

# Subject Name Solutions

4331104 – Winter 2022

Semester 1 Study Material

*Detailed Solutions and Explanations*

## Question 1(a) [3 marks]

What is modulation? What is the need of it?

### Solution

Modulation is the process of varying one or more properties (amplitude, frequency, or phase) of a high-frequency carrier signal with a modulating signal containing information.

**Need for modulation:**

- **Antenna size reduction:** Makes practical antenna size possible ( $\lambda = c/f$ )
- **Multiplexing:** Allows multiple signals to share the medium
- **Noise reduction:** Improves SNR by shifting to higher frequency bands
- **Range extension:** Increases transmission distance

### Mnemonic

“AMEN” - Antenna size, Multiplexing, Eliminate noise, New range

## Question 1(b) [4 marks]

Derive voltage equation for Amplitude modulation.

### Solution

For AM, the carrier signal is modulated by the message signal.

**Mathematical derivation:**

- **Carrier signal:**  $e_c(t) = A_c \cos(2\pi f_c t)$
- **Message signal:**  $e_m(t) = A_m \cos(2\pi f_m t)$
- **Instantaneous amplitude:**  $A_i = A_c + e_m(t)$
- **AM signal:**  $e_{AM}(t) = A_i \cos(2\pi f_c t)$
- **Substituting:**  $e_{AM}(t) = [A_c + A_m \cos(2\pi f_m t)] \cos(2\pi f_c t)$
- **Expanding:**  $e_{AM}(t) = A_c \cos(2\pi f_c t) + A_m \cos(2\pi f_m t) \cos(2\pi f_c t)$
- **Final equation:**  $e_{AM}(t) = A_c \cos(2\pi f_c t) + \frac{A_m}{2} \cos(2\pi(f_c + f_m)t) + \frac{A_m}{2} \cos(2\pi(f_c - f_m)t)$

### Mnemonic

“CAT” - Carrier, Addition, Three components (carrier + 2 sidebands)

## Question 1(c) [7 marks]

Classify Noise signal and explain flicker noise, shot noise and thermal noise.

### Solution

**Noise classification:**

Type	Sources	Characteristics
<b>External Noise</b>	Atmospheric, Space, Industrial, Man-made	Originates outside communication system
<b>Internal Noise</b>	Thermal, Shot, Transit-time, Flicker	Originates inside components

**Types of Internal Noise:**

- **Flicker Noise:**
  - Occurs at low frequencies (below 1 kHz)
  - Inversely proportional to frequency (1/f noise)
  - Common in semiconductor devices and carbon resistors
- **Shot Noise:**
  - Caused by random fluctuations of current carriers
  - White noise with constant power density
  - Occurs in active devices like diodes and transistors
- **Thermal Noise:**
  - Due to random motion of electrons in a conductor
  - Directly proportional to temperature and bandwidth
  - Present in all passive components
  - Also called Johnson noise or white noise

**Mnemonic**

“FAST” - Flicker (low frequency), Active (shot), Semiconductor (flicker), Temperature (thermal)

**Question 1(c) OR [7 marks]**

Write application of different band of EM wave spectrum.

**Solution****EM Spectrum Applications:**

Frequency Band	Frequency Range	Applications
<b>ELF</b> (Extremely Low Frequency)	3Hz - 30Hz	Submarine communication
<b>VLF</b> (Very Low Frequency)	3kHz - 30kHz	Navigation, time signals
<b>LF</b> (Low Frequency)	30kHz - 300kHz	AM radio, navigation
<b>MF</b> (Medium Frequency)	300kHz - 3MHz	AM broadcasting, maritime
<b>HF</b> (High Frequency)	3MHz - 30MHz	Shortwave radio, amateur radio
<b>VHF</b> (Very High Frequency)	30MHz - 300MHz	FM radio, TV broadcasting, air traffic control
<b>UHF</b> (Ultra High Frequency)	300MHz - 3GHz	TV broadcasting, mobile phones, WiFi, Bluetooth
<b>SHF</b> (Super High Frequency)	3GHz - 30GHz	Satellite communication, radar, WiFi
<b>EHF</b> (Extremely High Frequency)	30GHz - 300GHz	Radio astronomy, 5G, millimeter-wave radar
<b>Infrared</b>	300GHz - 400THz	Remote controls, thermal imaging, fiber optics
<b>Visible Light</b>	400THz - 800THz	Fiber optics, LiFi, photography
<b>Ultraviolet</b>	800THz - 30PHz	Sterilization, fluorescence, security
<b>X-rays</b>	30PHz - 30EHZ	Medical imaging, security screening
<b>Gamma rays</b>	>30EHZ	Medical treatments, nuclear detection

**Mnemonic**

“Every Very Lovely Monkey Has Visited Uncle Sam’s House Easily In Visible Upper Xtra Gamma” (first letter of each band)

**Question 2(a) [3 marks]**

State advantages of SSB over DSB.

### Solution

#### Advantages of SSB over DSB:

Advantage	Description
<b>Bandwidth Efficiency</b>	Uses half the bandwidth (only one sideband)
<b>Power Efficiency</b>	Requires less transmitter power (83.33% power saving)
<b>Reduced Fading</b>	Less susceptible to selective fading
<b>Less Distortion</b>	Reduced intermodulation distortion
<b>Simplified Receiver</b>	Simpler circuit design possible

### Mnemonic

“BPFDS” - Bandwidth, Power, Fading, Distortion, Simple

## Question 2(b) [4 marks]

Explain generation of FM using Phase lock loop technique.

### Solution

#### FM Generation using PLL:

A Phase-Locked Loop (PLL) generates FM signals by applying the modulating signal to the VCO control input.

