

# Experiment - 1

Aim: To study signal sampling and reconstruction techniques

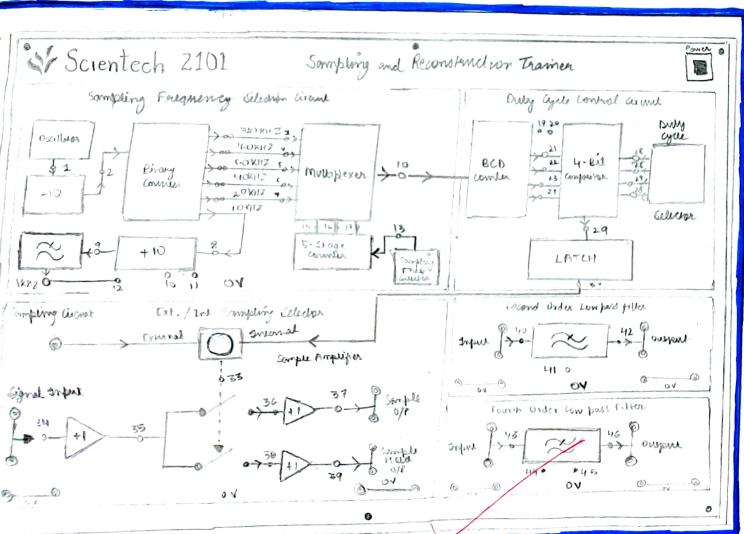
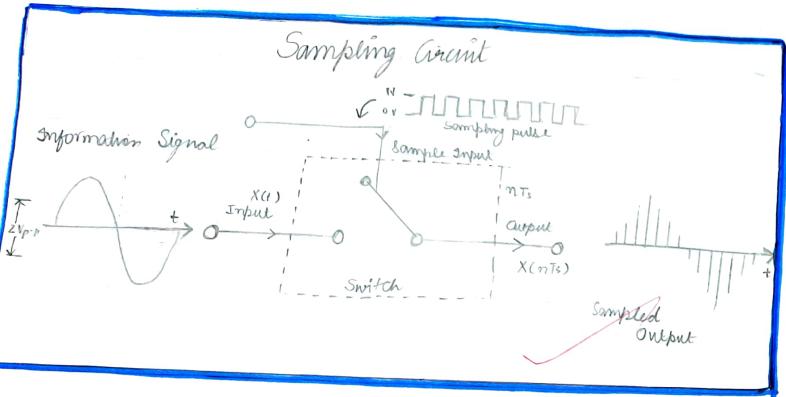
Objectives: Study of sampling and reconstruction of signal  
Study effect of sample / Hold sample on reconstructed Waveform  
Comparison of frequency response of 2<sup>nd</sup> and 4<sup>th</sup> order low pass filters

Equipments Required: Scientech 2101 with power / supply cords  
CRO with connecting probe  
connecting cords

## Theory:

Signals we use in real world, such as our voice are called analog signals. To process these signals in digital communication we need to convert analog to digital signals.

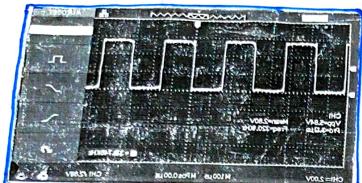
Analog signals are continuous in time and amplitude while digital signal is discrete time signal and has both time and amplitude discrete. SAMPLING is the process to convert continuous time signal to discrete time signal. The value of signal is measured at certain intervals in time which are called as samples. Sampling theorem states that a continuous signal band limited to  $f_m$  Hz can be completely represented by and reconstructed from the samples.



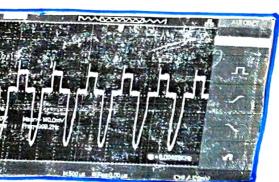
[ SIGNAL WAVE FORM VISUALIZATION ]



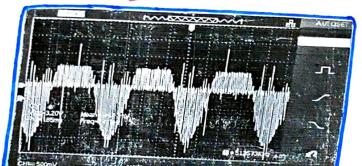
20 kHz sampling Frequency  
I/P signal waveform



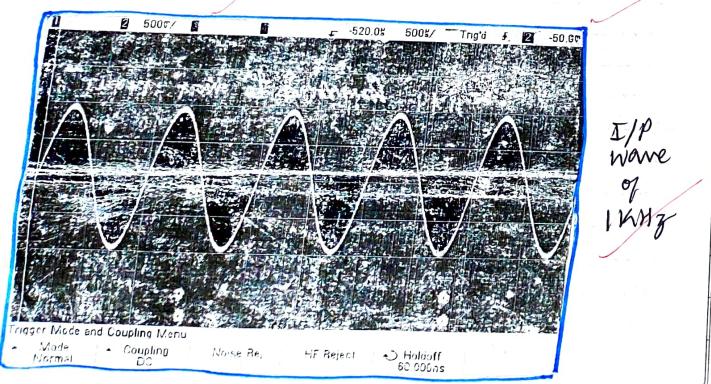
320 kHz sampling Frequency  
I/P signal waveform



Sampling O/P at sampling wave frequency 20 kHz



Sampling O/P at sampling wave frequency 320 kHz



I/P wave of 1 kHz

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per sec. Minimum frequency is called "Nyquist Rate". Thus for faithful reconstruction of information signal from its samples it is necessary that the sampling rate  $F_s > 2 F_{max}$  Hz.

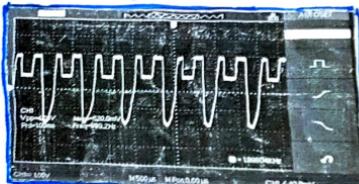
If the information signal is sampled at a rate lower than that stated by Nyquist criterion, then there is overlap between information signal and side bands of harmonics. Thus the lower and higher frequency components get mixed and cause unwanted signals to appear at the demodulator output this phenomenon is known as aliasing or fold-over distortion.

