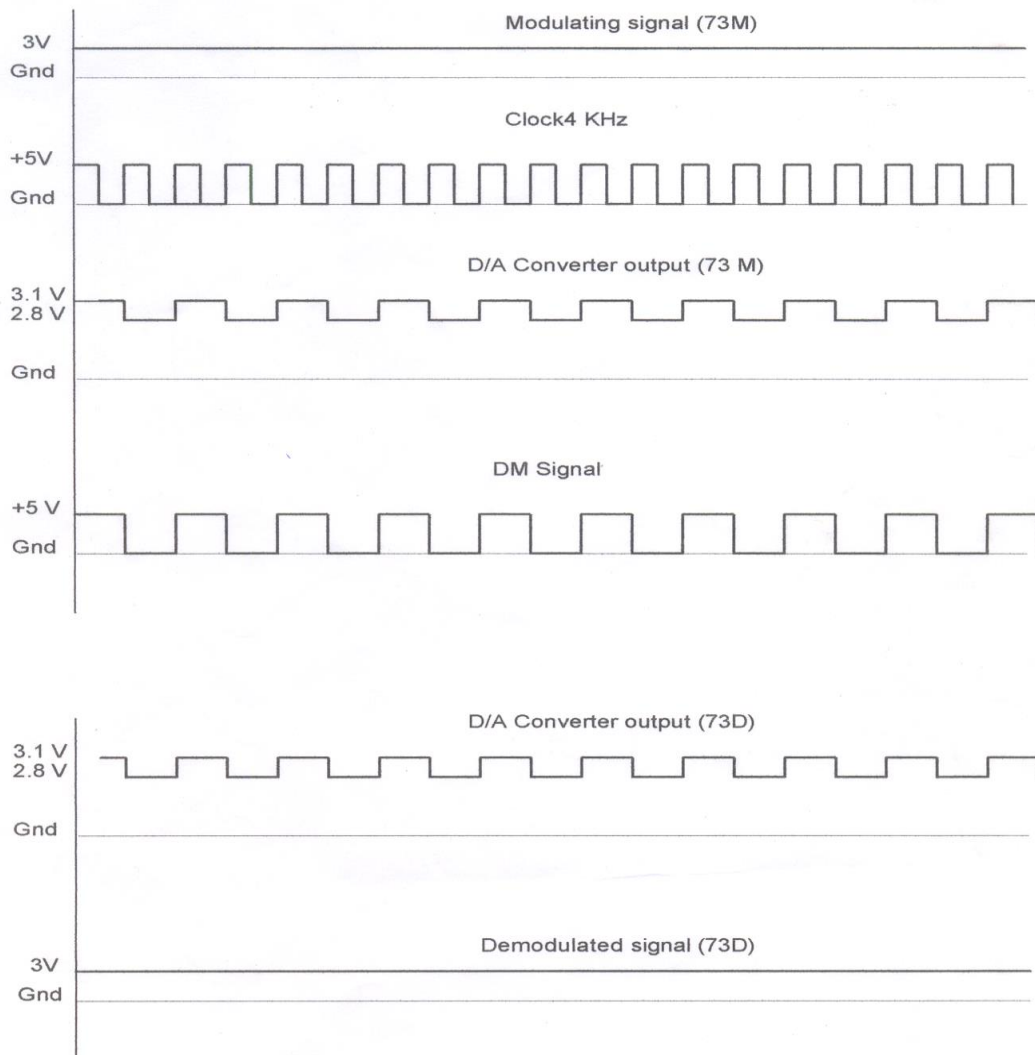
Demodulator



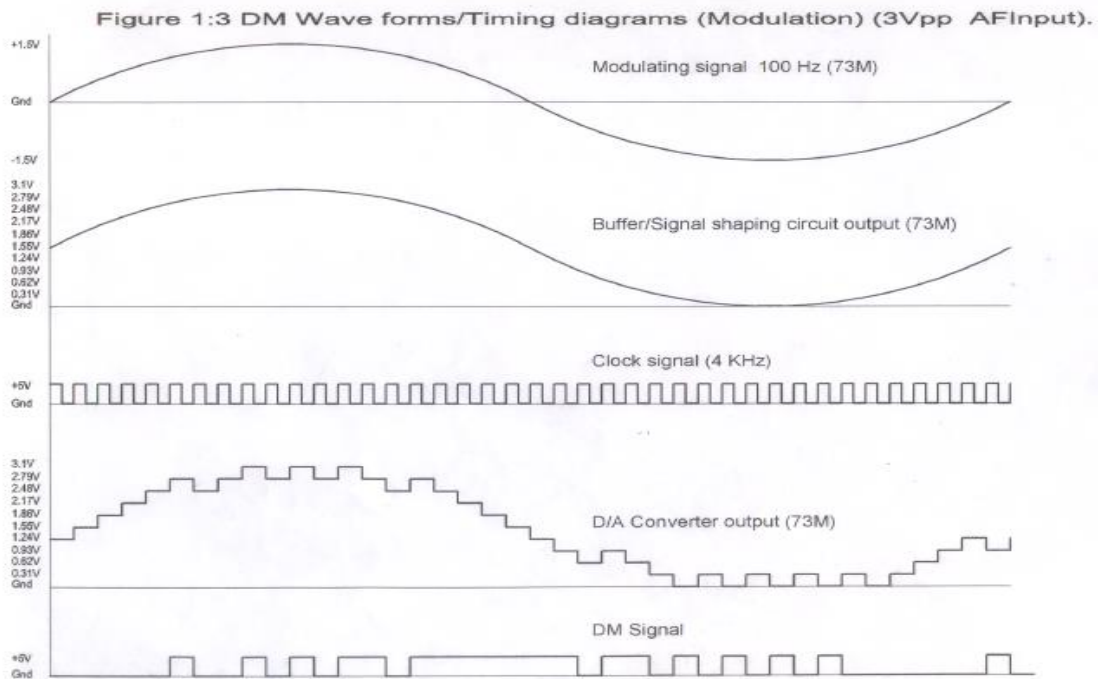
EXPECTED GRAPHS:

MODULATION WITH DC Input:

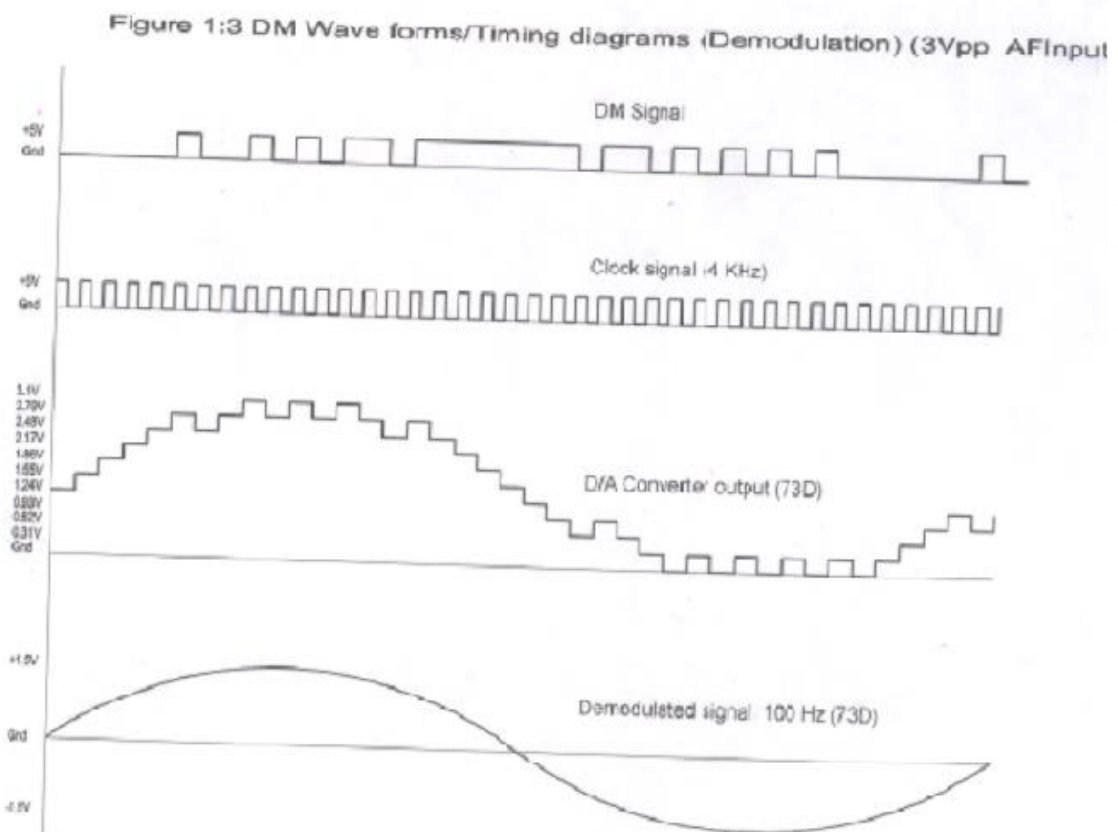
Figure 1:2 DM Wave forms/Timing diagrams (3 V DC Input).



MODULATION WITH AF Input:



DEMODULATION WITH AF Input:



PRECAUTIONS:

1. Connections must be tight.
2. Note down the comparator inputs carefully.

RESULT: Hence we obtained the digital data output of a given analog message signal by using delta modulator

VIVA QUESTIONS:

- What are the advantages of Delta modulator?
- What are the disadvantages of delta modulator?
- How to overcome slope overload distortion?
- How to overcome Granular or ideal noise?
- What are the differences between PCM & DM?
- Define about slope over load distortion?
- What is the other name of Granular noise?
- What is meant by staircase approximation?
- What are the disadvantages of Delta modulator?
- Write the equation for error at present sample?

TIME DIVISION MULTIPLEXING & DEMULTIPLEXING

AIM: To transmit a multiplexed output of different frequency message signals through a single channel using TDM system and recover back the original message signals through a demultiplexer at receiver end.