#### PLL FM Modulator:

#### Mermaid Diagram (Code)

```
{Shaded}
{Highlighting}[]
graph LR
    A[Modulating Signal] --> B[Summing Circuit]
    E[Reference Oscillator] --> F[Phase Detector]
    F --> G[Low Pass Filter]
    G --> B
    B --> H[VCO]
    H --> I[FM Output]
    I --> J[Feedback]
    J --> F
{Highlighting}
{Shaded}
```

#### Operation:

- **Reference Oscillator:** Provides stable reference frequency
- **Phase Detector:** Compares reference and feedback signals
- **Low Pass Filter:** Removes high-frequency components
- **VCO:** Generates output frequency that varies with control voltage
- **Modulating Signal:** Added to control voltage to produce FM output

### Mnemonic

“PROVE” - Phase detector, Reference oscillator, Output VCO, Voltage controlled

## Question 2(c) [7 marks]

Derive the equation for total power in AM, calculate percentage of power savings in DSB and SSB.

### Solution

#### Power in AM:

The AM wave equation:  $e_{AM}(t) = A_c[1 + m \cos(2\pi f_m t)] \cos(2\pi f_c t)$

**Power derivation:**

- **Total power:**  $P_T = P_c \left(1 + \frac{m^2}{2}\right)$
- Where  $P_c = \frac{A_c^2}{2R}$  (carrier power) and  $m$  is modulation index

**Power distribution:**

- **Carrier power:**  $P_c = \frac{A_c^2}{2R}$
- **Total sideband power:**  $P_{SB} = \frac{m^2 P_c}{2}$
- **Each sideband:**  $P_{LSB} = P_{USB} = \frac{m^2 P_c}{4}$

**Power savings:**

- **In DSB-SC:** No carrier power, so savings =  $\frac{P_c}{P_T} \times 100\% = \frac{1}{1 + \frac{m^2}{2}} \times 100\%$ 
  - For  $m=1$ , savings = 66.67%
- **In SSB:** No carrier and one sideband, so savings =  $\frac{P_c + P_{SB}/2}{P_T} \times 100\%$ 
  - For  $m=1$ , savings = 83.33%

**Mnemonic**

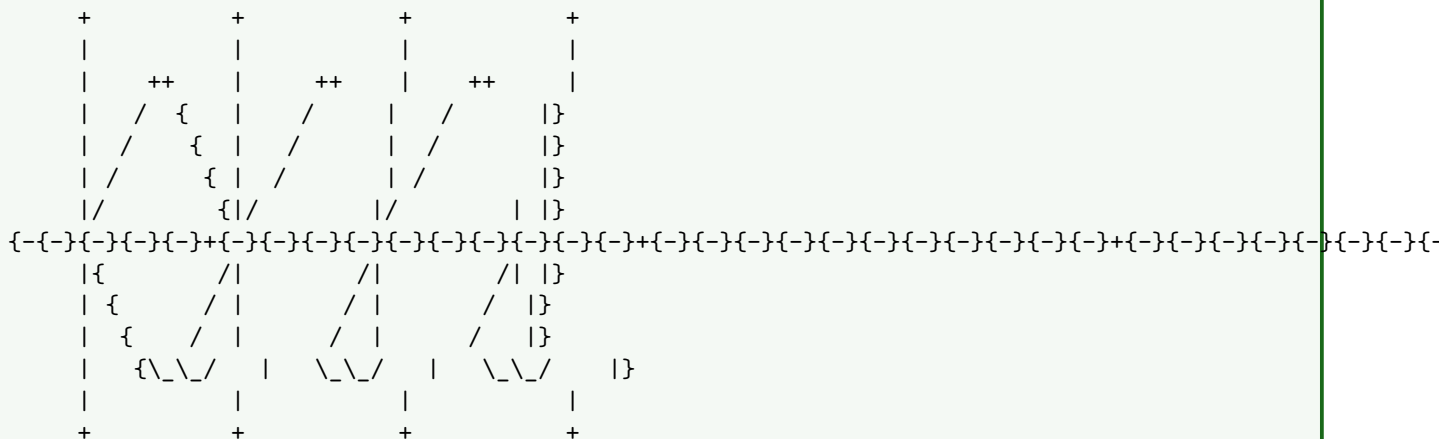
“CEPTS” - Carrier Eliminated Provides Tremendous Savings

**Question 2(a) OR [3 marks]**

Draw and explain Time domain and Frequency domain display of AM wave.

**Solution****Time and Frequency Domain of AM:****Diagram:**

Time Domain:



Frequency Domain:

**Time Domain:**

- Shows amplitude variation of carrier with time
- Envelope follows modulating signal

- Upper and lower envelopes = carrier peak  $\times(1)$

#### Frequency Domain:

- Shows frequency components and their amplitudes
- Carrier at frequency  $f_c$  with amplitude  $A_c$
- Two sidebands at  $f_c$  with amplitude  $mA_c/2$
- Bandwidth =  $2f_m$  (twice the modulating frequency)

#### Mnemonic

“EBS” - Envelope in time, Bandwidth in frequency, Sidebands symmetric

### Question 2(b) OR [4 marks]

Explain pre-emphasis & de-emphasis circuit.