• Procedure

1. Duty cycle selector switch was set to 5 (50% duty cycle) and sampling selector switch was set to internal.
2. 1 kHz sine wave was given as input and Oscilloscope was used to observe the wave. Readings were taken at 320 kHz and 20 kHz sampling rates.
3. For the respective sampling rates sample O/P waveforms were visualized.
4. Sample input was connected to 2<sup>nd</sup> and 4<sup>th</sup> order low pass filter.
5. We observed sample output lower frequencies introduced distortion which was caused as filter did not attenuate unwanted frequency component significantly.
6. As we increased duty cycle selector, we observed linear relationship with amplitude.
7. Same procedure was repeated for sample and hold waveform.
8. Outputs for different duty cycles were observed too but

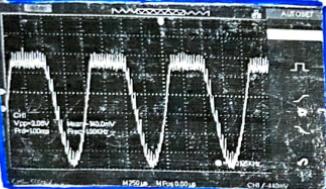
Teacher's Signature : \_\_\_\_\_

## [ SAMPLE AND HOLD O/P ]

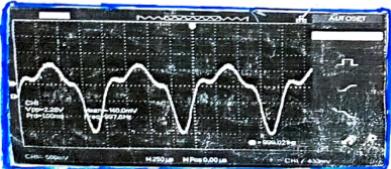


Sample and hold O/P wave for sampling wave of frequency 20kHz

Sample and hold O/P for sampling wave of frequency 320kHz

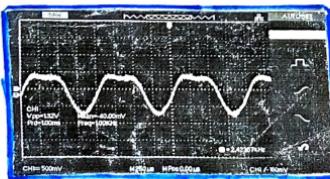


## [ OUTPUT WAVE WHEN SAMPLE OUTPUT PASSED THROUGH 4<sup>TH</sup> ORDER LOW PASS FILTER (LPF) ]



sample and hold O/P when passed through 4<sup>th</sup> order LPF for sampling frequency 20kHz

Sampling O/P when a wave I/P passed through 4<sup>th</sup> order LPF for the sampling frequency 320kHz I/P



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there was no change in amplitude in hold case

### Result :

It was observed from various cases that -

- As sample frequency increases, the output of sample port has more no. of samples of applied signal
- For transmitting the signal if a sample and hold amplifier is used just before transmission channel, the signal suffered from distortion as compared to when only sample amplifier was used
- As the order of LPF is increased at O/P, the recovered signal will be reconstructed more like transmitted signal

To perform verify when sample and sample and hold amplifiers were connected to II and IV order LPF Signal sampling was generated and various reconstruction techniques are verified and observed

## VIVA VOCE QUESTIONS

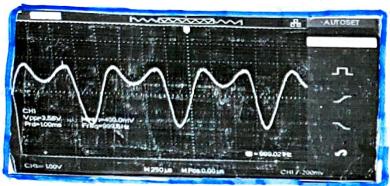
Ques 1. What do you mean by sampling?

Ans 1. To convert continuous time signal to discrete time signal, a process is used and this process is called sampling.

Ques 2. What is Sampling Theorem?

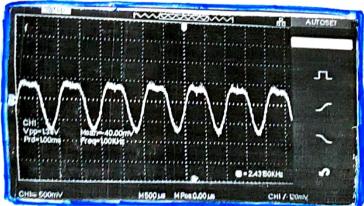
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[ OUTPUT WAVE WHEN SAMPLE AND HOLD WAVE IS PASSED THROUGH 4<sup>th</sup> ORDER LOW PASS FILTER ]



When sampling wave of frequency 20 kHz is given at input

In case of sampling wave of frequency 320 kHz



Ans 1. The sampling theorem states that a signal can be exactly reproduced if it is sampled at a frequency  $F_s$   
where  $F_s > 2F_{max}$  :  $F_{max}$  = maximum frequency in signal

Ques 2. What is Nyquist frequency?

Ans 2. The frequency  $2 \times F_{max}$  is called Nyquist sampling ratio rate.

Half of this value  $F_{max}$  is sometimes called as Nyquist frequency

Ques 3. List different sampling techniques,

Ans 3. There are 3 types of sampling techniques -

- 1) Ideal sampling or Instantaneous sampling or Impulse sampling
- 2) Natural sampling
- 3) Float top sampling.

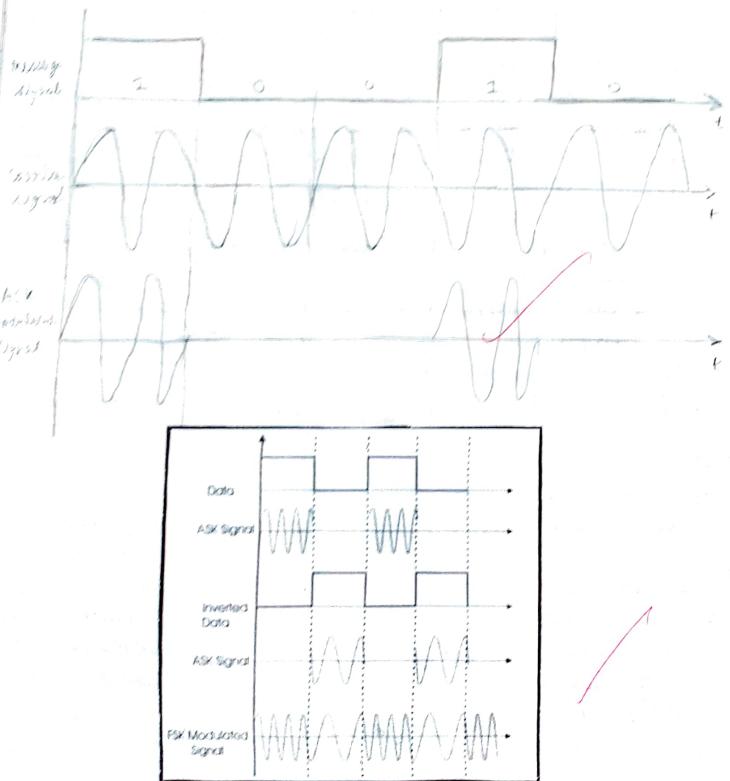
Ques 4. How aliasing is removed?

