

Spring 2017

(Test 1 Sol)

- ✓ 1. What is first order entropy? A binary source has the probabilities $p_1 = 0.2$ and $p_2 = 0.8$. Calculate first order entropy for the source. [10 points]

- ✓ 2. Complete the following matrices:

| | | |
|---|---|---|
| 1 | 5 | 9 |
| 3 | 2 | 4 |

1:2 up sampling H & V
Based on ZOH
→

| | | | | | |
|--|--|--|--|--|--|
| | | | | | |
| | | | | | |
| | | | | | |
| | | | | | |

| | |
|---|---|
| 1 | 7 |
| 3 | 6 |

1:2 up sampling H & V
Based on FOH
→

| | | | |
|--|--|--|--|
| | | | |
| | | | |
| | | | |
| | | | |

[20 points]

- ✓ 3. Describe pseudorandom noise quantization (dithering) with a block diagram. What are the advantages of dithering? [10 points]

- ✓ 4. For a uniform quantizer number of levels $L = 32$. Calculate the number of bits per level and SNR for the quantizer. If the number of output levels is doubled to $L = 64$ then the SNR will be changed by how many dB (specify increase or decrease) and how the bandwidth will be affected. [10 points]

- ✓ 5. With sketch describe compandor. Is it a uniform or nonuniform quantizer? [10 points]

- ✓ 6. Describe contrast quantization (sketch). What are the advantages? [10 points]

- ✓ 7. Describe what is half tone image generation (sketch)? [10 points]

- ✓ 8. Define and explain clearly with figures:

- Uniform Sampling
- Non-Uniform sampling
- Nyquist Theorem
- Aliasing

[20 points]

1) First order entropy:

It is defined as minimum theoretical bit rate at which a group of 'N' samples can be coded. It is given by

$$H(x) = - \sum_{i=1}^L P_i \log_2 P_i$$

P_i = probability of symbol 'i'

L = # of levels.

→ Given

$$P_1 = 0.2, P_2 = 0.8$$

$$\text{First order entropy } H(x) = - \sum_{i=1}^2 P_i \log_2 P_i$$

$$= - [0.2 \log_2 0.2 + 0.8 \log_2 0.8]$$

$$= - [-0.464 + (-0.257)]$$

$$= 0.721 \text{ bits / symbol}$$

2)

| | | |
|---|---|---|
| 1 | 5 | 9 |
| 3 | 2 | 4 |

1:2 up sam
→
ZOH

| | | | | | |
|---|---|---|---|---|---|
| 1 | 1 | 5 | 5 | 9 | 9 |
| 1 | 1 | 5 | 5 | 9 | 9 |
| 3 | 3 | 2 | 2 | 4 | 4 |
| 3 | 3 | 2 | 2 | 4 | 4 |

| | |
|---|---|
| 1 | 7 |
| 3 | 6 |

1:2 up
Sampling
→

2x2

| | | | |
|---|---|---|---|
| 1 | 0 | 7 | 0 |
| 0 | 0 | 0 | 0 |
| 3 | 0 | 6 | 0 |
| 0 | 0 | 0 | 0 |