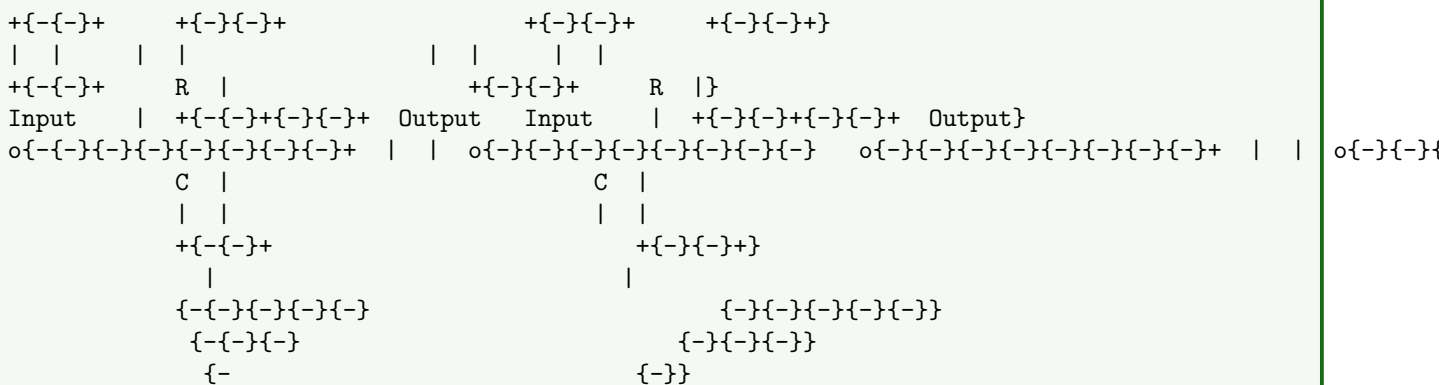
#### Solution

##### Pre-emphasis and De-emphasis:

##### Circuit Diagrams:

Pre-emphasis:

De-emphasis:



##### Purpose:

- **Pre-emphasis:** Boosts high-frequency components at transmitter
- **De-emphasis:** Attenuates high-frequency components at receiver

##### Operation:

- **Pre-emphasis:** High-pass RC circuit ( $R$  series,  $C$  parallel)
- **De-emphasis:** Low-pass RC circuit ( $R$  parallel,  $C$  series)
- Time constants are identical:  
 $= RC = 75 \text{ s}$  (standard)

##### Benefits:

- Improves SNR for higher frequencies in FM
- Compensates for higher noise power at high frequencies
- Restores original frequency response at receiver

#### Mnemonic

“BETH” - Boost (pre-emphasis), Emphasizes Treble, Helps SNR

### Question 2(c) OR [7 marks]

Compare AM, FM and PM.

#### Solution

##### Comparison of AM, FM and PM:

Parameter	AM	FM	PM
<b>Definition</b>	Amplitude varies with message signal	Frequency varies with message signal	Phase varies with message signal
<b>Mathematical expression</b>	$A_c[1 + m \cos(mt)] \cos(ct)$	$A_c \cos[ct + mf \sin(mt)]$	$A_c \cos[ct + mp \cos(mt)]$
<b>Bandwidth</b>	2fm (narrow)	$2(\Delta f + f_m)$ (wide)	$2(m_p + 1)f_m$ (wide)
<b>Power efficiency</b>	Low (carrier contains no info)	High (constant amplitude)	High (constant amplitude)
<b>Noise immunity</b>	Poor	Excellent	Excellent
<b>Circuit complexity</b>	Simple	Complex	Complex
<b>Applications</b>	AM broadcasting, aircraft communication	FM broadcasting, TV sound, mobile radio	Satellite communication, telemetry
<b>Modulation index</b>	$m = A_m/A_c$ (0 to 1)	$m_f = \Delta f/f_m$ (no limit)	$m_p = \Delta / f_m$ (no limit)

#### Mnemonic

“BANCP-MAP” - Bandwidth, Amplitude, Noise, Complexity, Power, Modulation, Applications, Parameters

### Question 3(a) [3 marks]

Define any FOUR characteristics of radio receiver.

#### Solution

##### Radio Receiver Characteristics:

Characteristic	Definition
<b>Sensitivity</b>	Minimum signal strength required for acceptable output
<b>Selectivity</b>	Ability to separate desired signal from adjacent signals
<b>Fidelity</b>	Accuracy in reproducing the original signal without distortion
<b>Image rejection</b>	Ability to reject image frequency interference
<b>Signal-to-noise ratio</b>	Ratio of desired signal to unwanted noise
<b>Stability</b>	Ability to maintain tuned frequency without drift

#### Mnemonic

“SFIS-SS” - Sensitivity, Fidelity, Image rejection, Selectivity, SNR, Stability

### Question 3(b) [4 marks]

Draw the block diagram of FM receiver. What is the use of Limiter in FM receiver.

#### Solution

##### FM Receiver Block Diagram:

##### Mermaid Diagram (Code)

```

{Shaded}
{Highlighting}[]
graph LR
    A[Antenna] --> B[RF Amplifier]
    B --> C[Mixer]
    D[Local Oscillator] --> C

```

```

    C {--}{ } E[IF Amplifier]}
    E {--}{ } F[Limiter]}
    F {--}{ } G[FM Detector]}
    G {--}{ } H[Audio Amplifier]}
    H {--}{ } I[Speaker]}
{Highlighting}
{Shaded}

```

#### Use of Limiter in FM Receiver:

- **Primary function:** Removes amplitude variations/noise
- **Operation:** Clips the signal to provide constant amplitude
- **Benefits:**
  - Eliminates AM interference
  - Improves SNR
  - Ensures proper FM detection
  - Prevents false frequency demodulation
- **Location:** Placed between IF amplifier and FM detector

#### Mnemonic

“CARE” - Clips Amplitude, Removes noise, Ensures constant signal

### Question 3(c) [7 marks]

Draw and explain block diagram of super heterodyne receiver.