Ans 4. Aliasing is removed by sample filtering out of all high frequencies components before sampling.

Ques 5. What is under sampling?

Ans 5. When sampling rate is lower than or equal to Nyquist rate, a condition defined as under sampling. It is impossible to rebuild the original signal according to Sampling theorem.

### Amplitude shift keying modulation waveform



## Experiment - 2

Aim : Study of amplitude shift keying modulation and demodulation

Equipments Required : ST 2156 and ST 2157 with power supply card, CRO / Oscilloscope with connecting probe, connecting cords

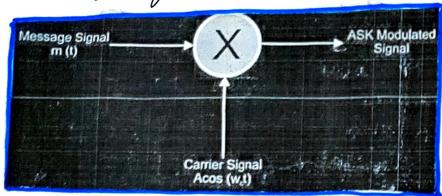
Theory :

ASK technique is the simplest method of modulating a carrier with a stream data stream to change the amplitude of the carrier wave every time the data changes. This modulation technique is known as Amplitude Shift Keying.

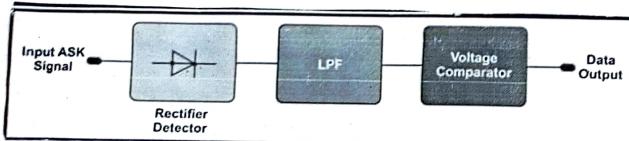
The simplest way of achieving ASK is by switching 'ON' the carrier whenever data bit '1' and switching it 'OFF' whenever the data bit is '0' i.e. the transmitter output the carrier for a '1' and totally suppresses the carrier for a '0'. This technique is also known as ON-OFF keying.  
 Data = 1 carrier transmitted  
 Data = 0 carrier suppressed

The ASK waveform is generated by a balanced modulator circuit, also known as a linear multiplier. The device multiplies the instantaneous signal at its two inputs. The O/P voltage being product of two I/P voltages at any instance of time. One of the I/P is AC coupled 'carrier' wave & the

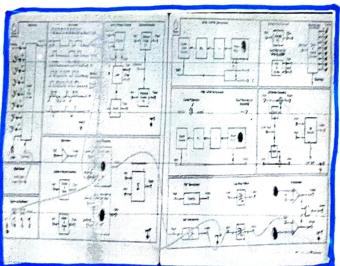
### Amplitude Shift keying modulator



### Amplitude Shift keying demodulator



### CIRCUIT DIAGRAM

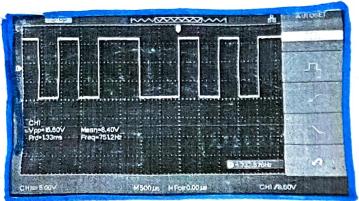


frequency. Generally, the carrier wave is a sinusoidal signal since any other waveform would increase bandwidth without providing any advantages. In other I/P which is the information signal to be transmitted, is DC coupled. It is known as modulating signal. The data stream applied is unipolar i.e. 0 volt for logic '0' & +5 volt for logic '1'. The O/P of balanced modulator is a sine wave, unchanged in phase when a data bit '1' is applied to it and is 0 when data bit '0' is applied. In ASK modulation result in great simplicity at the receiver, the method to demodulate the ASK waveform is to rectify it, pass it through the filter & 'shape up' the resulting waveform. The O/P is the original data stream.

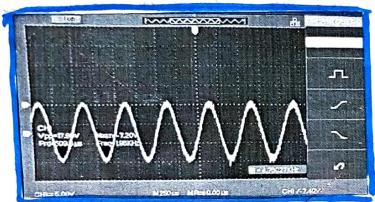
#### Procedure :

1. Power supplies of Scientech 2156 and 2157 were connected and connections were made according to given figure.
2. "Clock in" of CH1 and "Data in" of CH2 were connected and waveforms were observed.
3. Oscilloscope CH1 was connected to "NRZ(L)" and CH2 to "Output" of modulator circuit of Scientech 2156 and waveforms were observed.
4. Gain potentiometer of modulator circuit on Scientech 2156 was varied to adjust amplitude of ASK waveform.
5. On Scientech 2156 was connected to oscilloscope CH1 to "NRZ(L)" and CH2 to "Output" of comparator on Scientech 2157 and waveforms were observed.

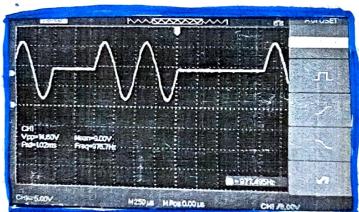
## Waveforms of ASK modulation



carrier signal



ASK signal



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### Observations :

1. The output at "Data In" is repeating sequence of bits generated by Data Source.
2. The output at Modulator circuit (1) is the ASK waveform which contains carrier transmitted for Data '1' and carrier suppressed data '0'.
3. The output at Comparator on Scientech 2157 is the same as "Data In" on Scientech 2156.

### Conclusion :

1. Amplitude Shift keying is fairly simple to implement in practice, but it is less efficient, because the noise inherent in the transmitter channel can deteriorate the signal so much that the amplitude changes in the modulated carrier wave due to noise addition may lead to the incorrect decoding the receiver.
2. The technique is not widely used in practice. Application wise, it is however used in diverse areas and old as emergency radio transmission and fiber-optic communications.

### VIVA VOICE QUESTIONS

- Ques 1. What are the types of Digital Modulation techniques?   
 Digital Modulation techniques is classified in

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coherent or non-coherent techniques depending on whether the receiver is equipped with a phase recovery circuit or not.