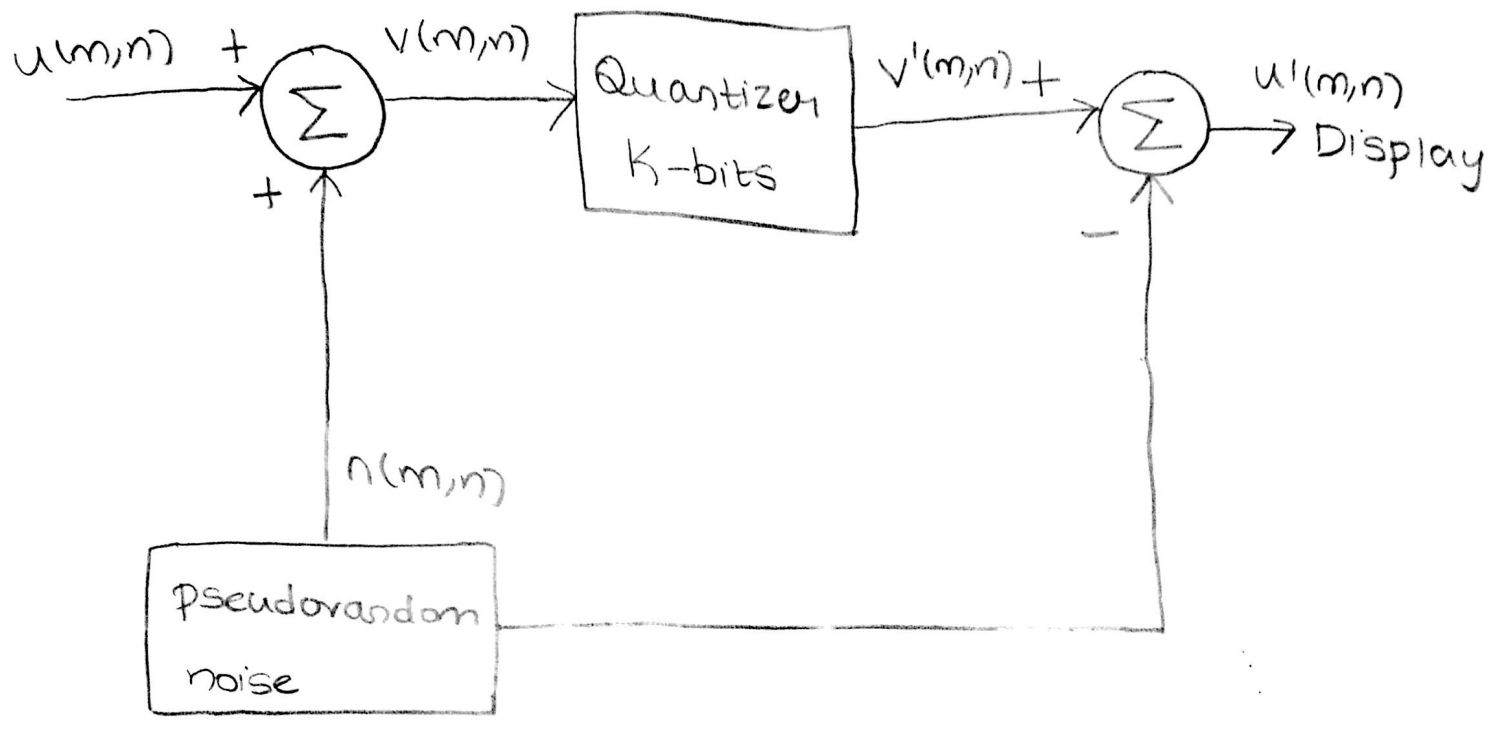
4x4

↓ FOH

| | | | |
|-----|------|-----|------|
| 1 | 4 | 7 | 3.5 |
| 2 | 4.25 | 6.5 | 3.25 |
| 3 | 4.5 | 6 | 3 |
| 1.5 | 2.25 | 3 | 1.5 |

4x4

3) pseudorandom noise quantization: (Dithering)



Block diagram

→ To Suppress the contouring effects, uniformly distributed pseudorandom noise is added to the input luminance values before quantization.

→ This pseudorandom noise is also called dither.

→ Advantage:

→ Same amount of pseudorandom noise is subtracted

From the quantizer output to display the image.

→ The effect is that noise breaks the contours by making pixels go above or below the decision levels.

Few more points:

→ During the display, the noise fills in the regions of contours in such a way that spatial averaging is minimized.

→ Amount of noise added should be small enough to maintain the spatial resolution and large enough to allow luminance values to vary randomly.

4) Given

$$L=32$$

To find number of bits?

$$L = 2^B = 2^5$$

\therefore Number of bits = 5

$$SNR = 10 \log_{10} L^2$$

$$= 10 \log(32^2)$$

$$= \underline{\underline{30.10 \text{ dB}}}$$

\rightarrow L is doubled - $L = 64$

$$L = 2^B = 2^6$$

\therefore Number of bits = 6

$$SNR = 10 \log_{10} L^2$$

$$= 10 \log(64^2)$$

$$= \underline{\underline{36.12 \text{ dB}}}$$

\therefore If L is doubled, SNR will increase by roughly 6 dB.