#### Solution

##### Super Heterodyne Receiver:

##### Mermaid Diagram (Code)

```

{Shaded}
{Highlighting}[]
graph LR
    A[Antenna] {--}{ } B[RF Amplifier]}
    B {--}{ } C[Mixer]}
    D[Local Oscillator] {--}{ } C}
    C {--}{ } E[IF Amplifier]}
    E {--}{ } F[Detector]}
    F {--}{ } G[Audio Amplifier]}
    G {--}{ } H[Speaker]}
    F {--}{ } I[AGC]}
    I {--}{ } B}
    I {--}{ } E}
{Highlighting}
{Shaded}

```

##### Function of each block:

- **Antenna:** Captures RF signals from electromagnetic waves
- **RF Amplifier:** Amplifies weak signals, provides selectivity
- **Local Oscillator:** Generates signal to mix with incoming RF
- **Mixer:** Produces IF by heterodyning RF with local oscillator
- **IF Amplifier:** Main amplification and selectivity at fixed frequency
- **Detector:** Extracts audio from modulated IF signal
- **Audio Amplifier:** Amplifies audio signal to drive speaker
- **AGC (Automatic Gain Control):** Maintains constant output level
- **Speaker:** Converts electrical signal to sound

##### Super Heterodyne Principle:

- Converts high-frequency RF to fixed IF for better amplification
- $IF = |RF \pm LO|$  (*typically 455kHz for AM, 10.7MHz for FM*)

### Mnemonic

“ARLMIDAS” - Antenna Receives, Local Mixes, IF Delivers, Audio Sounds

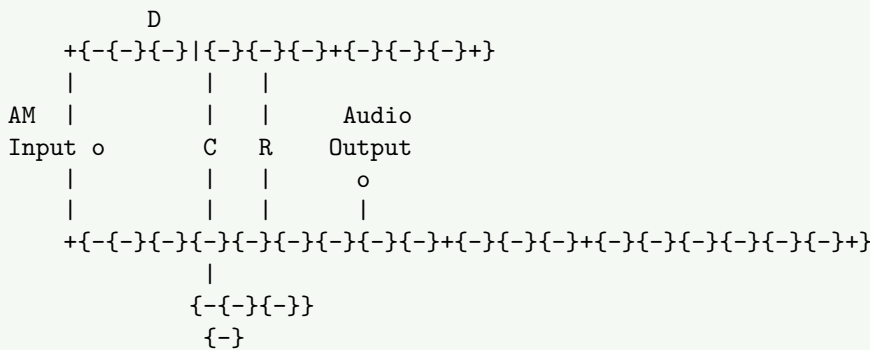
### Question 3(a) OR [3 marks]

Draw and explain block diagram for envelope detector.

#### Solution

**Envelope Detector:**

**Circuit Diagram:**



**Component Functions:**

- **Diode (D):** Rectifies AM signal (allows only positive half-cycles)
- **Capacitor (C):** Charges to peak of input, filters carrier frequency
- **Resistor (R):** Discharges capacitor, follows modulating signal envelope

**Operation:**

1. Diode conducts during positive half-cycles
2. Capacitor charges to peak voltage
3. During negative half-cycles, diode blocks
4. Capacitor discharges through resistor
5. RC time constant follows envelope variations

**RC Selection Criteria:**  $\frac{1}{f_c} \ll RC \ll \frac{1}{f_m}$

### Mnemonic

“DRIVER” - Diode Rectifies, RC Values Extract Envelope, Restores audio

### Question 3(b) OR [4 marks]

What is IF? Explain its importance in brief.

#### Solution

**Intermediate Frequency (IF):**

**Definition:** IF is a fixed frequency to which incoming RF signals are converted in superheterodyne receivers.

**Importance of IF:**

Aspect	Importance
Fixed Frequency	Allows optimized amplification at one frequency
Improved Selectivity	Fixed-tuned filters provide better adjacent channel rejection
Stable Gain	Consistent amplification across entire tuning range
Image Rejection	Helps reject image frequency interference
Simplified Tuning	Only local oscillator needs to be tuned for different stations
Better AGC	More effective gain control at fixed frequency