~~Ques 2. Define coherent Digital Modulation Techniques?~~

~~Ans 2. Coherent Digital Modulation techniques employs coherent detection technique in which local carrier generated at the receiver is phase locked with the carrier which is used in transmitter section. It is a synchronous detection method~~

~~Ques 3. state maximum the bandwidth requirement in ASK system?~~

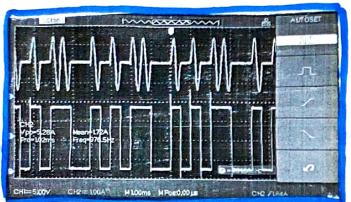
~~Ans 3. Maximum Bandwidth as  $2f_b$  Hz in ASK System.~~

~~Ques 4. what are advantages of digital communication?~~

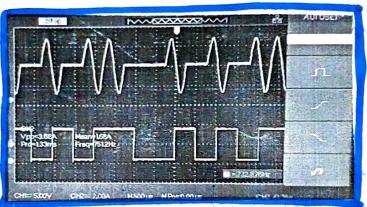
~~Ans 4. Reliability, noise effect is very less, power consumption is very less, various Digital IC's are available so circuit is not complex, cheap error detection and correction is also possible~~

~~Ques 5. How to convert an analog to digital signal?~~

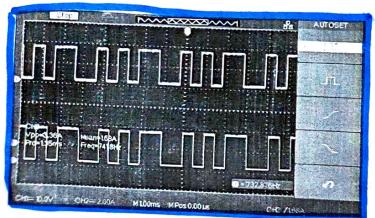
~~Ans 5. Anti aliasing filter, sampler, Quantizer, encoder~~



data vs ASK  
signal  
(↑ frequency)



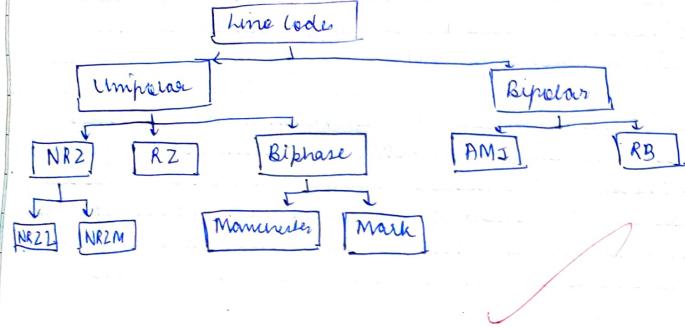
ASK signal  
vs  
Output data  
(zoomed in)



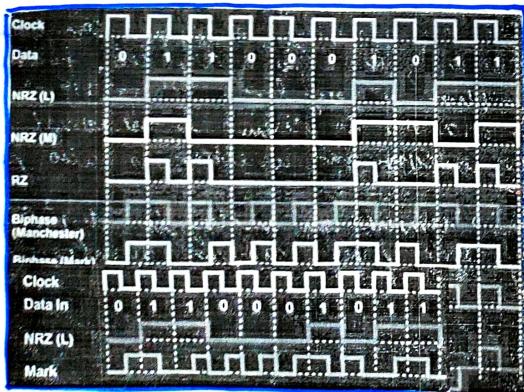
[ data at the  
input  
vs  
data at output ]

{ modulated signal obtained  
and signal to be modulated }

Classification of line codes :



Various data formatting techniques :



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## Experiment - 3

Aim : To perform different line codes - NRZ-I, NRZ-M, RZ, Biphase (Manchester), Biphase (Mark) and AMI

Equipment Required : ST 2156 and ST 2157 with power supply card  
CRO with connecting probe  
Connecting cables

### Theory :

Line coding and decoding consists of representing the digital signal to be transported by an amplitude and time discrete signal that is optimally tuned for specific properties of physical channel. The waveform pattern of link is called line encoding or voltage and current to represent 1s and 0s of digital signal on a transmission line called line encoding. It is commonly used in computer communication networks over short distances.

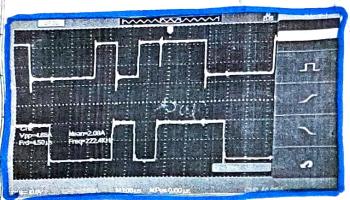
Various advantages and disadvantages - Minimize transmitter hardware

- Facilitates synchronization
- Ease error detection & correction
- Minimize spectral content
- Eliminates a dc component

The manchester code is quite popular. It is known as

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## Different data formatting Techniques

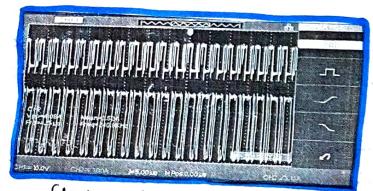


Input Clock  
vs

Clock observed

$$\text{data} = 1 \ 0 \ 1 \ 0 \ 0 \ 1 \ 1$$

NRZ - L      (Clock vs NRZ-L waveform observed)



Clock vs NRZ - L



NRZ - L

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a self clocking because there is always transition during bit intervals. Consequently, long strings of zeros or ones do not cause clocking problems.

### Different Data formatting techniques :

#### 1. Non return to zero level (NRZ - L) :

Representation : +5V for data bit 1 and 0V for 0

Bandwidth : Low Bandwidth

DC level : High DC component

Timing Information : No info (for long streams of 1s & 0s)

#### 2. Non-return to zero-level (NRZ - M) :

Representation : Level transition for bit 1 and unchanged for 0

Bandwidth : Low bandwidth

DC level : High DC component

Timing Information : No timing info (for long stream of 0s)

#### 3. Return to Zero (RZ) :

Representation : 0V for bit 0 for bit 1, for half bit duration (+5V) and rest as 0V

Bandwidth : Twice as required for NRZ

DC level : High DC component

Timing Information : No timing info (for long streaming 0s)

#### 4. Biphase (Manchester) :

Representation : +5V for first half bit and 0V for second half bit, 0V for first half and +5V for rest.

Bandwidth : Twice as required for NRZ

DC level : No DC component

Timing info : Good clock recovery

#### 5. Biphase (Mark.)

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Representation : for any bit first half +5V and rest 0 and in next half -5V and next bit. Bit 0 pattern remains same. Bit 1 Phase Reversal.

Bandwidth : Twice as required for NRZ.

DC level : No DC component.

Timing Info : Ground Clock Recovery.