EQUIPMENT REQUIRED:

1. TDM Multiplexer trainer
2. TDM De-Multiplexer trainer
3. Storage Oscilloscope
(Note: Storage oscilloscope is desired for satisfactory observation of TDM wave forms)
4. Digital Multimeter.
5. 2 No's co-axial cables (standard accessories with trainer)

THEORY:

Time Division Multiplexing is the technique for simultaneous transmission of more than one intelligence signals on one channel and the single carrier by utilizing the time scale without loss of identity of an individual signal. PACIFIC Educational trainer AET-55 (55M & 55D) is a self sustained kit for the demonstration of 2 Channel Time Division Multiplexer & De-multiplexer system. This system provides visual understanding of actual TDM concepts, and this consists of:

1. TDM Multiplexer:

1. Regulated power supply.
2. Audio Frequency signal generators.
3. 2 channel TDM multiplexer.
4. Clock generator.
5. Logic source.

2. TDM De-Multiplexer:

1. Regulated power supply.
2. 2 channel TDM de-multiplexer.
3. Clock generator.

4. 2 no's of passive low pass filters.
5. 2 no's of audio amplifiers.
6. Logic source.

CIRCUIT DESCRIPTION:

Regulated power supply:

This consists of a bridge rectifier followed by Capacitor filters and three terminal regulators 7805 and 7905 to provide regulated DC voltages of $\pm 5V @ 300$ mA to the on board circuits. These supplies have been *internally connected to the circuits*, so no external connections are required for operation.

Audio Frequency (AF) Signal generators:

Sine wave signals of 400 Hz & 200 Hz are generated from AF generator-1 and AF generator -2 respectively to use as a message signals to be transferred. These generators are Op-Amp based Wein bridge Oscillators using IC TL084. IC TL084 is a FET input general purpose Operational Amplifier. Amplitude control is provided in the circuit to vary the output amplitude of AF signal.

Clock generator:

A TTL compatible clock signal of variable frequency is provided on board to use as a control (timing) signal to the multiplexer & de-multiplexer circuits. This circuit has been designed based on the application of CMOS integrated circuits in linear mode. Here IC 4069 is used as an active device. 4069 is a CMOS hex inverters integrated circuit. Three inverters are used to form a oscillator and other three are connected as a buffer to isolate oscillator from output and to improve current capability.

Logic source:

As name indicates this provides logic signals i.e. Logic '1' (represents +5V) and Logic '0' (represents 0V). This is a simple two way switch followed by a buffer circuit and this is applicable in single step operation of multiplexer & de-multiplexer circuits.

Low pass filters:

These are a series of simple RC networks provided on board to reconstruct the message signals from PAM signals i.e. output of the de-multiplexer. RC values are chosen such that the cutoff frequency would be at 200 Hz.

Amplifiers:

These are Op-amp (IC TL084) based non-inverting variable gain amplifiers provided on board to amplify the recovered message signals i.e. output of the Low pass filter to desired level. Amplitude control is provided in circuit to vary the gain of the amplifier between 0 and 10. AC/DC Switch facilitates to couple the input signal through capacitor or directly to the amplifier input.

2 Channel TDM multiplexer:

Figure 1:1 shows the TDM Multiplexer using 4052. CD 4052 IC is a 4 channel analog multiplexer/de-multiplexer and is used as an active component in this circuit. One of the control signals (pin 9) of 4052 is grounded so that 4052 will act as a two channel multiplexer and other control input (pin 10) is being terminated as control signal of TDM Multiplexer. The message signals that are to be transmitted has to be connected to CH1 and CH2 inputs of the CD 4052. TL084 has been used as a buffer at input of the CD4052 in order to avoid loading effect on signal sources. For pin configuration and other operating conditions of IC 4052 you can go through the data sheet provided along with manual.

Operation:

When control signal is 1 i.e. control signal is logic high (+5V), output of the 4052 is connected to CH1 and whatever signal at CH1 is there, it will be passed on to output. Similarly when control signal is 0 (0V), output of the multiplexer (4052) is connected to CH2 and the signal at this input will be passed on to output. In summary:

- If control signal is 1..... CH1 will be connected to the multiplexer output
- If control signal is 0..... CH2 will be connected to the multiplexer output.

When we drive this control signal with a clock signal (square wave) of desired frequency the output of the multiplexer will be connected to CH1 and CH2 automatically in sequence depend on clock signal condition i.e. ON/OFF time. So

multiplexer samples each the input signals in sequence and transmits over a single channel. If you make one of the input signal amplitude to zero, output of the multiplexer is a simple natural top PAM signal otherwise it is a Time Division Multiplexed PAM. In the same way we can use any pulse modulation technique (i.e. PWM, PPM, PCM) instead of PAM, here we have selected PAM.

Output of the TDM multiplexer has been terminated with RCA Socket which is useful to connect a cable (represents actual transmission line) to transmit TDM-PAM signal to de-multiplexer (55D).

2 Channel TDM De-multiplexer:

Figure 1:2 shows the TDM De-multiplexer using 4052. IC CD 4052 is a 4 channel analog multiplexer/de-multiplexer and is used as an active component in this circuit. One of the control signals (pin 9) of 4052 is grounded so that 4052 will act as a two channel de-multiplexer and other control input (pin 10) is being terminated as control signal of TDM de-multiplexer. The TDM signal that has to be de-multiplexed is connected to a common input (pin 13). De-multiplexed outputs are taken as CH1 (from pin 14) and CH2 (pin 12). Outputs of the de-multiplexer can be connected to low pass filters for smoothing the output. TL084 has been used as a buffer at input of the CD4052 in order to avoid loading effect on signal sources. For pin configuration and other operating conditions of IC 4052 you can go through the data sheet provided along with manual.

Operation:

When control signal is 1 i.e. control signal is logic high (+5V), input of the 4052 is connected to CH1 output and the signal which is at input will be passed on to output. Similarly when control signal is 0 (0V), input of the multiplexer (4052) is connected to CH2 output and the signal at the input will be passed on to output. In summary:

- If control signal is 1.....De-multiplexer Input will be connected to CH1output.
- If control signal is 0.....De-multiplexer Input will be connected to CH2 output.

When we drive this control signal with a clock signal (square wave) of desired frequency the input of the de-multiplexer will be connected to CH1 output and CH2 output automatically in sequence depend on clock signal condition i.e. ON/OFF time. So a part of the TDM-PAM signal will be passed on to CH1 and CH2 in sequence depend on clock signal condition i.e. to CH1 during ON time and to CH2 during OFF time. Output of CH1 and CH2 are simple natural top PAM signals. The original AF signals can be recovered from these PAM signal by smoothening with low pass filters. These recovered AF signals can be amplified with the help of amplifiers provided on board if desired.

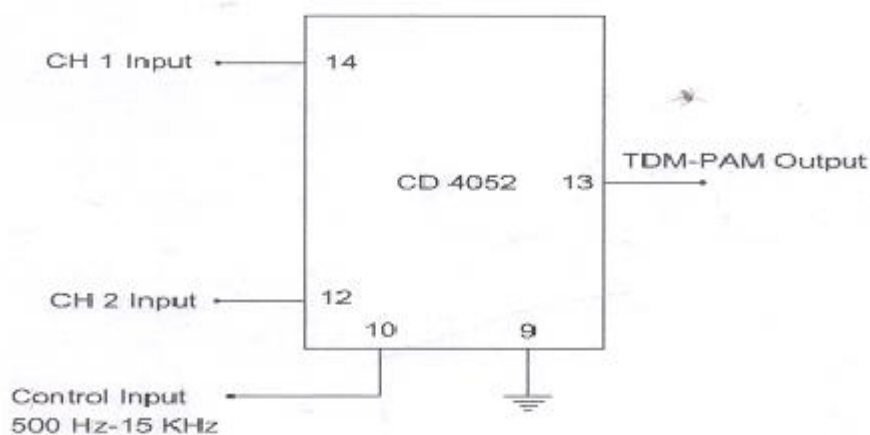
Synchronization:

De-multiplexer at the receiver should be operate in synchronize with the multiplexer which is at transmitter in order to recover the original signals without any loss. In actual communication system a synch signal will be added to the TDM signal and the clock generator at receiver will be triggered with the synch signal. In general these synch signals will be a over sized in amplitude or in width depends on the modulation technique which we adapts. In this trainer synch signal is not added in TDM, anyhow facility has been provided to study the effect of synchronization on recovered signal. For this a clock generator (which represents a local oscillator) is provided in de-multiplexer kit , and we can use the same clock which is used at transmitter .

BLOCK DIAGRAM:

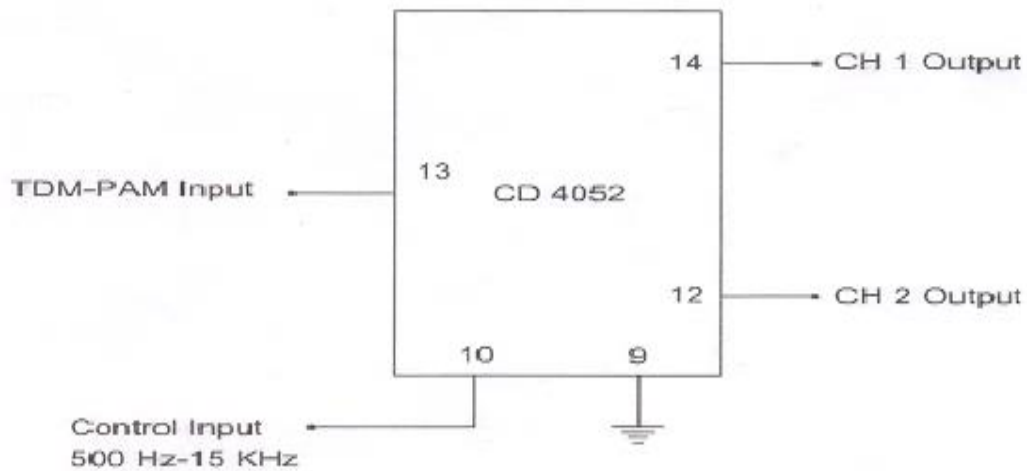
Multiplexer

Figure 1:1 TDM Multiplexer



Demultiplexer

Figure 1:2 TDM De-Multiplexer



EXPERIMENTAL PROCEDURE:

Multiplexer:

1. Study the theory of operation.
2. Connect the trainer TDM Multiplexer to the mains and switch on the power supply.
3. Measure the output of the regulated power supply i.e. +5V and -5V with the help of digital multimeter.
4. Observe the output of the AF generator-1 using CRO, it should be a Sine wave of 400 Hz frequency with 3 V_{pp} amplitude.
5. Observe the output of the AF generator-2 using CRO it should be a Sine wave of 200 Hz frequency with 3 V_{pp} amplitude.
6. Verify the operation of logic source with multimeter/scope, output should be +5V in logic 1 position and 0V in logic 0 position.
7. Observe the output of the Clock generator using CRO, it should be a Square wave of 500 Hz to 15 KHz frequency with 5 V_{pp} amplitude.
8. Now connect the CH1 & CH2 Inputs of the TDM multiplexer to the outputs of the AF Generator 1 and 2 respectively.