Penalty:

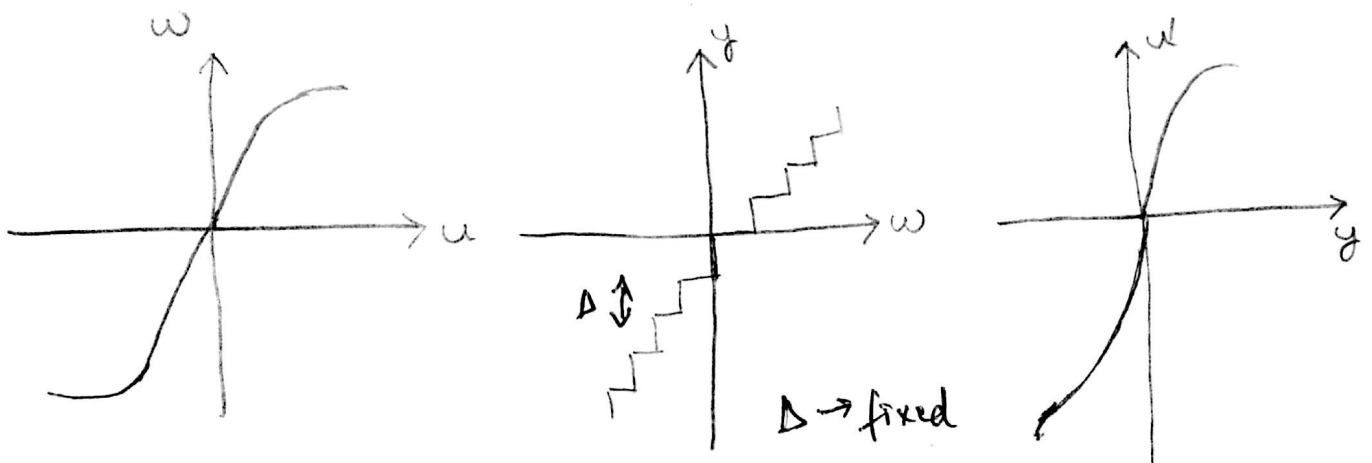
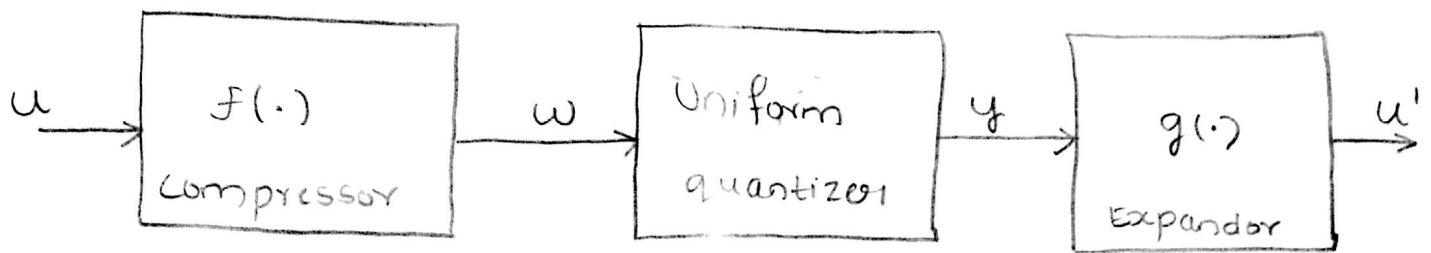
If L increases, then Bandwidth will be doubled.

\uparrow

$\uparrow\uparrow$

5) Compandor: (compressor-expandor)

→ Compandor is a uniform quantizer preceded and succeeded by a non-linear transformation.



→ First, input luminance value is passed through a non-linear memoryless transformation $f(\cdot)$ to yield output w .