**Typical IF Values:**

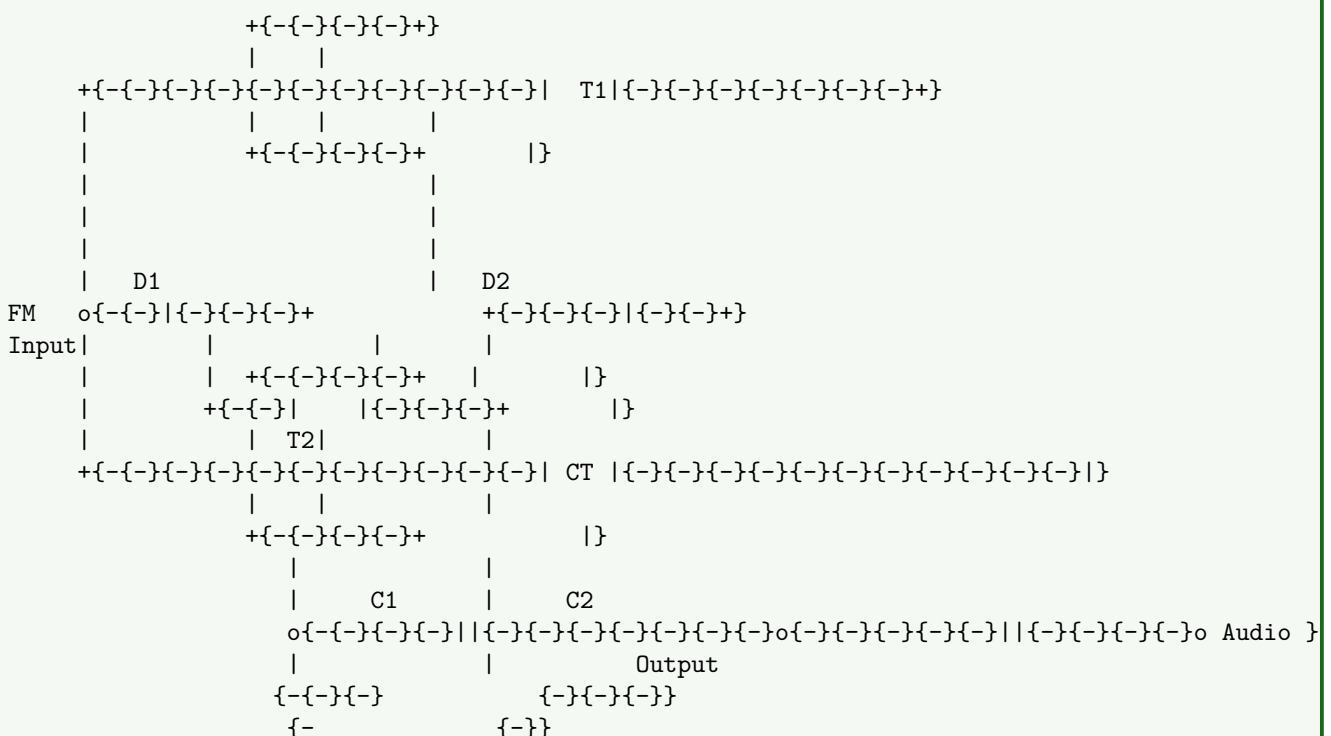
- AM receivers: 455 kHz
- FM receivers: 10.7 MHz
- Television: 45 MHz

**Mnemonic**

“FIGS-ST” - Fixed frequency, Improved selectivity, Gain stability, Simplified tuning

**Question 3(c) OR [7 marks]**

Explain phase discriminator circuit for FM detection.

**Solution****Phase Discriminator for FM Detection:****Circuit Diagram:****Operation:**

1. **Center-tapped transformer (T2)** creates  $180^\circ$  phase difference
1. **Primary transformer (T1)** sets reference phase
2. **Diode D1 and D2** form phase comparators
3. **When carrier at center frequency:**
  - Equal currents through both diodes
  - Equal voltages across C1 and C2
  - Net output is zero
4. **When frequency deviates:**
  - Phase changes
  - Unequal diode currents
  - Output voltage proportional to frequency deviation

**Advantages:**

- Good linearity
- Reduced distortion
- Better noise performance than slope detector

### Mnemonic

“PERFECT” - Phase Ensures Rectification For Extracting Carrier Transitions

### Question 4(a) [3 marks]

Explain quantization process and its necessity.

#### Solution

##### Quantization Process:

**Definition:** Quantization is the process of mapping continuous analog values to discrete digital levels.

##### Process:

1. Sampling converts continuous-time signal to discrete-time
2. Range of amplitudes divided into finite number of levels
3. Each sample assigned to nearest quantization level
4. Difference between original and quantized value is quantization error

##### Necessity of Quantization:

Necessity	Explanation
<b>Digital Processing</b>	Enables digital storage and manipulation
<b>Error Control</b>	Allows error detection and correction
<b>Noise Immunity</b>	Digital signals more resistant to noise
<b>Storage Efficiency</b>	More efficient than storing analog values
<b>Transmission</b>	Digital signals can be regenerated without error

### Mnemonic

“DENSE” - Digital conversion, Error control, Noise immunity, Storage, Efficient transmission

### Question 4(b) [4 marks]

Give difference between DM and ADM.

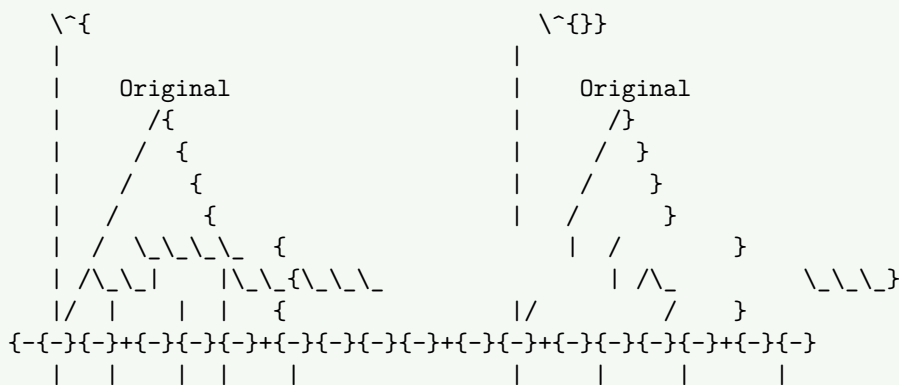
#### Solution

##### Difference between DM and ADM:

Parameter	Delta Modulation (DM)	Adaptive Delta Modulation (ADM)
<b>Step Size</b>	Fixed	Variable (adapts to signal)
<b>Slope Overload</b>	Common at steep signals	Reduced with adaptive step
<b>Granular Noise</b>	High for small signals	Reduced with smaller steps
<b>Signal Tracking</b>	Slow for rapidly changing signals	Better tracking of signal variations
<b>Complexity</b>	Simple	Moderate
<b>Bit Rate</b>	Higher for good quality	Lower for same quality
<b>Error</b>	More sensitive	More robust
<b>Performance</b>		

### Diagram:

DM:



ADM:



### Mnemonic

“SAVAGES” - Step size, Adaptable, Variable tracking, Avoids overload, Granular noise reduction, Error performance, Signal fidelity

### Question 4(c) [7 marks]

Draw & explain block diagram of PCM system.