9. Connect Control input of the TDM multiplexer to the output of the logic source.
10. Put control signal (logic source) at logic 1 condition and observe the output of the TDM multiplexer with the help oscilloscope, by this we can notice that the output of the TDM multiplexer is a signal which has been connected to CH1 input. In this condition the signal at CH2 input has no effect on multiplexer output.
11. Similarly put logic source at logic 0 position and observe the output of the TDM multiplexer. Now notice that the output of the TDM multiplexer is a signal which has been connected to the CH2 input and the signal at CH1 input has no effect on multiplexer output.
12. Now disconnect logic source and connect clock output to the control input.
13. Observe TDM wave form using CRO at different values of clock frequency, input signal voltage levels and sketch them.

Note1: After setting the clock frequency and input signals to desire values put storage scope in STOP mode so that you can view stable display of waveforms.

Note2: Sample wave forms given in Figure 1:3, 1:4 are drawn at 1 KHz sampling clock, you can take at any clock frequency.

14. Similarly you can observe and plot the TDM waveforms for different inputs i.e. DC signals alone, AC&DC instead of AC signals.

Note1: DC Signals (voltages) can be connected from an external sources and care should be taken in case voltage levels i.e. maximum voltage input voltage must be in range of $\pm 4.8V$.

Note2: You can use even normal scope, when you observe the TDM wave form for DC inputs.

De-multiplexer:

15. Study the theory of operation.
16. Connect the trainer TDM DE-multiplexer to the mains and switch on the power supply.
17. Measure the output of the regulated power supply i.e. +5V and -5V with the help of digital multimeter.
18. Verify the operation of logic source with multimeter/scope, output should be +5V in logic1 position and 0V in logic 0 position.
19. Observe the output of the Clock generator using CRO, it should be a Square wave of 500 Hz to 15 KHz frequency with 5 Vp amplitude.

20. Connect TDM-PAM signal to input of TDM de-multiplexer from TDM multiplexer with the help of co-axial cable (supplied with trainer).
21. Connect control input to logic source output.
22. Keep CRO in dual mode; connect one input to CH1 output and another input to CH2 output.
23. Put logic source to 1 position and observe CH1 and CH2 outputs. You can notice that the entire TDM signal is transferred to CH1 output and has no signal at CH2 output.
24. Similarly put logic source to 0 positions and observe CH1 and CH2 outputs. Now the entire TDM signal is transferred to CH2 output and has no signal at CH1 output. By the above two steps you can notice that the entire TDM signal is transferred to CH1 output when control input is 1 and to CH2 output when control input is 0.
25. Now disconnect logic source and connect clock from the transmitter (i.e., AET- 55M) through a coaxial cable.
26. Observe CH1 and CH2 outputs. You will notice that the outputs are natural top sampled PAM signals.
27. Connect CH1, CH2 outputs to low pass filters and observe the output of the filters and compare them with the original AF Signals (at multiplexer inputs) using CRO. You will notice that both the signals are same in frequency and shape. Signal amplitude may be attenuated during smoothing process and this can be achieved by taking amplifiers output. Select AC/DC coupling depending on the input signal.
28. Observation on effect of synchronization:
29. Disconnect clock from multiplexer (AET-55M) and connect to local oscillator (i.e., clock generator output from AET-55D) with remaining setup as it is. Observe CH1 and CH2 outputs and compare them with the previous results. These signals are little bit distorted in shape. This is because lack of synchronization between clock at multiplexer and clock at de-multiplexer. You can get further perfection in output wave forms by adjusting the locally generated clock.

PRECAUTIONS:

1. Connections must be tight.
2. Waveforms must be noted carefully.

VIVA QUESTIONS:

- What is meant by multiplexing technique and what are the different types of Multiplexers?
- Briefly explain about TDM&FDM?
- What is the transmission band width of a PAM/TDM signal?
- Define crosstalk effect in PAM/TDM system?
- What are the advantages of TDM system?
- What are major differences between TDM&FDM?
- Give the value of T_s in TDM system?
- What are the applications of TDM system and give some example?
- What is meant by signal overlapping?
- Which type of modulation technique will be used in TDM?

RESULT:

Hence four message signals are transmitted at a time through a single communication channel, using TDM system and again de-multiplex these four message signals at receiver is observed.

FREQUENCY SHIFT KEYING

AIM: Study the operation of FSK modulation & Demodulation and to plot the FSK wave forms for Binary data at different frequencies.

EQUIPMENT REQUIRED:

- 1.Frequency Shift Keying system trainer
- 2.Dual trace Oscilloscope
- 3.Digital multimeter
- 4.Digital frequency counter

THEORY:

Frequency Shift Keying (FSK) is a modulation/ Data transmitting technique in which carrier frequency is shifted between two distinct fixed frequencies to represent logic 1 and logic 0. The low carrier frequency represents a digital 0 (space) and higher carrier frequency is a 1 (mark). FSK system has a wide range of applications in low speed digital data transmission systems. Wave forms are shown in Figure 1:3. FSK Modulating & Demodulating circuitry can be developed in number of ways, familiar VCO and PLL circuits are used in this trainer. Figure 1:1 and Figure 1:2 shows the FSK modulator and demodulator respectively.

FSK Modulator:

Figure 1:1 shows the FSK modulator using IC XR 2206. IC XR 2206 is a VCO based monolithic function generator capable of producing Sine, Square, Triangle signals with AM and FM facility. In this trainer XR2206 is used generate FSK signal. Mark (Logic 1) and space (logic 0) frequencies can be independently adjusted by the choice of timing potentiometers F0 & F1. The output is phase continuous during transitions. The keying signals i.e. data signal is applied to pin 9.

FSK Demodulator:

Figure 1:2 shows FSK Demodulator, is a combination of PLL (LM565) and comparator (Op-amp). The frequency-changing signal at the input to the PLL drives the phase detector to result in rapid change in the error voltage, which is applied to the

input of the comparator. At the space frequency, the error voltage out of the phase detector is below the comparison voltage of the comparator. The comparator is a non-inverting circuit, so its output level is also low. As the phase detector input frequency shifts low (to the mark frequency), the error voltage steps to a high level, passing through the comparison level, causing the comparator output voltage to go high. This error voltage change will snap the comparator output voltage between its two output levels in a manner that duplicates the data signal input to the XR2206 modulator.

The free running frequency of the PLL (no input signal) is set midway between the mark and space frequencies. A space at 2025 Hz and mark at 2225 Hz will have a free running VCO frequency of 2125 Hz.

CICUIT DIAGRAM:

Figure 1:1 FSK Modulator using XR2206

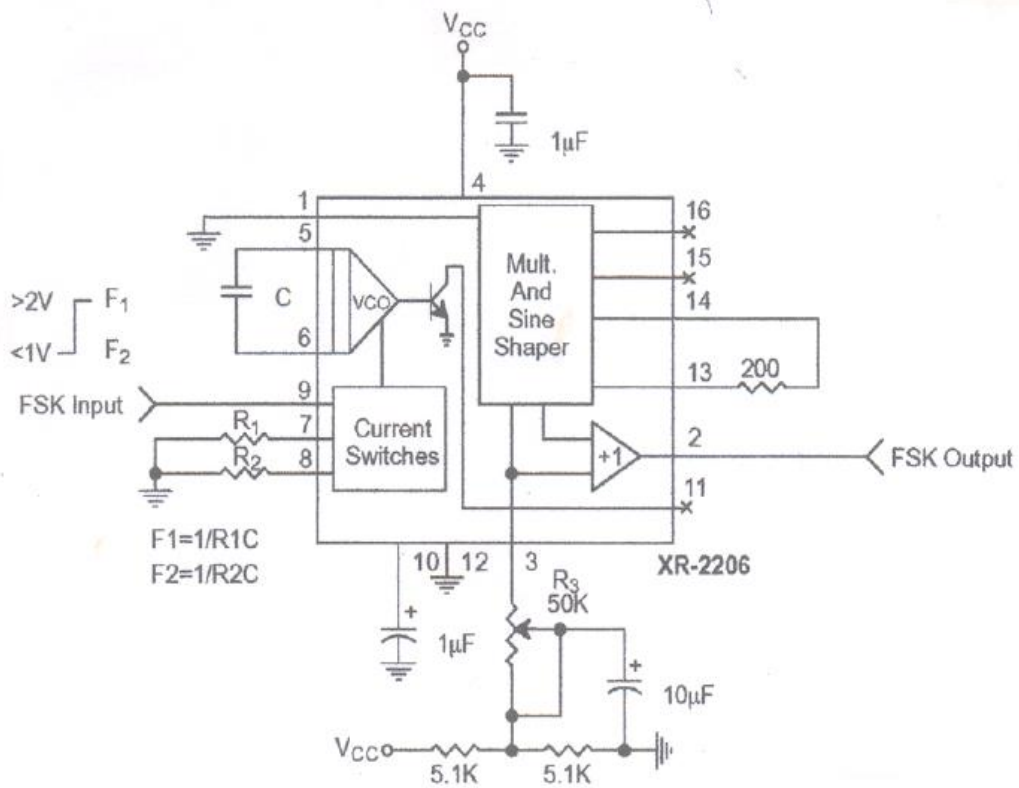
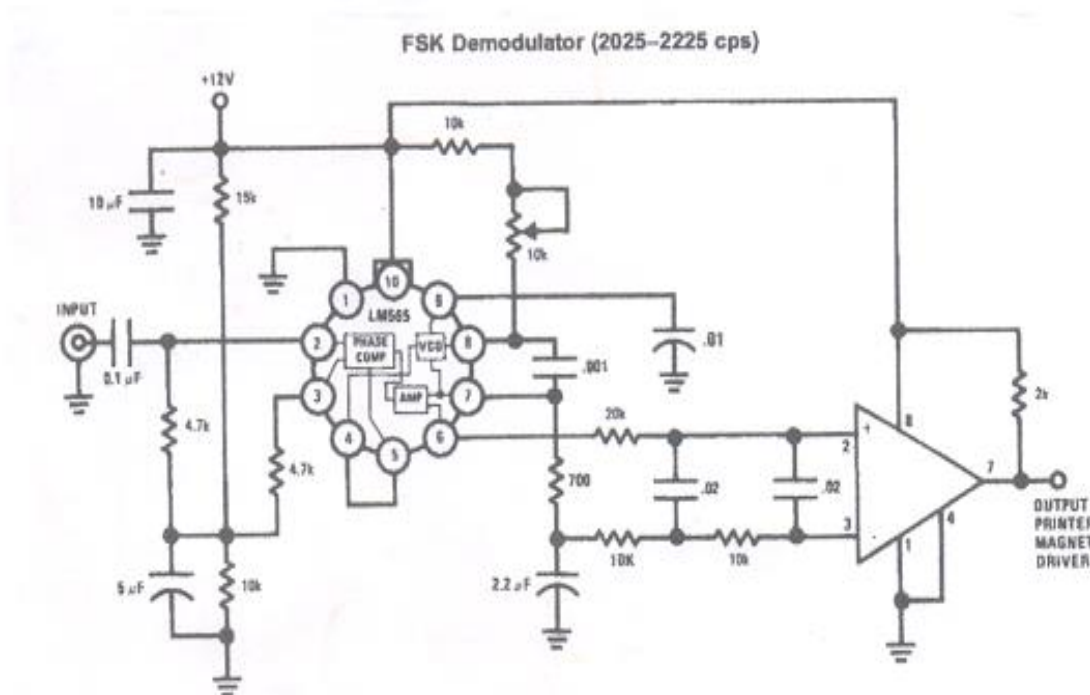


Figure 1:2 FSK De Modulator using LM565



EXPERIMENTAL PROCEDURE:

1. Study the theory of operation.
2. Connect the trainer to mains and switch on the power supply.
3. Measure the output voltage of the regulated power supply i.e. +12V with the help of digital multimeter.
4. Verify the operation of the logic source using digital multimeter. Output should be zero volts in Logic 0 position and 12V in logic 1 position.
5. Observe the output of the data signal using Oscilloscope. It should be a square wave of 20 Hz to 180 Hz @ 10Vpp. (For frequency variation potentiometer is provided)

FSK Modulation:

1. Connect output of the logic source to data input of the FSK Modulator.
2. Set logic source switch in 0 positions.
3. Connect FSK modulator output to Oscilloscope as well as frequency counter.
4. Set the output frequency of the FSK modulator as per your desire (say 1.2 KHz) with the help of control F0 which represents logic 0.

5. Set logic source switch in 1 position.
6. Set the output frequency of the FSK modulator as per your desire (say 2.4 KHz) with the help of control F1 which represents logic 1.

Note: We have chosen F0 as 1.2 KHz and F1 as 2.4 KHz for ease of operation, in fact you may set any value.

3. Now connect data input of the FSK modulator to the output of the data signal generator.
4. Keep CRO in dual mode connect CH1 input of the oscilloscope to the input of the FSK modulator and CH2 input to the output of the FSK modulator.
5. Observe the FSK signal for different data signal frequencies and plot them. By this we can observe that the carrier frequency is shifting between two predetermined frequencies as per the data signal i.e. 1.2 KHz when data signal is 0 and 2.4 KHz when data input is 1 in this case.
6. Compare these plotted wave forms with the theoretically drawn in figure 1:3.

FSK Demodulation:

7. Again connect input of the FSK modulator to the logic source and put data source switch in 0 positions.
8. Connect the frequency counter to the output of the FSK modulator output.
9. Set FSK output frequency to 2025 Hz with the help of FO control.
10. Now put data source switch in 1 position and set the FSK output frequency to 2225 Hz with the help of F1 control without disturbing the FO.

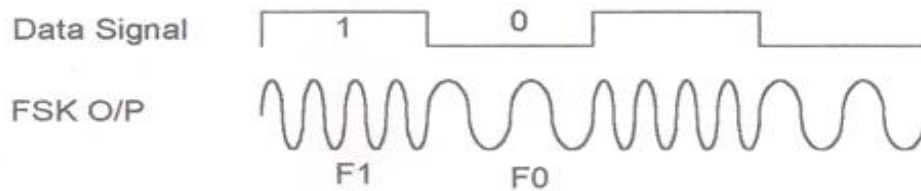
Note: As per one of the standards, for proper demodulation of FSK signal the F0 should be 2025 Hz and F1 should be 2225 Hz.

11. Disconnect the FSK input of the modulator from logic source and connect to the data signal generator.
12. Observe the output of the modulator using CRO and compare them with given waveforms in figure 1:3.
13. Now connect the FSK modulator output to the FSK input of the demodulator.
14. Connect CH1 input of the Oscilloscope to the data signal at modulator and CH2 input to the output of the FSK demodulator (keep CRO in dual mode).
15. Observe and plot the output of the FSK demodulator for different frequencies of data signal. Compare the original data signal and demodulated signal; by

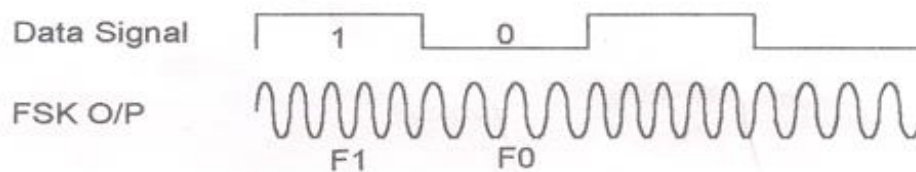
this we can observe that there is no loss in process of FSK modulation and demodulation.

EXPECTED WAVEFORMS

FSK Wave form for F0 (1.2 KHz) and F1 (2.4 KHz)



FSK Wave form for F0 (2025 Hz) and F1 (2225 Hz)



VIVA QUESTIONS:

1. Define Binary FSK signal?
2. What is meant by carrier swing?
3. Define Frequency deviation of FSK signal?
4. What are the advantages of this FSK signal?
5. Give the differences between FSK & FM?

RESULT:

Hence obtained the FSK signal for a given input data and also obtained the demodulated data the given FSK signal.

PHASE SHIFT KEYING

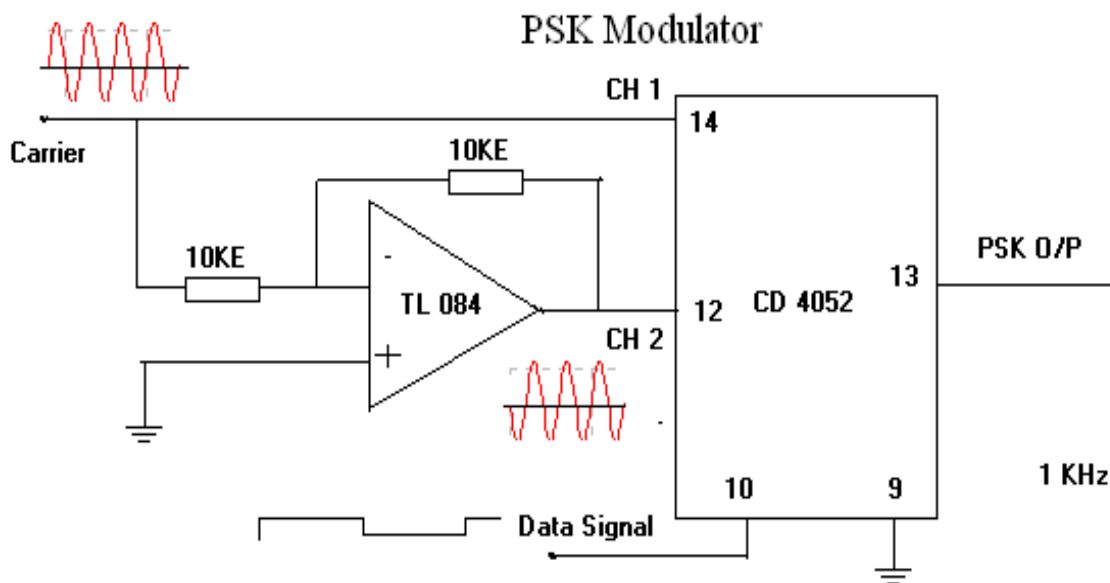
AIM: Study the operation of PSK (Binary) Modulation & Demodulation and to plot the PSK wave forms for Binary data at different frequencies.