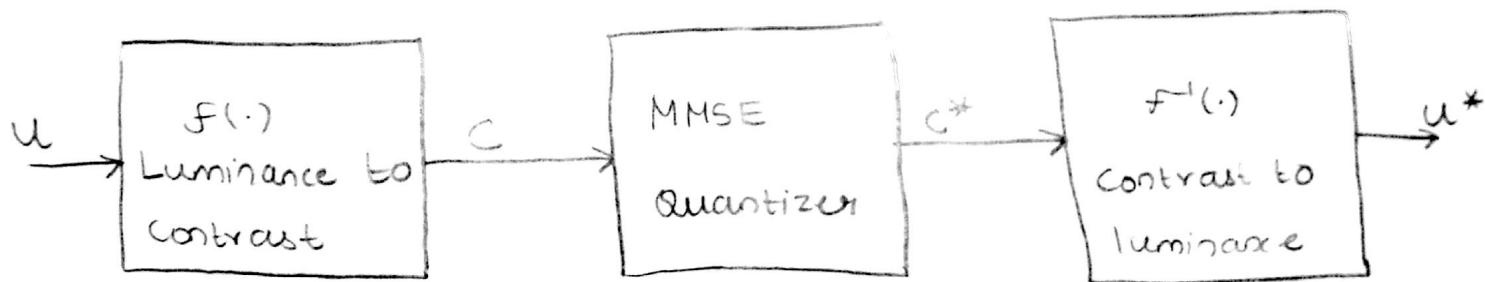
→ ' w ' will undergo uniform quantization to yield output y .

(4)

→ 'y' is passed through a non-linear transformation to get the output u' .

→ The overall transformation from u to u' is nonuniform quantizer.

6) Contrast quantization:



→ In Contrast quantization, quantization is performed on contrast values instead of luminance values.

→ First, luminance values are converted to Contrast using non-linear transformation $F(\cdot)$.

→ Contrast values ' c ' are passed through a MMSE

quantizer.

→ Contrast values are given by

$$\begin{aligned} C &= \alpha \ln(1 + \beta u) , & 0 \leq u \leq 1 \\ C &= \alpha u^\beta , & \alpha = 1, \beta = \frac{1}{3} \end{aligned}$$

↓
Suggested values.

→ Contrast undergoes quantization using 4 to 5 bpp rather than 6bpp taken by uniform quantizer.

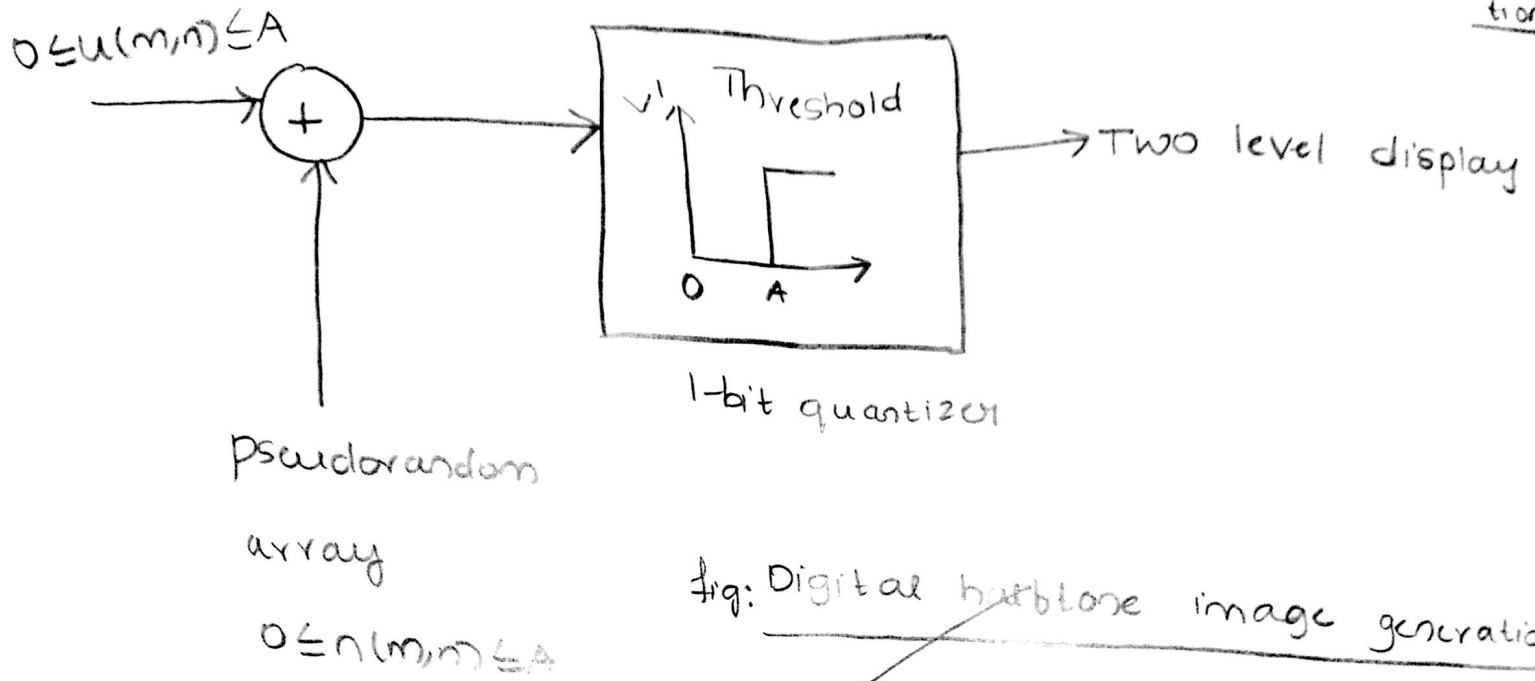
→ Finally, contrast values are converted to the luminance values using inverse transformation $S^{-1}(\cdot)$.