### Solution

#### PCM System Block Diagram:

#### Mermaid Diagram (Code)

```
{Shaded}
{Highlighting}[]
graph LR
    subgraph "PCM Transmitter"
        A[Input Signal] --> B[Anti-aliasing Filter]
        B --> C[Sample & Hold]
        C --> D[Quantizer]
        D --> E[Encoder]
        E --> F[Parallel to Serial]
    end

    F --> G[Transmission Channel]

    subgraph "PCM Receiver"
        G --> H[Serial to Parallel]
        H --> I[Decoder]
        I --> J[Reconstruction Filter]
        J --> K[Output Signal]
    end
    {Highlighting}
    {Shaded}
```

#### PCM Transmitter:

- **Anti-aliasing Filter:** Limits input signal bandwidth to satisfy Nyquist criterion
- **Sample & Hold:** Converts continuous signal to discrete-time samples
- **Quantizer:** Approximates sample amplitudes to nearest discrete levels
- **Encoder:** Converts quantized levels to binary code
- **Parallel-to-Serial:** Converts parallel bits to serial for transmission

#### PCM Receiver:

- **Serial-to-Parallel:** Converts serial data back to parallel form
- **Decoder:** Converts binary code back to amplitude levels
- **Reconstruction Filter:** Smooths stepped output to recover analog signal

#### PCM Parameters:

- **Sampling rate:**  $f_s > 2f_m$  (Nyquist rate)
- **Quantization levels:**  $L = 2^n$  ( $n$  = number of bits)
- **Resolution:** Smallest distinguishable change =  $V_{max}/L$
- **Bit rate:**  $R = n \times f_{sbits}/second$

#### Mnemonic

“SAFE-PETS” - Sample, Amplify, Filter, Encode, Pulse train, Extract, Transform, Smooth

### Question 4(a) OR [3 marks]

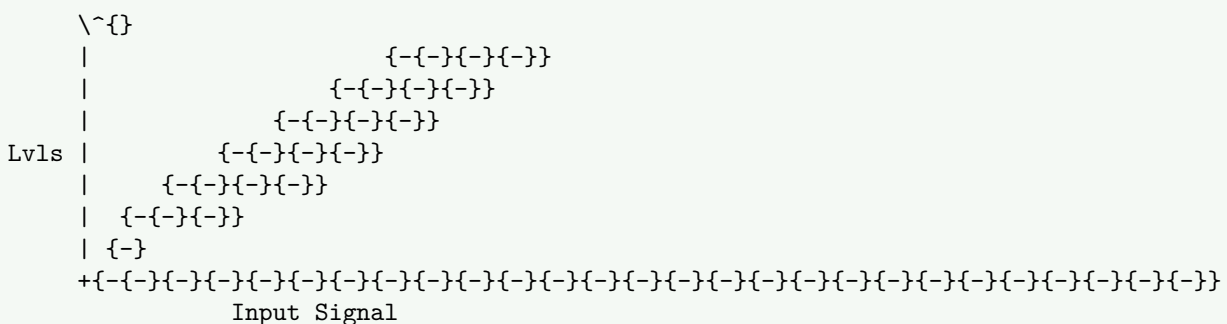
Define quantization. Explain non uniform quantization in brief.

#### Solution

**Quantization Definition:** Quantization is the process of converting continuous amplitude values to a finite set of discrete levels in analog-to-digital conversion.

#### Non-uniform Quantization:

##### Diagram:



#### Characteristics:

- Unequal step sizes throughout the amplitude range
- Smaller steps for low amplitudes, larger for high amplitudes
- Better matches human perception (logarithmic response)
- Improves SNR for small signals without increasing bit rate

#### Implementation Methods:

- **Companding:** Compressing at transmitter, expanding at receiver
- **Logarithmic coding:**  $\mu$ -law (North America) and A-law (Europe)
- **Adaptive quantization:** Adjusts levels based on signal statistics

#### Mnemonic

“CLASP” - Compressed Levels, Adaptive Steps, Small steps for small signals, Perceptual matching

### Question 4(b) OR [4 marks]

Explain Adaptive delta modulation with its application.

#### Solution

#### Adaptive Delta Modulation (ADM):

##### Diagram:

Mermaid Diagram (Code)

{Shaded}

```

{Highlighting}[]
graph LR
    A[Input Signal] --> B[Comparator]
    B --> C[1-bit Quantizer]
    C --> D[Transmission Channel]
    D --> E[Step Size Control]
    E --> F[Integrator]
    F --> Feedback --> B
    F --> G[Output Signal]
    C --> Controls --> E
{Highlighting}
{Shaded}

```

#### Operation:

- Adapts step size based on input signal slope
- Increases step size for rapid changes (prevents slope overload)
- Decreases step size for slow changes (reduces granular noise)
- Uses previous bits pattern to determine slope changes

#### Advantages:

- Better signal tracking than DM
- Lower bit rate for same quality
- Reduced slope overload and granular noise
- Wider dynamic range

#### Applications:

- Speech and audio compression
- Voice-grade communication channels
- Digital telephony systems
- Video signal encoding
- Telemetry systems

#### Mnemonic

“ADAPT” - Automatically Decides Appropriate Pulse Transitions

### Question 4(c) OR [7 marks]

What is sampling? Explain types of sampling in brief.

#### Solution

**Sampling Definition:** Sampling is the process of converting a continuous-time signal to a discrete-time signal by taking measurements at regular intervals.