APPARATUS:

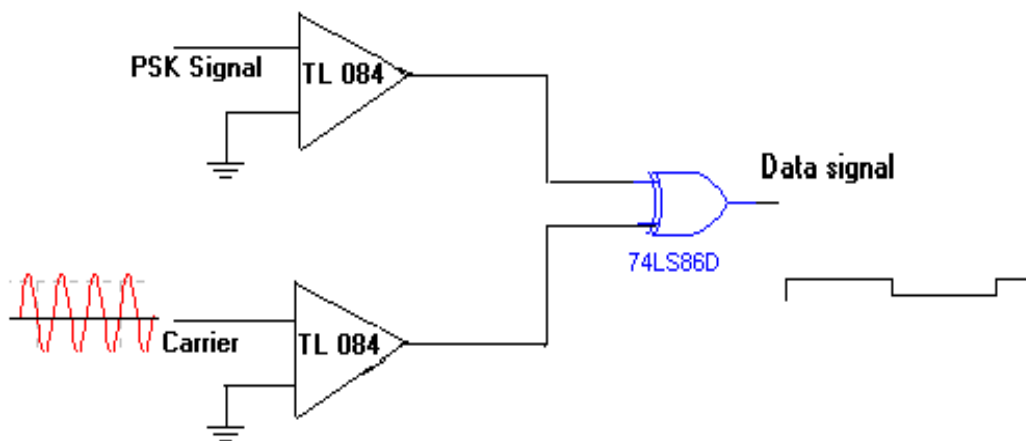
1. Phase Shift keying trainer-
2. Dual trace Oscilloscope
3. Digital multimeter
4. Patch chords

THEORY: Phase Shifting Keying (PSK) is a modulating / Data transmitting technique in which phase of the carrier signal is shifted between two distinct levels. In a simple PSK (i.e. Binary PSK) unshifted carrier $V \cos W_o t$ is transmitted to indicate a 1 condition, and the carrier shifted by 180° i.e. $-V \cos W_o t$ is transmitted to indicate a 0 condition. Wave forms are shown in Figure PSK Modulating & Demodulating circuitry can be developed in number of ways; one of the simple circuits is used in this trainer.

BLOCK DIAGRAM:



PSK Demodulator



PROCEDURE:

1. Study the theory of operation.
2. Connect the trainer to mains and switch on the power supply.
3. Measure the output of the regulated power supply i.e +5V and -5V with the help of digital multimeter.
4. Observe the output of the carrier generator using CRO, it should be an 8 KHz Sine with 5 Vpp amplitude.
5. Observe the various data signals (1 KHz, 2 KHz and 4 KHz) using CRO.

Modulation

6. Connect carrier signal to carrier input of the PSK Modulator.
7. Connect data signal say 4 KHz from data source to data input of the modulator.
8. Keep CRO in dual mode.
9. Connect CH1 input of the CRO to data signal and CH2 to the output of the PSK Modulator
10. Observe the PSK o/p Signal with respect to data signal and plot the wave forms
Compare the plotted waveforms with given wave forms.

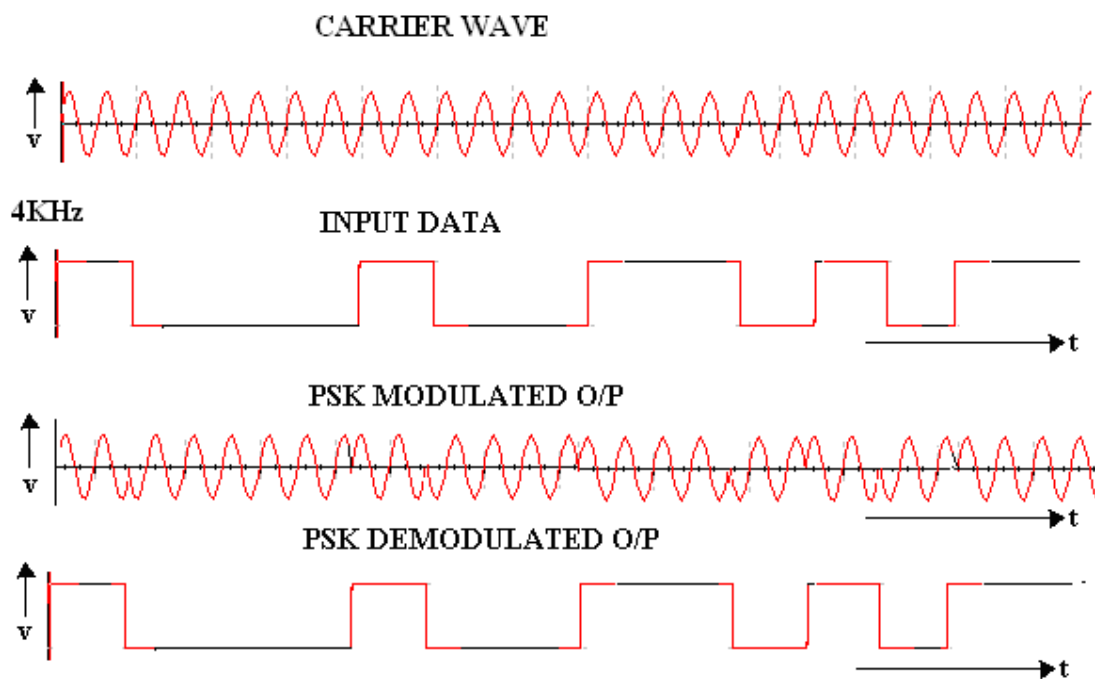
Demodulation:

11. Connect the PSK output to the PSK input of the demodulator.
12. Connect carrier to the carrier input of the PSK demodulator

Note: In actual communication system reference carrier is generated at receiver.

13. Keep CRO in dual mode.
14. Connect CH1 to the data signal (at Modulator) and CH2 to the output of the demodulator.
15. Compare the demodulated signal with original data signal, By this we can notice that there is no loss in modulation and demodulation process.
16. Repeat the steps 7 to 15 with different data signals i.e 2 KHz and 1 KHz.

EXPECTED WAVEFORMS:



RESULT:

QUESTIONS:

- Explain the concept of PSK?
- Compare ASK, FSK, PSK?
- Draw the waveforms of PSK?
- What is M-ary signaling? What are its advantages over 2-ary signaling?
- Explain the demodulation scheme of PSK?.
- What is the advantage of PSK over ASK, FSK?
- Will the smaller variations in the signal can be detected reliably by PSK?

DIFFERENTIAL PHASE SHIFT KEYING

AIM: Study the characteristics of differential phase shift keying.

APPARATUS:

1. Differential Phase Shift Keying Kits
2. C.R.O (20MHz)
3. Digital multimeter.
4. No's of coaxial cables (standard accessories with trainer)

THEORY:

DPSK: Phase Shift Keying requires a local oscillator at the receiver which is accurately synchronized in phase with the un-modulated transmitted carrier, and in practice this can be difficult to achieve. **Differential Phase Shift Keying (DPSK)** over comes the difficult by combining two basic operations at the transmitter (1) differential encoding of the input binary wave and (2) phase shift keying – hence the name differential phase shift keying. In other words DPSK is a no coherent version of the PSK.

The differential encoding operation performed by the modulator is explained below Let $b(t)$ be the binary message to be transmitted. An encoded message stream $b(t)$ is generated from $b'(t)$ by using a logic circuit The first bit in $b(t)$ is arbitrary which may be chosen as 1 or 0 . The subsequent bits in $b(t)$ are determined on the basis of the rule that when $b'(t)$ is 1 $b(t)$ does not change its value fig 1.4 shows two possible bit streams $b(t)$ and the respective phases. In the first bit stream, the initial bit (arbitrary) is 1 and in the second bit stream, the initial bit is 0 EX-NOR gate can be used to perform this operation as its output is a 1 when both the input are same, and a 0 when the inputs are different.

Figure 1.4:

$b'(t)$	0	1	1	0	0
$b(t)$	1	0	0	0	1
Phase	0^0	180^0	180^0	180^0	0^0

B(t)	0	1	1	1	0	1
Phase	180^0	0^0	0^0	0^0	180^0	0^0

Figure 1.5 Example for Complete DPSK operation (with arbitrary bit as 0):

Message signal(to be transmitted)	0	1	1	0	0	
Encoded data(differential data)		0	1	1	1	0
Transmitted signal phase:		180^0	0^0	0^0	0^0	180^0
Received signal phase :		180^0	0^0	0^0	0^0	180^0
Encoded data(differential data)		0	1	1	1	0
Message signal (Demodulation)		0	1	1	0	0

Dpsk Demodulator:

Fig 1.1 shown the DPSK modulator. This consists of PSK modulator and differential encoder.

PSK Modulator: IC CD 4052 is a 4 channel analog multiplexer and is used as an active component in this circuit. One of the control signals of 4052 is grounded so that 4052 will act as a two channel multiplexer and other control is being connected to the binary signal i.e., encoded data . Un shifted carrier signal is connected directly to CH1 and carrier shifted by 180^0 is connected to CH2. Phase shift network is a unity gain inverting amplifier using Op-Amp (TL084).

When control signal is at high voltage, output of the 4052 is connected to CH1 and un shifted (or 0 phase) carrier is passed on to output. Similarly when control signal is at zero voltage output of 4052 is connected to CH2 and carrier shifted by 180^0 is passed on to output.

Differential encoder: This consists of 1 bit delay circuit and an X-NOR Gate. 1 bit delay circuit is formed by a D-Latch. Data signal i.e., signal to be transmitted is connected to one of the input of the X-NOR gate and other one being connected to out of the delay circuit. Output of the X-NOR gate and is connected to control input of the multiplexer (IC 4052) and as well as to input of the D-Latch. Output of the X-NOR gate is 1 when both the inputs are same and it is 0 when both the inputs are different.

Dpsk Demodulator:

Fig 1.2 shows the DPSK Demodulator. This consists of 1 bit delay circuit, X-NOR Gate and a signal shaping circuit. Signal shaping circuit consists of a Op-amp based zero crossing detector followed by a D-latch. Receiver DPSK signal is converted to square wave with the help of zero crossing and this square wave will pass through the D-Latch. So output of the D-latch is an encoded data. This encoded data is applied to 1 bit delay circuit as well as to one of the inputs of X-NOR gate. And output of the delay circuit is connected to another input of the X-NOR gate. Output of the X-NOR gate is 1 when both the inputs are same and it is 0 when both the inputs are different.