Advantages:

- It decreases the ~~contouring~~ effect.
- Experimental results show 2% increase in the contrast, makes visible to human eye.

7) Half tone image generation:

→ They are binary images used for grayscale reproduction



→ To the input luminance values, random values from pseudorandom array are added.

→ The output $v(m,n)$ is sent to a 1-bit quantizer

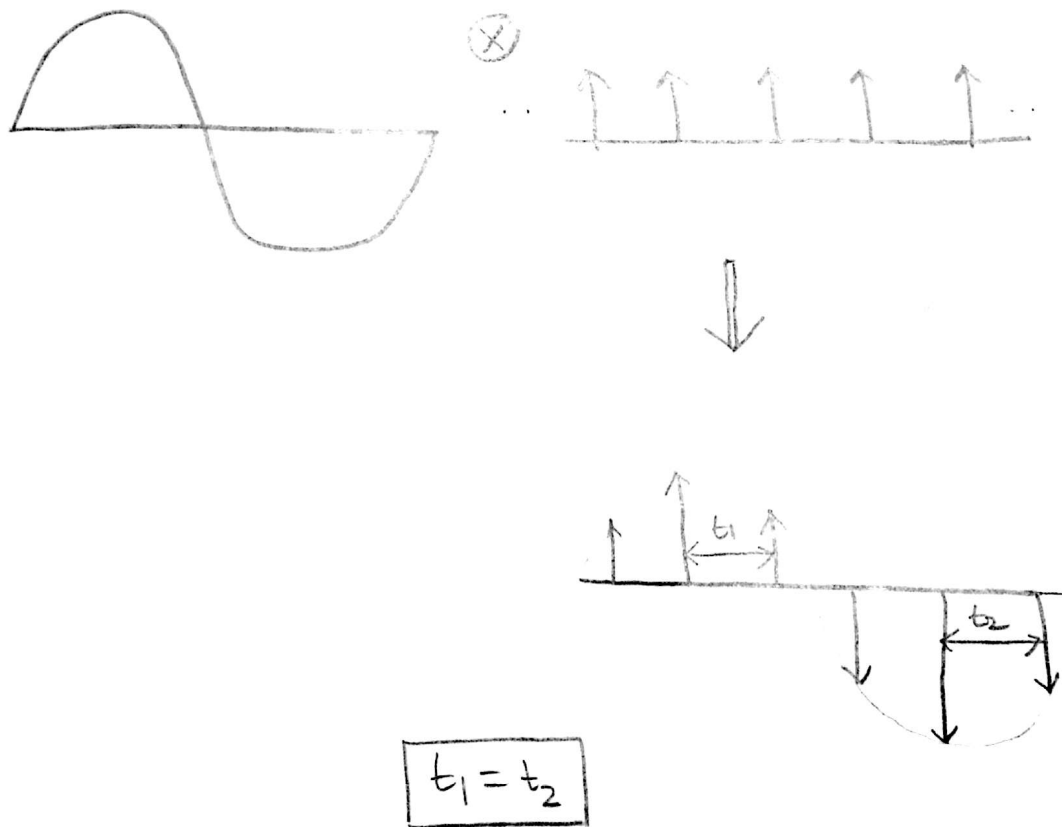
The output of the quantizer (0/1) represents black or white dot.