#### 6. Return to Bias (RB) :

Representation : During the first half of a period, it is +5V for bit 1 and then -5V for bit 0 and during the second half of bit time, it returns to the bias level.

Bandwidth : Twice as required for NRZ.

DC Level : depends on density of 1's and 0's.

Timing Info : Ground clock recovery (Self clocking).

#### 7. Alternate Mark Inversion (AMI)

Representation : Like RB, AMI returns to bias level during second half and during first half period, it can be +ve or -ve for a bit 1 either and changes alternatively.

Bandwidth : Twice as required for NRZ.

DC level : No DC component.

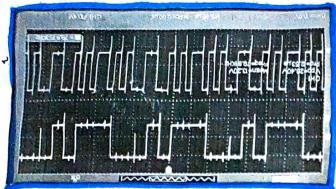
Timing info : No info (for long sequence of 0's)

#### Procedure :

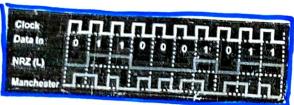
1. ~~Scientech 2156 was connected to power supply and the connections were made.~~
2. Power was switched on and various observations were made by visualizing waveforms at various pins.

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## Biphase Manchester

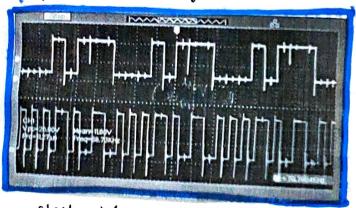


Clock vs Biphase Manchester



Biphase Manchester

## Biphase Mark



Clock vs Biphase Mark

2. Oscilloscope is connected to ~~clock in, NRZ (L), NRZ(M)~~, RZ, Biphase (manchester), Biphase (mark), RB → AMI and waveforms were observed.

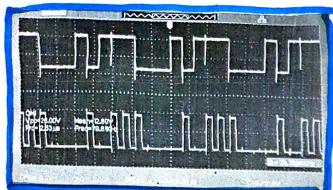
## Observations:

1. The output at 'Data in' is a repeating sequence of bits generated by parallel to serial converter.
2. The NRZ(L) is same as 'Data in' but it is one bit shifted.
3. Verified all formatting techniques according to example patterns given on the Scientech 215G board.

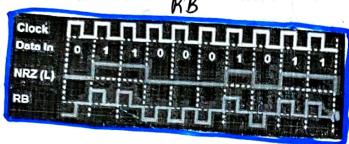
## Conclusion:

1. NRZ(L) waveform simply goes low for one bit time to represent data '0' and high for one bit time to represent a data '1'.
2. In NRZ (M) line coding, the present level is related to the previous level when logic '1' is to be transmitted change in level occurs and for logic '0' the level remains unchanged.
3. In RZ line codes, the maximum signal frequency is RZ signal occurring when always  $\frac{1}{2}$  is transmitted change in level. It's equivalent to sending 2 logic levels in each clock period. Thus bandwidth required is twice as required for the NRZ waveform.
4. The biphase manchester codes always contain atleast one transition per bit-line irrespective of data being transmitted. Hence, the maximum frequency of the biphase system is equal to data clock rate when a

RB (Return to Bias)

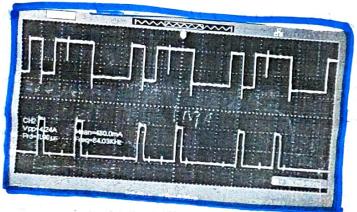


clock vs RB waveform

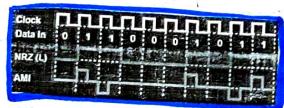


RB

Alternate Mark Inversion (AMI)



clock vs AMI waveform



AMI

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Mean of consecutive data '1' & '0' is transmitted.

5. Biphase Mark has sequence transmitted level constant in case of 0 bits for 1 sequence transmitted levels hence phase reversal from one I to another.
6. The Biphase mark code being nearly digital to Biphase (e. Manakov) requires band bandwidth as double to NRZ (L).
7. The maximum signal frequency in RB code is equal to data frequency; the bandwidth required is same as that for NRZ, biphase codes and double that for NRZ codes.
8. The maximum transmission rate for AMI can occur only during a stream of all '1's thus bandwidth required is twice as for NRZ codes.

Result :

Different line coding techniques were observed and verified.

### VIVA VOCE QUESTIONS

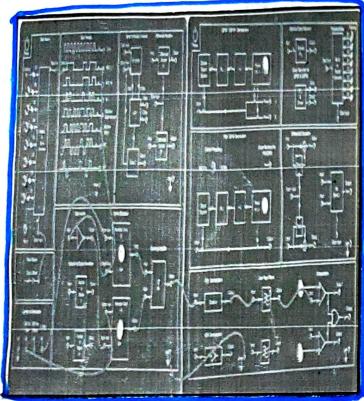
Ques 1. What is Equalization?

Ans. Equalization is the reversal of distortions occurred by a signal transmitted through a channel.

They are used to render frequency response flat. Varying intervals.

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## Circuit Diagram



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Ques 2. What is Baseband binary data transmission system?

~~An~~ Baseband transmission system transfer digital signals in the form of a train of pulses. Signals transferred in baseband signal are not moved from their original frequency bands as no sinusoidal carrier modulation is used.

Ans 3. What do you mean by Eye diagram?

~~An~~ It is an oscilloscope display in which digital signal from receiver is relatively sampled and applied to the vertical inputs, while the data rate is used to trigger horizontal sweep.

Ques 4. What are other names of Bipolar and Manchester coding?

~~An~~ Bipolar → AMI (Alternate pulse inversion)

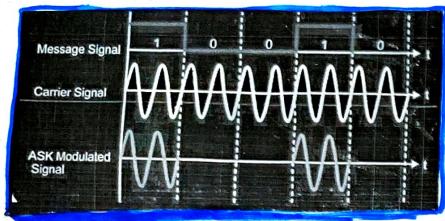
~~An~~ Manchester → phase Encoding

Ques 5. Name some coding techniques?