#### Types of Sampling:

Type	Description	Diagram
<b>Ideal Sampling</b>	Instantaneous samples of infinitesimal duration	Impulses at sampling instants
<b>Natural Sampling</b>	Samples have finite width, amplitude follows input	Original signal visible during sampling duration
<b>Flat-top Sampling</b>	Samples have constant amplitude during sampling interval	Step-like appearance, used in sample-and-hold

**Diagrams:**

**Ideal Sampling:**

**Natural Sampling:**

**Flat{-top Sampling:}**

**Sampling Parameters:**

- Sampling period ( $T_s$ ):** Time between consecutive samples
- Sampling frequency ( $f_s$ ):** Number of samples per second ( $f_s = 1/T_s$ )
- Nyquist rate:** Minimum sampling rate ( $f_s > 2f_m$ ) to avoid aliasing

Mnemonic
“INFS” - Ideal (impulses), Natural (follows signal), Flat-top (constant), Sufficient rate

Mnemonic
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Question 5(a) [3 marks]

Define bit rate and baud rate.

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Solution			
Bit Rate and Baud Rate:			
Parameter	Definition	Formula	Unit
<b>Bit Rate</b>	Number of binary digits (bits) transmitted per second	$R = f_s \times n$	bits per second (bps)
<b>Baud Rate</b>	Number of signal elements or symbols transmitted per second	$B = f_s$	symbols per second (baud)

**Relationship:**

- For binary signaling: Bit Rate = Baud Rate
- For M-ary signaling: Bit Rate = Baud Rate  $\times \log_2 M$ 
  - Where M = number of different signal elements

**Example:**

- 4-QAM (M=4): Each symbol carries  $\log_2 4 = 2 \text{ bits}$
- If Baud Rate = 1000 symbols/s, then Bit Rate = 2000 bits/s

Solution			
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- | Solution                |   |                    |                           |
|-------------------------|---|--------------------|---------------------------|
| Bit Rate and Baud Rate: |   |                    |                           |
| Parameter               | Definition  | Formula            | Unit                      |
| <b>Bit Rate</b>         | Number of binary digits (bits) transmitted per second       | $R = f_s \times n$ | bits per second (bps)     |
| <b>Baud Rate</b>        | Number of signal elements or symbols transmitted per second | $B = f_s$          | symbols per second (baud) |

**Relationship:**

  - For binary signaling: Bit Rate = Baud Rate
  - For M-ary signaling: Bit Rate = Baud Rate  $\times \log_2 M$ 
    - Where M = number of different signal elements

**Example:**

  - 4-QAM (M=4): Each symbol carries  $\log_2 4 = 2 \text{ bits}$
  - If Baud Rate = 1000 symbols/s, then Bit Rate = 2000 bits/s

Solution			
Bit Rate and Baud Rate:			
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**Mnemonic**

“BBSM” - Bits per second, Baud for Symbols, Modulation determines relationship

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**Question 5(b) [4 marks]**

**Explain working of DPCM.**

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Solution
<p><b>Differential Pulse Code Modulation (DPCM):</b></p> <p><b>Block Diagram:</b></p> <p style="text-align: center;">Mermaid Diagram (Code)</p> <pre> graph LR     subgraph Shaded         direction TB         A[Input] --&gt; B[Subtractor]         B --&gt; C[Adder]         C --&gt; D[Output]         C --&gt; E[Quantizer]         E --&gt; F[Feedback]         F --&gt; B     end     style Shaded fill:#f0f0f0     style B fill:#f0f0f0     style C fill:#f0f0f0     style D fill:#f0f0f0     style E fill:#f0f0f0     style F fill:#f0f0f0     style A fill:#f0f0f0     style I[Initial Value] fill:#f0f0f0     I --&gt; B     </pre>

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```

subgraph "Transmitter"
direction LR
A[Input] --> B[Difference]
B --> C[Quantizer]
C --> D[Encoder]
D --> E[Output]
F[Predictor] --> B
C --> F
end