→ Halftone image generation in grayscale is due to spatial averaging performed by the eye.

8)

a) Uniform sampling:

→ During sampling, an analog signal is said to be uniformly sampled, if the sampling interval is constant (between any two samples).

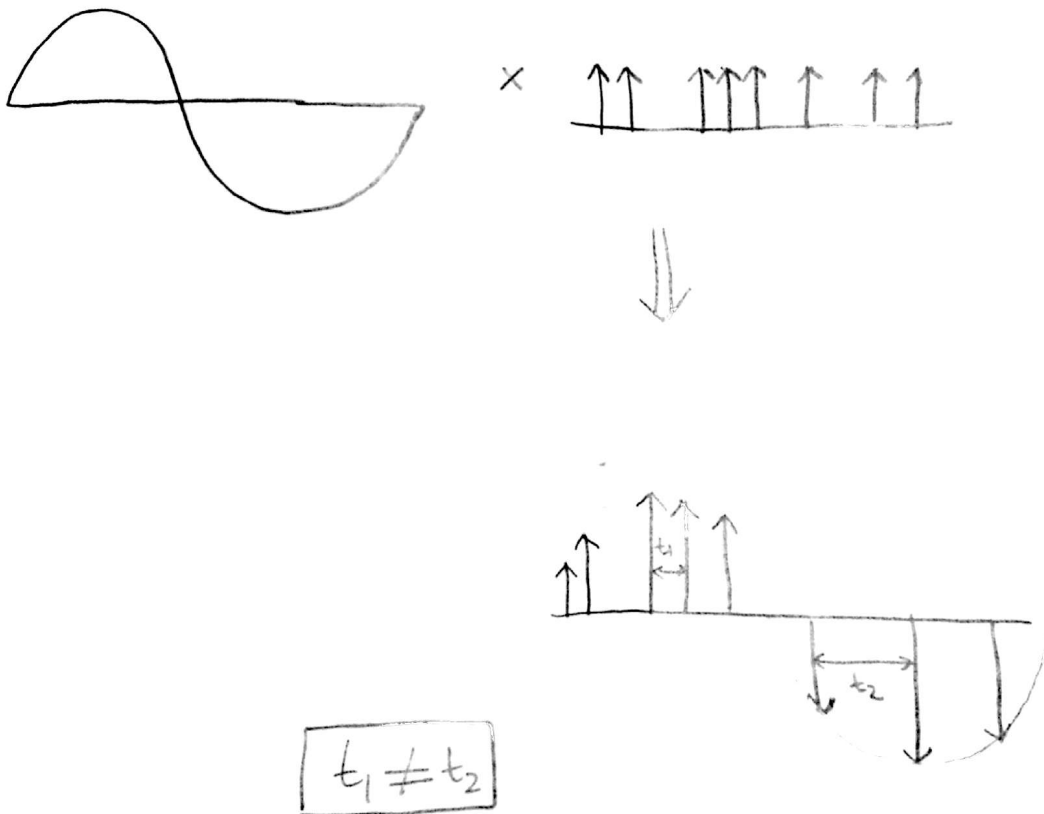


b) Non-uniform sampling:

→

⑥

→ During sampling, an analog signal is said to be ^{non-}uniformly sampled, if the sampling interval is not same between any two samples.



c) Nyquist theorem:

It states that, an analog signal can be successfully reconstructed from its samples, if it is sampled at a rate equal to or greater than the.

twice the bandwidth of the signal.