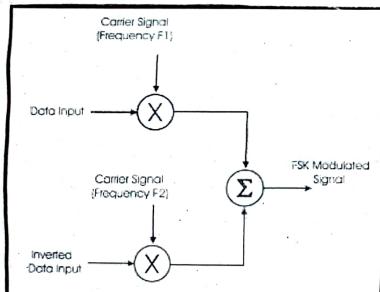
~~An~~ pulse code demodulation, differential pulse code modulation, delta modulation, adaptive delta modulation

Expt. No. 4

Generation of FSK waveform from the sum of two ASK waveforms



### Frequency Shift keying modulator



## Experiment - 4

Aim: To perform frequency shift keying modulation and demodulation

Equipments required:

- OT 2156 and ST 2157 with power cords
- CRO with connecting probes
- Connecting cords

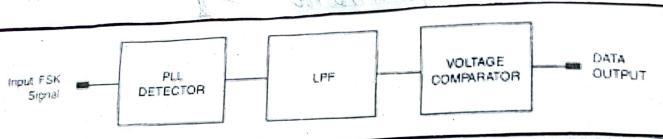
Theory:

FSK waves are apparent it's sum of two ASK waves. Thus the functional block of FSK requires 2 ASK modulator each and have different carrier frequencies but digital data is inverted in one of the modulator. These 2 ASK modulated signal are applied to the summing amplifier to get FSK modulated signal.

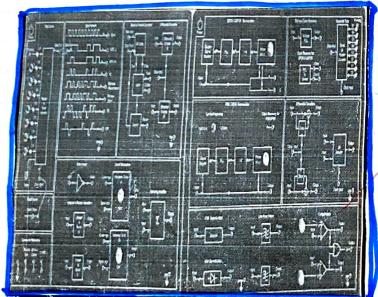
The modulation of FSK can be carried out by phase lock loop as known to it locks I/P frequency by generating corresponding voltage to be fed at voltage controlled oscillator, if any frequency deviation is at its input encountered thus PLL detector is followed by frequency changes and generates proportional o/p voltage it contains carrier waves/ components. Therefore, signal is passed through low pass filter to remove them. The resulting wave is used for data processing.

Teacher's Signature:

## Frequency Shift Keying demodulator



## Circuit Diagram



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

The amplitude level may be very low due to channel attenuations. The signal is shaped VP by feeding it to the voltage comparator. The functional block diagram of FSK demodulator is displayed.

### Procedure :

1. Scientech 2156 and 2157 were connected to the power supply and connections were made.
2. The leads were connected as displayed in the circuit diagram and power supply was turned 'ON'.
3. On Scientech 2156 oscilloscope CH1 was connected to 'Clock In' and CH2 to 'Data In' and waves were observed.
4. On Scientech 2156, oscilloscope CH1 was connected to 'NRZ(L)' and CH2 to O/P of the summer Amplifier on Scientech 2156 and waveforms were observed.
5. modulators were adjusted to adjust amplitude of FSK from summer Amplifier O/P.
6. On Scientech 2156 oscilloscope CH1 was connected to 'NRZ(L)' and CH2 to Output of comparator and waveforms were observed.

### Observation :

The output at summer amplifier is the FSK waveform, data bit '0' or FSK signal was at lower frequency and data bit '1' the FSK signal was at high frequency.

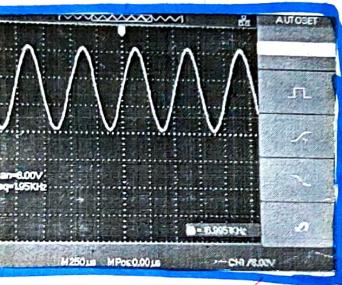
Teacher's Signature : \_\_\_\_\_

& demodulation

data

modulation and )

on



FSK signal

The output comparator on scientex is same as the data inputted.

Conclusion:

The amplitude change in FSK waveform does not matter, therefore FSK modulation technique is very reliable even in noisy and fading channel.

Result :

FSK modulation and demodulation was observed and understood.

### VIVA VOCE QUESTIONS

Ques 1. What is the bandwidth of BFSK ?

Ans 1  $\pi f_b$

$$B_o = \pi Df + 2B$$

$B = R_b$  (Bandwidth of rectangular pulse)

$$B_o = \pi(Df + R_b)$$

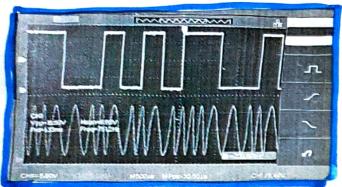
If RC shaping,

$$B_o = \pi Df + (1+\alpha) R_b$$

Ques 2. What is meant by orthogonal BFSK ?

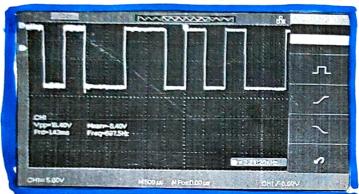
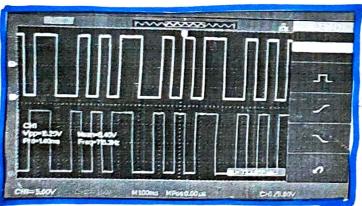
Ans 2 Orthogonality is the property which

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Data vs FSK signal

Data output at  
comparator  
(demodulated signal)  
obtained

Data vs  
data obtained (Data Output)

Results in 0 dot product of 2 vectors.  
Sti extension of concept of 1<sup>r</sup> vector to any  
space dimension.  
So here we want to look the balanced  
orthogonal approach onto carrier ~~no~~ frequency  
~~lock in amplifier~~

Ques 3. What kind of receiver is used in BFSK  
and FSK?

In FSK - conventional frequency demodulator  
for BFSK detection we use synchronous  
detector

Ques 4. What is advantage of FSK over ASK?