subgraph "Receiver"
direction LR
G[Input] --> H[Decoder]
H --> I[Output]
H --> J[Predictor]
J --> I
end
{Highlighting}
{Shaded}

```

#### Working Principle:

- Encodes difference between current sample and predicted sample
- Prediction based on previous samples (correlation)
- Smaller dynamic range of differences allows fewer bits per sample

#### Advantages:

- Higher compression ratio than PCM
- Reduced bit rate for same quality
- Exploits signal correlation
- Improved SNR performance

#### Mnemonic

“DEEP” - Difference Encoded, Efficient Prediction, Exploits correlation, Preserves quality

### Question 5(c) [7 marks]

The binary data 1011001 is to be transmitted using following line coding techniques: (i) Unipolar RZ and NRZ (ii) Polar RZ and NRZ (iii) AMI (iv) Manchester. Draw all the waveforms.

#### Solution

##### Line Coding Waveforms for 1011001:

Data:	1	0	1	1	0	0	1
	v	v	v	v	v	v	v

##### Unipolar

NRZ:     \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ \\_ |   |   \\_ \\_ \\_ \\_ | \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ \\_

##### Unipolar

RZ:     \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ |   \\_ \\_ \\_ \\_ \\_ \\_ \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ \\_ \\_ \\_ \\_

##### Polar

NRZ:     \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ \\_ |   |   \\_ \\_ \\_ \\_ | \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ \\_

##### Polar

RZ:     \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ |   \\_ \\_ \\_ \\_ \\_ \\_ \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ \\_ \\_ \\_ \\_

AMI:     \\_ \\_ \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ \\_ \\_ \\_ \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ |   \\_ \\_ \\_ \\_ \\_ \\_ \\_ \\_ \\_ \\_ |   \\_ \\_ \\_ \\_ \\_ \\_ \\_

[illegible]

### Description of Line Coding Techniques:

Technique	Logic 1	Logic 0	Characteristics
<b>Unipolar NRZ</b>	High level	Zero level	No return to zero between bits
<b>Unipolar RZ</b>	Pulse for half bit	Zero level	Returns to zero for half bit
<b>Polar NRZ</b>	Positive	Negative	No return to zero between bits
<b>Polar RZ</b>	Positive pulse	Negative pulse	Returns to zero for half bit
<b>AMI</b>	Alternating +/-	Zero level	Alternates polarity for consecutive 1s
<b>Manchester</b>	High	Low	Transition in middle of bit

### Mnemonic

“UPAM” - Unipolar, Polar, AMI, Manchester encoding options

Question 5(a) OR [3 marks]

**Compare RZ and NRZ coding with example.**

## Solution

### Comparison of RZ and NRZ Coding:

Parameter	Return-to-Zero (RZ)	Non-Return-to-Zero (NRZ)
<b>Signal levels</b>	Returns to zero in each bit	Maintains level for full bit period
<b>Bandwidth</b>	Higher ( $\approx 2 \times NRZ$ )	Lower
<b>Self-clocking</b>	Better (transitions in every bit)	Poorer (may have long runs without transitions)
<b>Power requirement</b>	Higher	Lower
<b>Bit synchronization</b>	Easier	More difficult
<b>Implementation</b>	More complex	Simpler
<b>DC component</b>	Less	More

Example for binary data 101:

Data:	1	0	1
	y	y	y

**NRZ:**

RZ: \\_ \\_ \\_ | \\_ \\_ \\_ | \\_ \\_ \\_ | \\_ \\_ \\_ | \\_ \\_ \\_ | \\_ \\_ \\_ | \\_ \\_ \\_ | \\_ \\_ \\_ |

### Mnemonic

“BPSIDC” - Bandwidth, Power, Synchronization, Implementation, DC component

**Question 5(b) OR [4 marks]**

Explain delta modulation in brief.



## Solution

### Delta Modulation (DM):

#### Block Diagram:

#### Mermaid Diagram (Code)

```
{Shaded}
{Highlighting}[]
graph LR
    A[Input Signal] --> B[Comparator]
    B --> C[1-bit Quantizer]
    C --> D[Transmission]
    D --> E[Integrator]
    E -- Feedback --> B
    E --> F[Integrator]
    F --> G[Output Signal]
{Highlighting}
{Shaded}
```

#### Working Principle:

- Encodes only the difference between samples using 1 bit
- Comparator checks if input is higher/lower than predicted value
- Integrator accumulates the bits to approximate original signal
- Output is series of 1s and 0s representing up/down steps

#### Limitations:

- **Slope Overload:** Cannot track rapidly changing signals
- **Granular Noise:** Small variations around steady signal

#### Advantages:

- Simplest form of differential encoding
- Low bit rate (1 bit per sample)
- Simple implementation
- Hardware efficiency

## Mnemonic

“SIDE” - Single-bit, Integrates Differences, Encodes changes

## Question 5(c) OR [7 marks]

Explain PCM-TDM system.

## Solution

### PCM-TDM System:

#### Block Diagram:

#### Mermaid Diagram (Code)

```
{Shaded}
{Highlighting}[]
graph TD
    subgraph "Transmitter"
        A1[Channel 1] --> B1[LPF]
        A2[Channel 2] --> B2[LPF]
        A3[Channel 3] --> B3[LPF]
        A4[Channel n] --> B4[LPF]
        B1 --> C[Multiplexer]
        B2 --> C
        B3 --> C
        B4 --> C
        C --> D[Sample & Hold]
        D --> E[Quantizer]
    end
```

```

E {-}{-}{ } F[Encoder]}
F {-}{-}{ } G[Frame Generator]}
G {-}{-}{ } H[Line Coder]}
H {-}{-}{ } I[Transmission Medium]}
end

subgraph "Receiver"
I {-}{-}{ } J[Line Decoder]}
J {-}{-}{ } K[Frame Sync]}
K {-}{-}{ } L[Decoder]}
L {-}{-}{ } M[Demultiplexer]}
M {-}{-}{ } N1[LPF]}
M {-}{-}{ } N2[LPF]}
M {-}{-}{ } N3[LPF]}
M {-}{-}{ } N4[LPF]}
N1 {-}{-}{ } O1[Channel 1]}
N2 {-}{-}{ } O2[Channel 2]}
N3 {-}{-}{ } O3[Channel 3]}
N4 {-}{-}{ } O4[Channel n]}
end
{Highlighting}
{Shaded}

```

#### PCM-TDM Operation:

Stage	Process
<b>Filtering</b>	Band-limits each channel to prevent aliasing
<b>Multiplexing</b>	Samples each channel sequentially
<b>Conversion</b>	Quantizes samples and converts to binary code
<b>Framing</b>	Adds sync bits and channel identification
<b>Transmission</b>	Sends frame over communication medium
<b>Demultiplexing</b>	Separates channels from received frame
<b>Reconstruction</b>	Converts digital samples back to analog signals

#### System Parameters:

- **Channel Capacity:** N channels
- **Sampling Rate:**  $f_s$  per channel
- **Quantization:** n bits per sample
- **Frame Structure:** 1 sample from each channel + sync
- **Total Bit Rate:**  $N \times n \times f_s + \text{overhead}$

#### Mnemonic

“MOST-FDR” - Multiplex, Quantize, Sample, Transmit, Frame, Demultiplex, Reconstruct