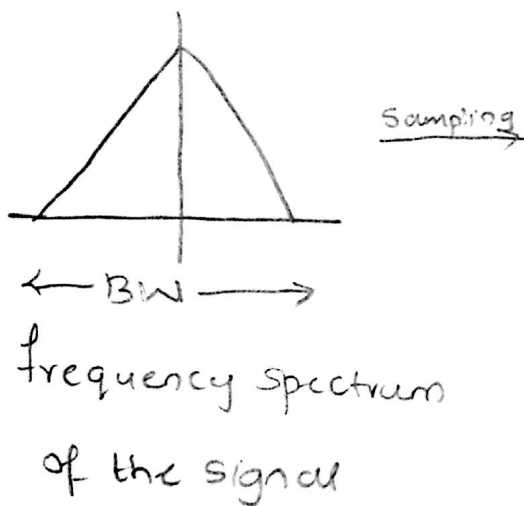
let min freq = f_{min}

max freq = f_{max}

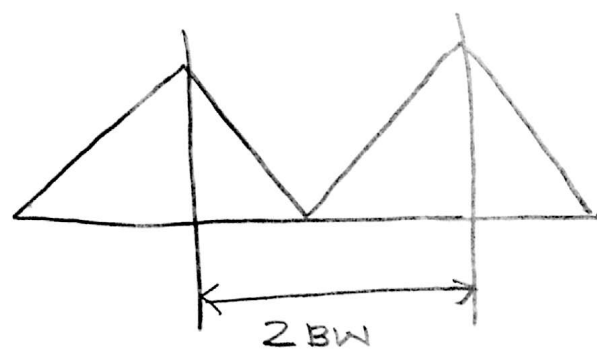
Bandwidth of the signal $BW = f_{max} - f_{min}$

$f_s > 2BW$

let $f_s = 2BW$



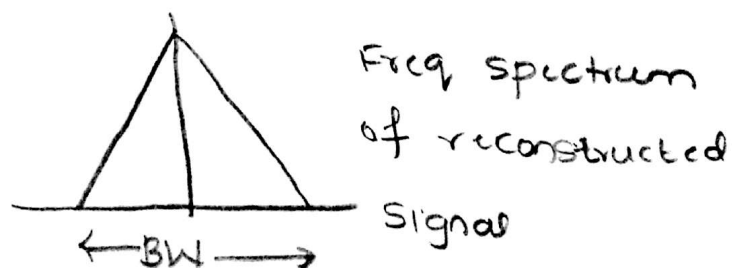
Sampling



Frequency spectrum of the sampled signal



low pass filter



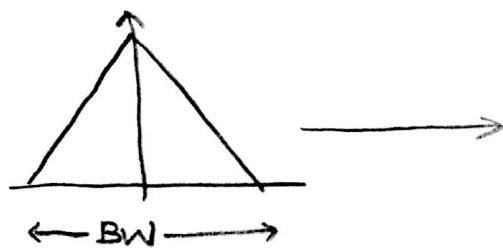
d) Aliasing:

⑦

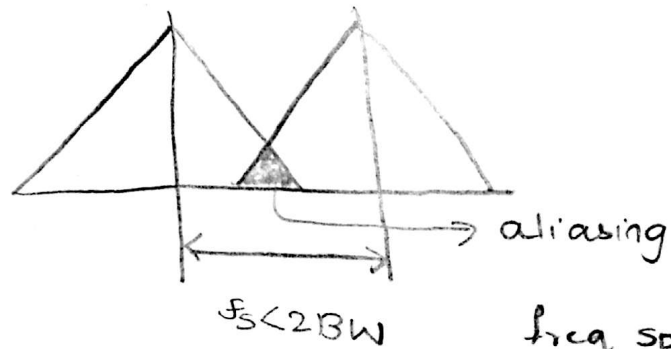
→ If the Nyquist theorem is not followed and $f_s < 2BW$ then the frequency spectrum of sampled signal looks like overlapping version of the frequency spectrum of the signal.

→ This results in the foldover frequencies.

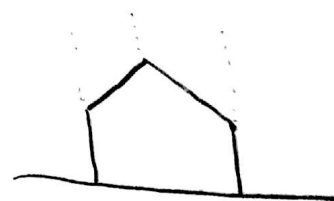
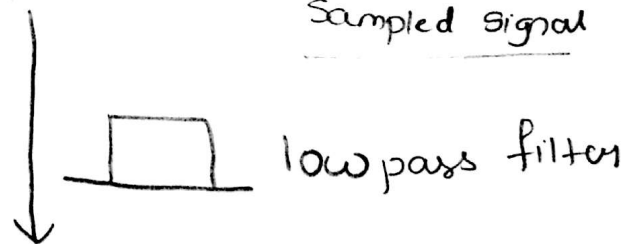
→ This is known as aliasing effect.



frequency spectrum
of the signal



freq. spectrum of
sampled signal



reconstructed signal

freq. spectrum
of reconstructed
signal.