BLOCK DIAGRAM:

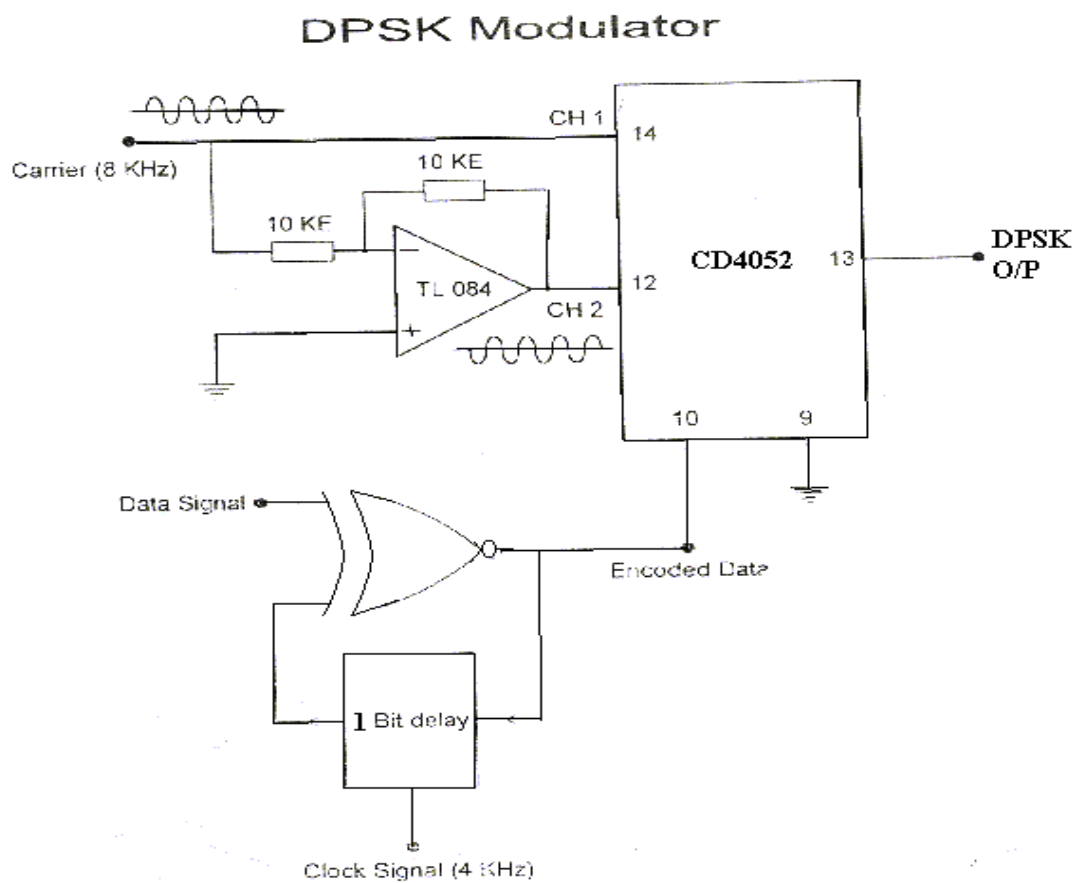
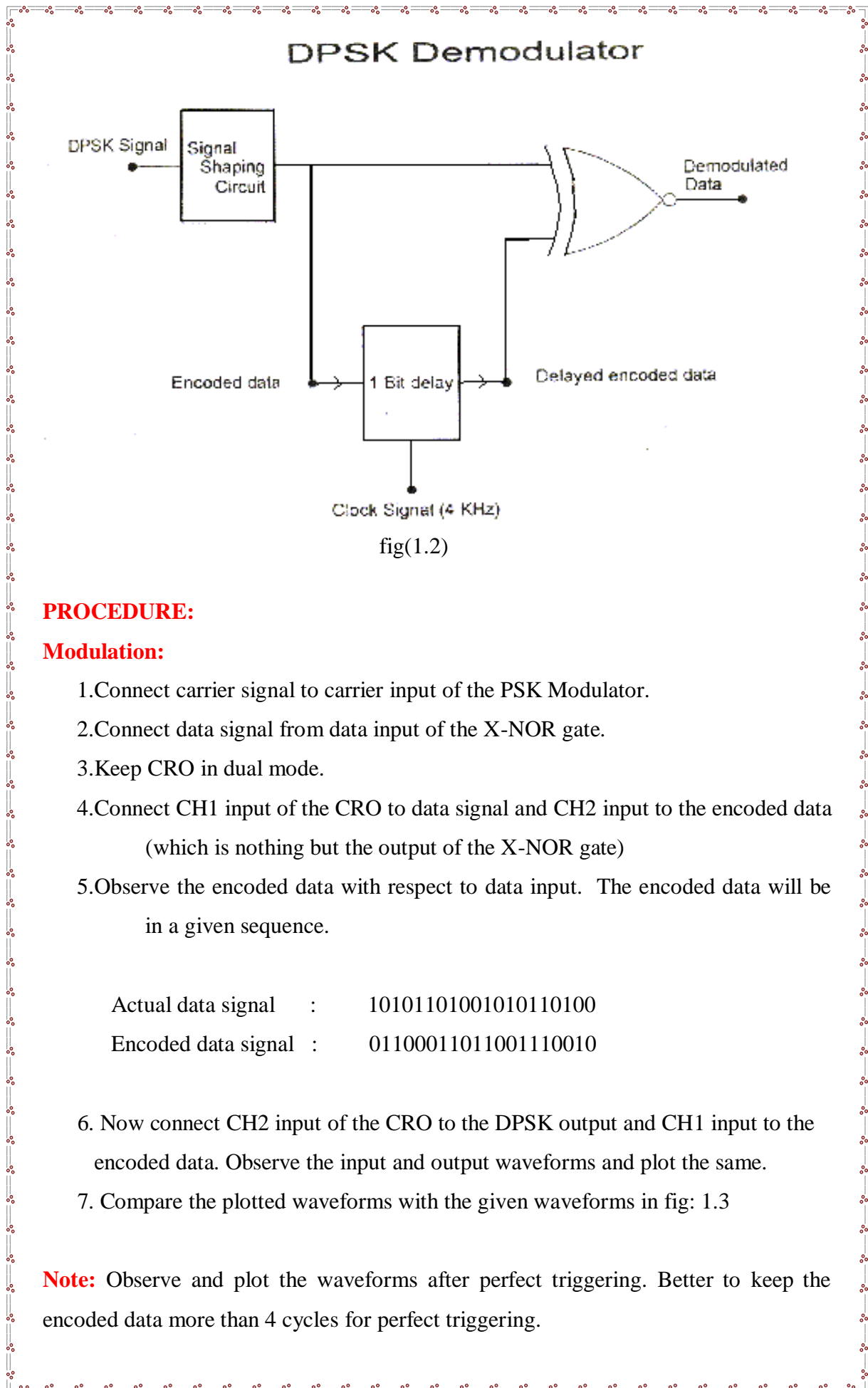


fig (1.1)



PROCEDURE:

Modulation:

1. Connect carrier signal to carrier input of the PSK Modulator.
2. Connect data signal from data input of the X-NOR gate.
3. Keep CRO in dual mode.
4. Connect CH1 input of the CRO to data signal and CH2 input to the encoded data (which is nothing but the output of the X-NOR gate)
5. Observe the encoded data with respect to data input. The encoded data will be in a given sequence.

Actual data signal : 10101101001010110100

Encoded data signal : 01100011011001110010

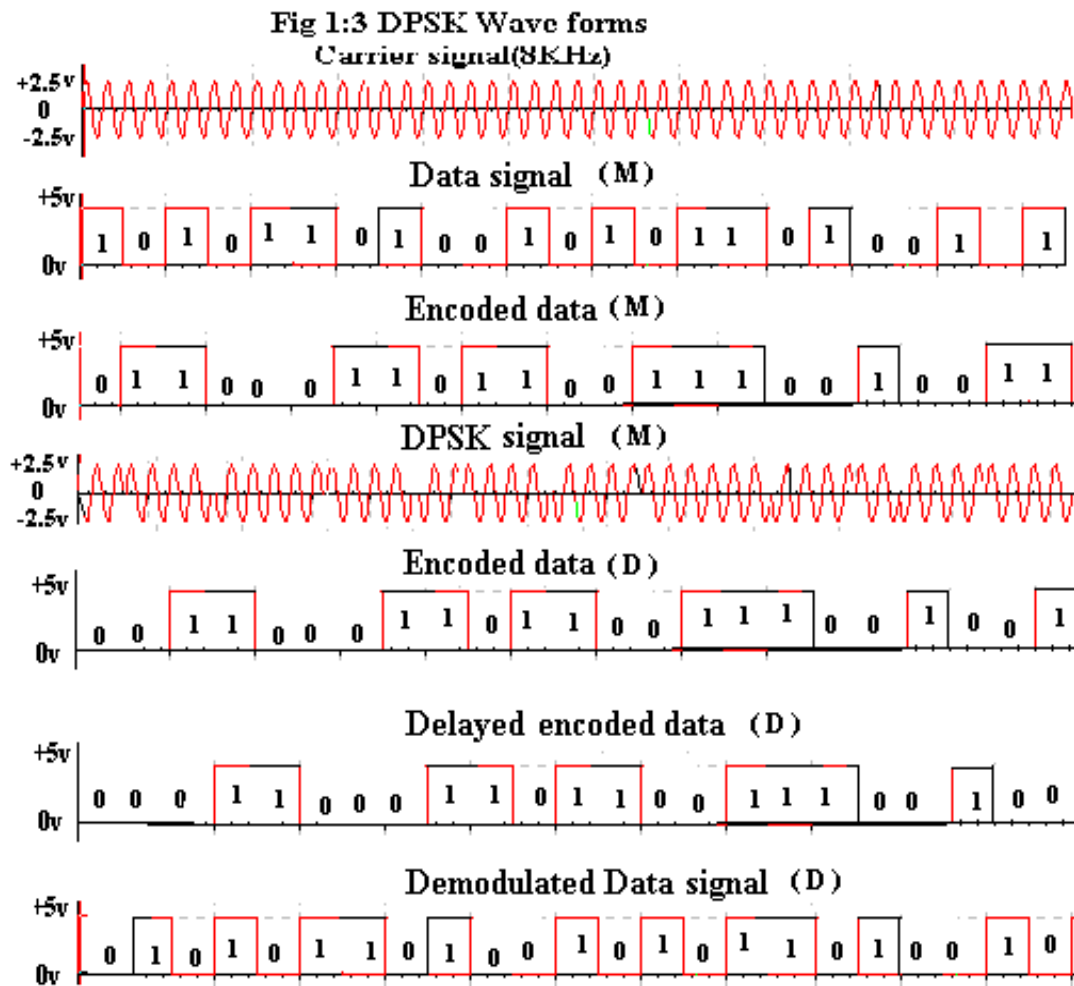
6. Now connect CH2 input of the CRO to the DPSK output and CH1 input to the encoded data. Observe the input and output waveforms and plot the same.
7. Compare the plotted waveforms with the given waveforms in fig: 1.3

Note: Observe and plot the waveforms after perfect triggering. Better to keep the encoded data more than 4 cycles for perfect triggering.