- ~~→ It has a low probability~~
- ~~→ Easy to implement~~
- ~~→ High data rate~~
- ~~→ better noise immunity than ASK~~
- ~~→ probability of error-free reception is high~~
- ~~→ Easy to decode~~
- ~~→ high SNR~~
- ~~→ operate in virtually any wires~~
- ~~→ easier implementation of transmitter and receiver for low data rate applications~~
- ~~→ Good sensitivity~~
- ~~→ low noise~~
- ~~→ Power requirement is constant.~~

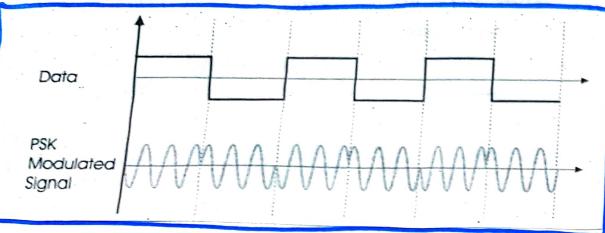
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Ques 5. What is Binary Phase shift keying (BPSK) ?  
Ans For each one bit of binary data (0 and 1) carrier phase will be changed (two different shifts; 0, 180)

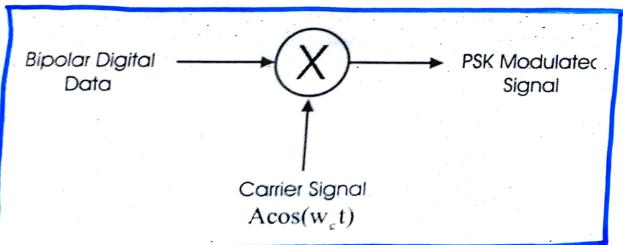
Ques 6. Compare ASK, PSK and FSK ?

Ans Bandwidth  $ASK < PSK < FSK$   
Power  $ASK < PSK = FSK$

Probability of error  $ASK > PSK > FSK$   
~~SNR~~  $ASK < PSK < FSK$

Diagrams and Waveforms :

Phase Shift Keying Waveform.



Phase Shift Keying Modulator

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## Experiment - 5

Aim : To perform Phase shift keying modulation and demodulation

Equipment Required : ST2156 and GT21057 with power supply  
CRO with connecting probe  
connecting cords

Theory :

Phase shift keying involves the phase change of the carrier wave between  $0^\circ$  and  $180^\circ$  in accordance with the data levels to be transmitted. Phase shift keying is also known as phase reversal keying (PRK). The PSK waveform for a given data is as shown in figure.

For Binary PSK

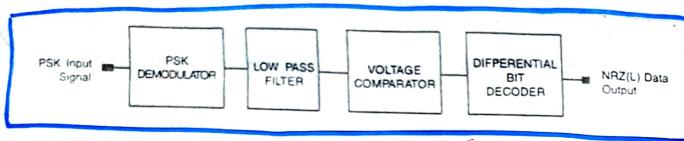
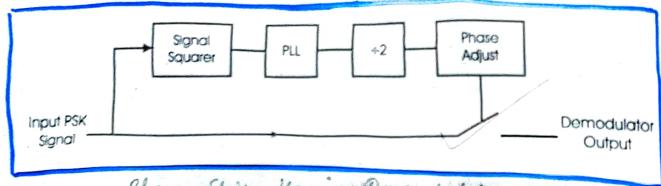
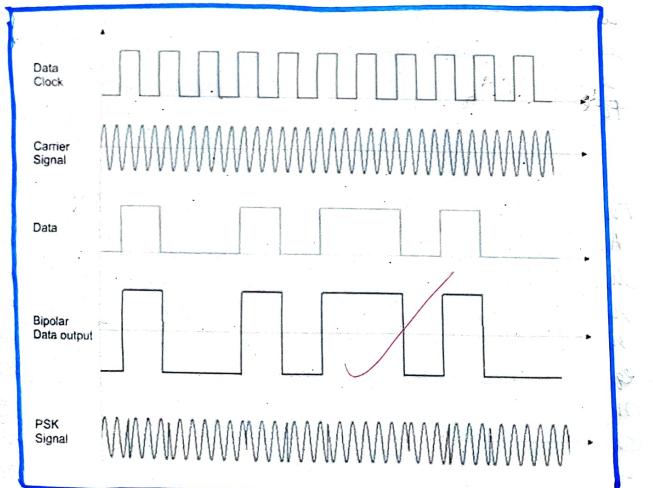
$$s_0(t) = \text{Acos}(wt) \quad \text{represents binary } '0'$$

$$s_1(t) = \text{Acos}(wt + \pi) \quad \text{represents binary } '1'$$

Functionally, the PSK modulator is very similar to the ASK modulator. Both uses balanced modulator to multiply the carrier with the modulating signal. But in contrast to ASK technique, the digital signal applied to the modulation input for PSK generation is bipolar i.e. have equal positive and negative voltage levels. When the modulating input signal is positive the output of modulator is a sine wave in phase with the carrier input. Whereas as for the negative voltage levels, the output

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*Phase Shift Keying Receiver System**Waveforms of PSK Modulation*

of modulator, is a sine wave which is shifted one quarter wave say  $180^\circ$  from the carrier input. This happens because the carrier input is now multiplied by negative constant level. The functional block representation of the PSK modulator.

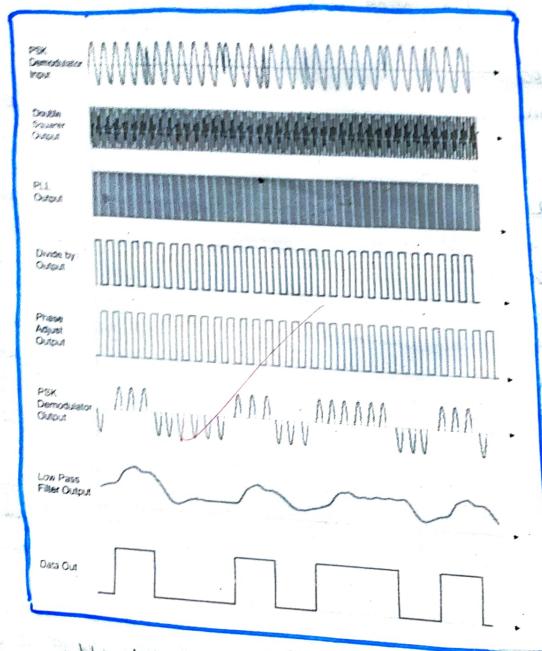
*Procedure:*

1. Connected the power supplies of scientech 2156 and Scientech 2157 but do not turn on the power supplies until connections are made for this experiment.
2. Made the the connections as per the circuit diagram
3. The circuit was turned ON, and scientech 2156 oscilloscope was connected to CH1 to "Clock In", and CH2 to "Data In" and observed the waveforms.
4. Scientech 2156 was connected CH2 oscilloscope, CH2 (CL) and CH2 to "Output" of modulator circuit (1) on Scientech 2156 and observe the waveform.
5. Adjust the "Gain" potentiometer of modulator circuit (1) on Scientech 2156 to adjust the amplitude of PSK waveform at output of modulator circuit on Scientech 2156.
6. Various comparisons and waveforms were obtained and verified.

*Observations:*

1. The output at "Data In" is repeating sequence of bits generated by Data Source.
2. The "Output" of modulator circuit (1) is Phase Shift Keying modulated signal.

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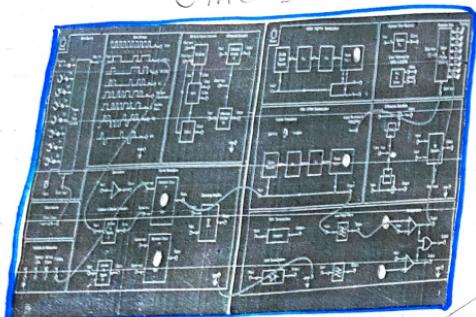
Waveforms of PSK Demodulation

3. The output of double squaring circuit is sinusoidal signal (carrier signal) but frequency is four times higher than that of carrier used for modulation.
4. The output of Phase Lock Loop (PLL) is clock signal of same frequency as that of the output of double squaring circuit and output of divide by two ( $1/2$ ) is clock signal of frequency two times less than the output of PLL signal.
5. The output of PSK demodulator is a signal having group of positive half cycles and group of negative half cycles of the ~~the~~ carrier signal.
6. A low pass filter removes high frequency component from demodulated PSK signal and it makes the signal smooth.
7. The variation in reference voltage potentiometer affects the data, to recover data correctly potentiometer adjustment is necessary.
8. The phase adjust potentiometer on Scientech 2157 matches the phase of regenerated clock and carrier with input clock and carrier signal.

#### Conclusion :

Now we can observe that the final data stream can be either the original data stream or its inverse, this is because the sine wave is symmetrical  $\rightarrow$  the receiver has no way of detecting whether the incoming phase of the signal is  $0^\circ$  or  $180^\circ$ . This phase ambiguity creates possibilities for the receiver output i.e.

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CIRCUIT DIAGRAM→ OBSERVED WAVEFORMS

I/p data clock vs FSK demodulated O/P



final data stream can be either the original data stream or its inverse

result = PSK modulation and demodulation was observed and verified.

VIVA QUESTIONS

Ques 1. What is the bandwidth of BPSK?  
Ans Binary Phase Shift Keying is a digital modulation scheme that conveys data by changing or modulating two different phases of a reference signal (the carrier wave). The constellation points chosen are usually positioned with uniform angular space.

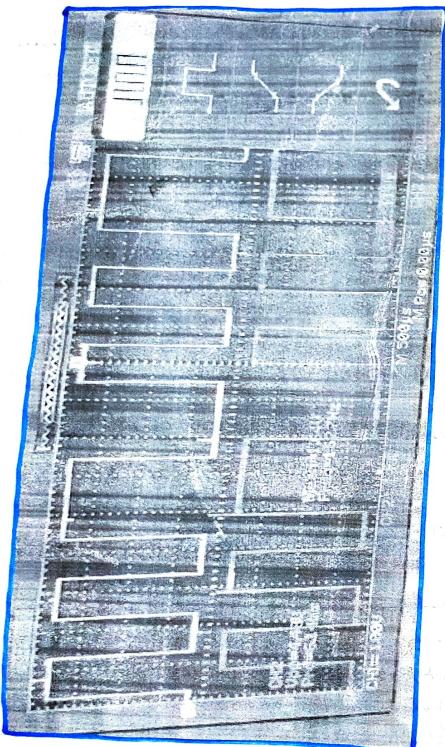
$$\begin{aligned} \text{BW} &= f_c + f_b - f_c - f_b \\ &= \text{highest frequency} - \text{lowest frequency} \\ &= f_c + f_b - (f_c + f_b) \\ \text{BW} &= 2f_b = 2f_b \end{aligned}$$

Ques 2. What is the role of bit synchronizer in BPSK receiver?

Ans Highly integrated ultra-low-power binary phase shift keying receivers are used for short-range wireless communications. These receivers consist of a power divider, two injection-locked RC oscillators with limiting buffers, and an XOR output stage.

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OBSERVED WAVEFORM

I/P data vs Output Data obtained after PSK demodulation.

Ques 3. What is the disadvantages of BPSK receiver?

- Ans 1. One bit is carried by one single analog carrier. Hence data rate in bits per second is same as the symbol rate. This half compared to QPSK modulation technique and many times less compare other higher modulation techniques such as 16QAM, 64QAM, etc.
2. Due to above reason, BPSK is not bandwidth efficient technique compared to other modulation types.

Ques 4. What type of receiver is used for BPSK detection?

Ans Synchronous detectors are used.

To detect the original binary sequence of 0s and 1s, the BPSK signal or channel output is applied to a receiver that consists of:

- Phase lock modulator
- Low pass filter
- Sampler
- Decision making device

Ques 5. What is the basic difference between coherent and non-coherent BPSK?

Ans In Coherent detection the local carrier generated at the receiver is phase locked with the carrier at the transmitter. Hence it is also called synchronous detection. In non-coherent detection the local carrier generated at the receiver is may not be phase.