Demodulation:

1. Connect DPSK signal to the input of the signal shaping circuit from DPSK transmitter with the help of coaxial cable (supplied with trainer).
2. Connect clock from the transmitter (i.e. DPSK Modulator) to clock input of the 1 bit delay circuit using coaxial cable.
3. Keep CRO in dual mode. Connect CH1 input to the encoded data (at modulator) and CH2 input to the encoded data (at demodulator).
4. Observe and plot both the waveforms and compare it with the given waveforms. You will notice that both the signals are same with one bit delay.
5. Keep CRO in dual mode. Connect CH1 input to the data signal (at modulator) and CH2 input to the output of the demodulator.
6. Observe and plot both the waveforms and compare it with the given waveforms. You will notice that both the signals are same with one bit delay.
7. Disconnect clock from transmitter and connect to local oscillator clock (i.e., clock generator output from De Modulator) with remaining setup as it is. Observe demodulator output and compare it with the previous output. This signal is little bit distorted. This is because lack of synchronization between clock at modulator and clock at demodulator. You can get further perfection in output waveform by adjusting the locally generated clock frequency by varying potentiometer.

EXPECTED WAVEFORMS



RESULT:

QUESTIONS:

1. How does DPSK differ from PSK?
2. Explain theoretical modulation & demodulation of DPSK using arbitrary bit Sequence and assuming initial bit 0 and 1?
3. What is the advantage of DPSK over PSK?
4. Why do we need 1 bit delay in DPSK modulator & demodulator?
5. What does a synchronous detector (multiplier) do in DPSK demodulator?
6. What is the relation between carrier frequency & the bit interval 'T'?
7. What are the disadvantages of DPSK?
8. Is the error rate of DPSK is greater than PSK?
9. What is the expression for DPSK error?
10. What are the applications of DPSK?

PULSE AMPLITUDE MODULATION & DEMODULATION

AIM: To generate a Pulse amplitude modulated signal and also demodulate the original signal.

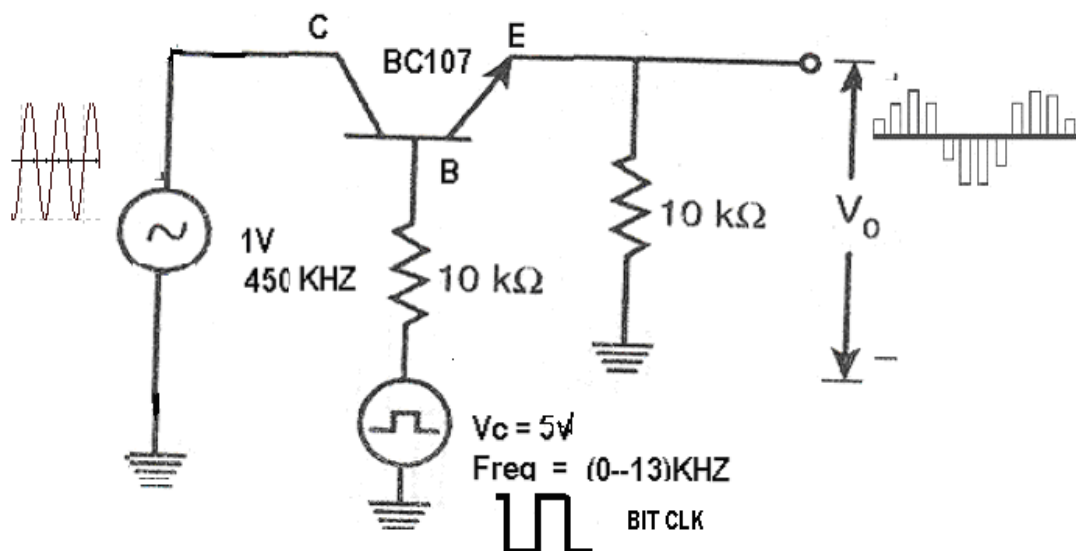
APPARATUS:

1. BC107 transistor-1
2. 10K Ω resistors -2
3. Function generators-2
4. CRO
5. Bread board
6. Connecting wires, probes

THEORY:

In pulse Amplitude Modulation the amplitude of the carrier pulses varies in accordance with the instantaneous values of message signal and the pulse width is fixed, in practice this is realized by a simple mechanical commutator or by electronic circuit. Natural PAM signal sampling occurs when finite width is used in the modulators and tops of the pulses are forced to follow the magnitude of modulating waveform. Flat topped PAM system is quite often used because of the generating the modulating waveform and spectrum. Finite width pulses are used but they are flat topped after modulation. Reconstruction of original signal is possible by passing the modulated signal through a low pass filter.

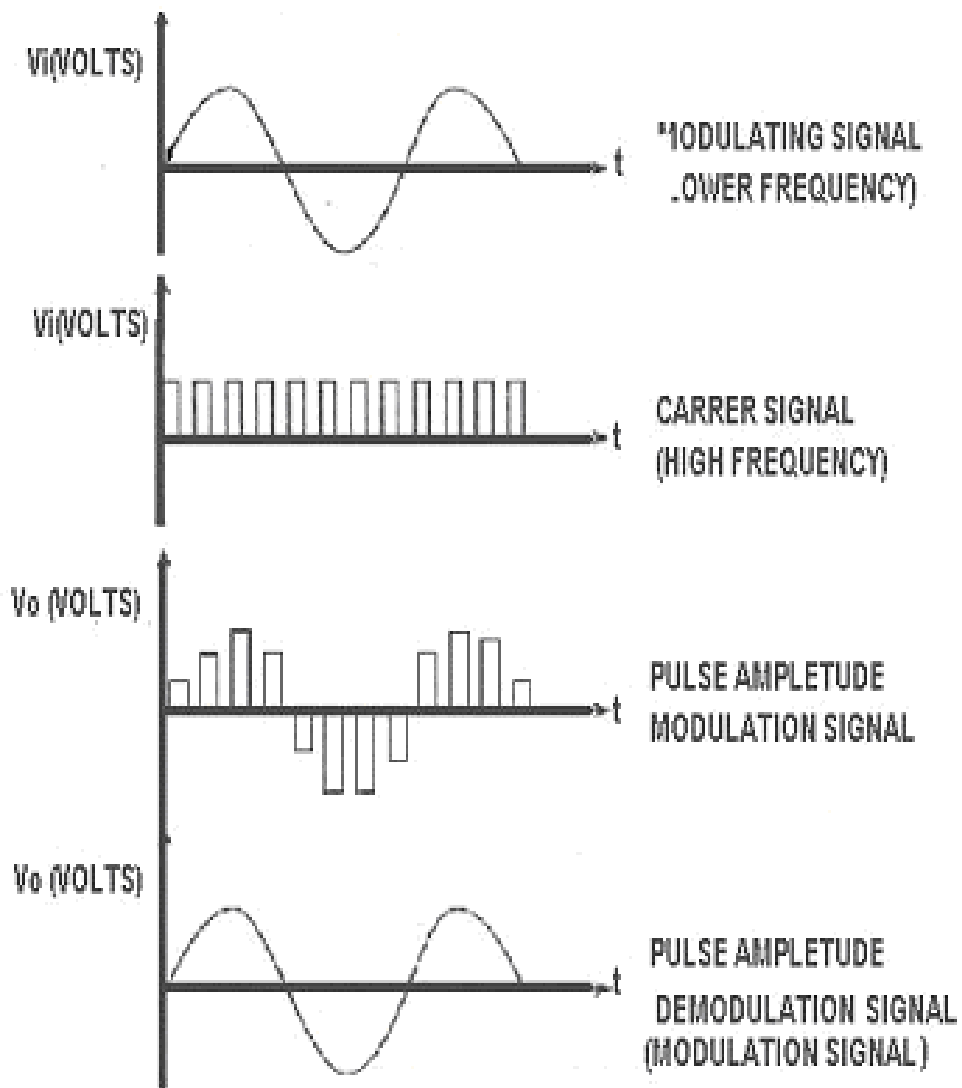
CIRCUIT DIAGRAM:



PROCEDURE:

1. Connections are made as per the circuit diagram.
2. Modulating signal is given to a collector terminal of a BC 107 transistors and carrier signal of High frequency is given to a base terminal of a transistor and switch ON the power supply.
3. Then the modulated output is observed across the transistor emitter terminal.
4. The output characteristics of a PAM signal is plotted on a graph.

OUTPUT WAVEFORM:



PRECAUTIONS:

1. Connections must be tight.
2. The width of the signal must be noted carefully.

VIVA QUESTIONS:

- What are the classifications of pulse modulation techniques?
- What is the transmission bandwidth of Pulse amplitude modulation?
- What are the Draw backs in Pulse amplitude modulated signal?
- What do you mean by synchronization in PAM?
- Write the standard equation of a PAM in frequency domain?
- What is meant by Aperture effect?
- Draw the frequency spectrum of a PAM signal?
- What is the time domain representation of a PAM signal?
- What are the major differences between PAM & PWM?
- Which type of sampling technique is used in PAM signal?

RESULT:

The output characteristic of a Pulse amplitude modulated/demodulated signal is obtained.

GENERATION AND DETECTION OF QPSK

AIM: To generate QPSK wave and detect it using QPSK module.

APPARATUS:

1. QPSK Modem Kit.
2. Power supply with regulated supply of +5V, \pm 12V.
3. CRO and CRO Probes.
4. Patch chords and Jumper wires.

THEORY:

QPSK Kit Description

QPSK Transmitter:

XR 2206 generates a master clock of frequency 10 KHz. The clock divider circuit consists of two numbers of 74HC161. Patch cord is used for selecting data rate between 600 bps or 300 bps. A 1C 74HC161 form a divide by 8 circuits which is used for getting a word pulse (WP). CD4014 generates the data. It converts the 8 bit parallel data in serial form. Data pattern can be selected through the DIP switch SW1. QPSK system (fig4.1) requires four signals $-\sin$, \sin , \cos and $-\cos$ which are generated using four number of UA741 through signal processing operation like inversion and differentiation. 74HC161 and CD 4094 form a two bit shift register used for serial to parallel conversion (Bit splitter). The two bit parallel output is used for selecting one of the four signal generated by IC's UA741. CD14053 and UA741 form a multiplexer and adder which is used for selecting one of the four signals of the QPSK modulation depends on the value of two bits of B_0 and B_1 .

QPSK Receiver:

QPSK receiver (fig4.2) consists of band pass filter and two parallel branches of multiplier, a low pass filter and a comparator. In coming QPSK signal is fed two parallel branches of multiplier, LPF and level converter. Multiplier is used to remove the carrier frequency and recover the and band signal. Then it is passed through a LPF to restrict the noise and made to fall within the base band signal bandwidth. The level converter is enables the decision circuit to decide '1' or '0' of transmitted bits. The local carrier is used in the multiplier are sine and cos signal, which ultimately generates two bit streams B_0 and B_1 at the output of level converters. These outputs are then fed to 2 bit parallel to serial bit combiner to recover the original data.

CIRCUIT DIAGRAM:

QPSK Transmitter

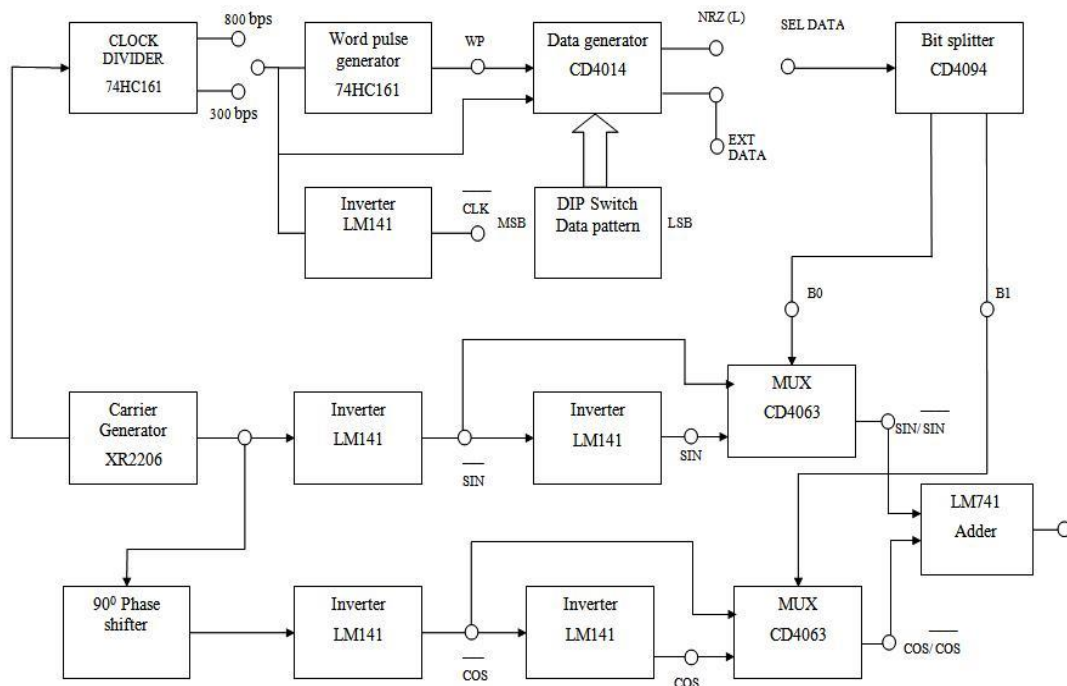


Fig4.1: Block diagram of a QPSK transmitter module.

QPSK Receiver

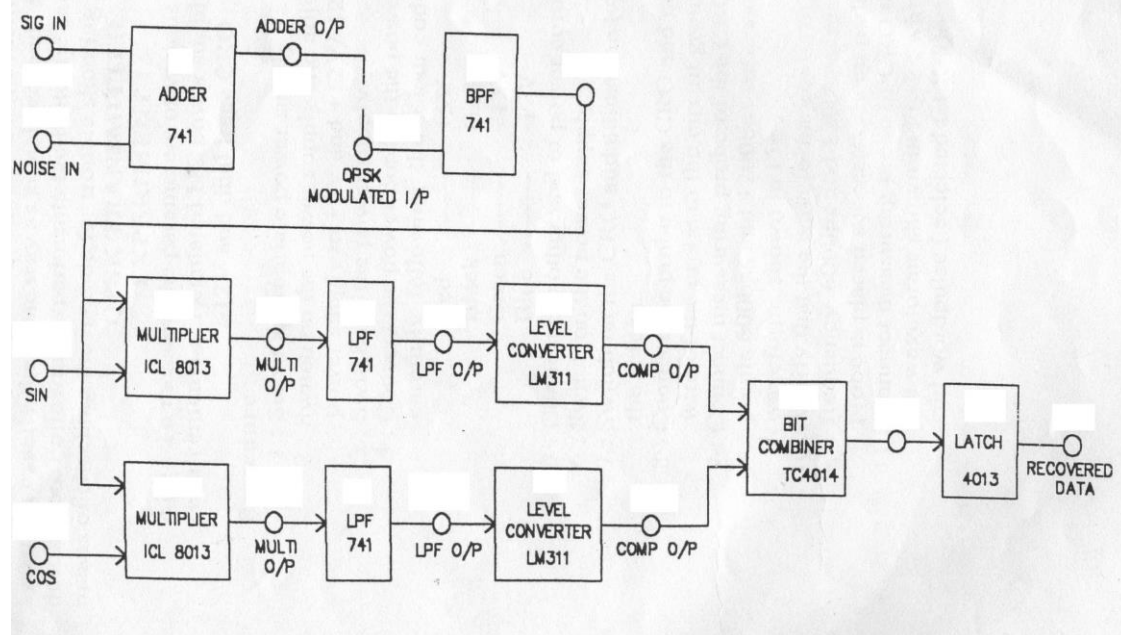


Fig4.2: Block diagram of a QPSK receiver module

PROCEDURE:

Transmitter

1. Connect the power supply to the QPSK Kit by using connector wire +5V, +12V, -12V (Fig4.1).
2. Select the input bit stream by combination of dip switches here input 8 bits data is selected. Select the proper transmission rate 600bps or 1200 bps.
3. Select NRZ (L) data at the output of data generator.
4. Measure the clock signal, word pulse and NRZ (L) input data stream. If transmission rate is 1200 bps then Word pulse width is $(1/1200=833 \mu\text{Sec})$.
5. Check the odd stream and even stream data output
6. If binary input data { 1 1 0 1 1 0 0 0 } the odd stream $B_1= 1 1 0 0 11 00$ and $B_0= 1 1 1 1 0 0 0 0$ as seen in figure
7. The four carrier signal generator with phase is selected according to the combination of the digits of input data stream (odd and even stream).
8. The output QPSK signal is verified with reference of Word pulse. Here word pulse is the start of 8 bits input data

Receiver

1. The QPSK modulated signal is fed to the receiver input (fig4.2).
2. Here odd stream B_1 offset and even stream B_0 offset is used to track the transmitter odd stream and even stream. So connect transmitter odd stream to one channel of CRO and receiver odd stream to another channel of CRO, and check the same bits pattern and repeat same for even stream.
3. Verify the demodulated output with transmitted message signal.
4. The input waveform and expected output waveforms are shown in fig 4.3.

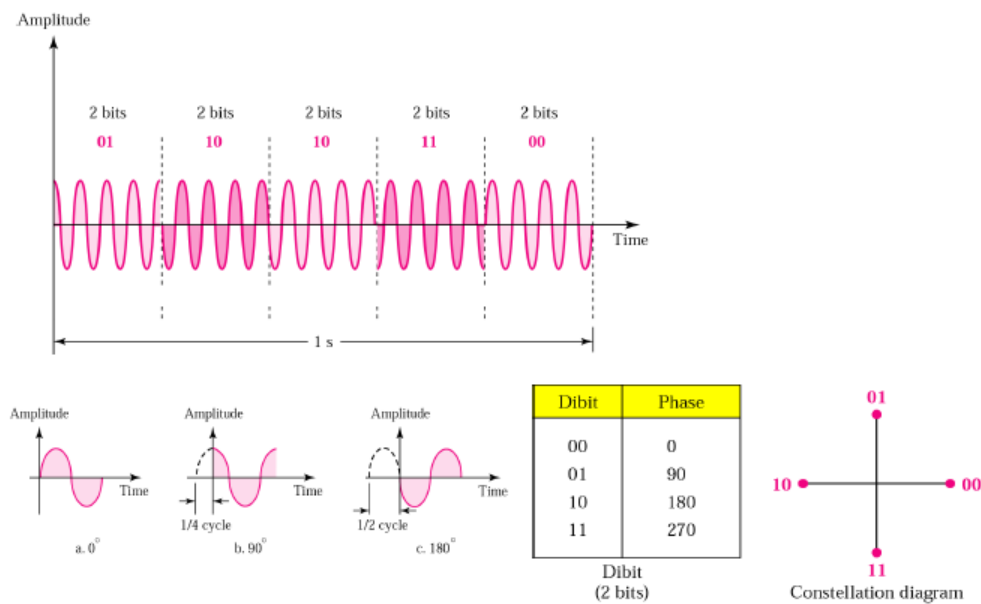


Fig4.3: QPSK output signal with constellation diagram

RESULT: