#### 1

# Gate Assignment 4

### Yashas Tadikamalla - AI20BTECH11027

Download all python codes from

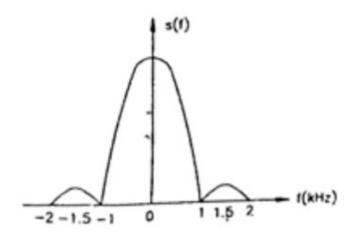
https://github.com/YashasTadikamalla/EE3900/blob/main/GateAssignment4/codes

and latex-tikz codes from

https://github.com/YashasTadikamalla/EE3900/blob/main/GateAssignment4/GateAssignment4.tex

## 1 Problem (EC-1997 Q1.10)

A deterministic signal has the power spectrum given in the figure. The minimum sampling rate needed to completely represent this signal is



- 1) 1*KHz*
- 2) 2*KHz*
- 3) 3KHz
- 4) None

#### 2 Solution

**Definition 2.1** (Normalised sinc function). A normalised sinc function is defined as

$$sinc(x) = \begin{cases} 1, & x = 0\\ \frac{sin(\pi x)}{\pi x}, & x \neq 0 \end{cases}$$
 (2.0.1)

**Definition 2.2** (Power spectrum). *Power Spectral density, or simply, Power spectrum, denoted by* s(f) *is defined as* 

$$s(f) = |X(f)|^2$$
 (2.0.2)

**Theorem 2.1** (Sampling Theorem). If a signal contains no frequency components above W Hz, then the sampling rate at which the continuous time signal needs to be sampled uniformly, so as to completely recover the original signal is given by

$$f_s \ge 2W \tag{2.0.3}$$

**Definition 2.3** (Nyquist rate). *Minimum sampling rate is also called as Nyquist rate. It is given by* 

$$f_s = 2W \tag{2.0.4}$$

Given, power spectrum of a deterministic signal. From (2.2), Fourier transform of the given band limited signal is **truncated normalised sinc pulse**. As no frequency component exceeds 2KHz,

$$W = 2KHz \tag{2.0.5}$$

From (2.0.4),

$$f_s = 2W = 4KHz \tag{2.0.6}$$

Hence, option 4 is the correct answer.

To verify the validity of (2.1), let's see what happens if we sample at a rate lower than Nyquist rate.

Let our original continuous time signal be x(t). Consider impulse train  $x_i(t)$  given by

$$x_i(t) = \sum_{n = -\infty}^{\infty} \delta(t - nT_s)$$
 (2.0.7)

where  $T_s$  is the sampling period. ( $f_s = \frac{1}{T_s}$  is the sampling frequency). The sampled signal would be

$$x_s(t) = x(t)x_i(t)$$
 (2.0.8)

$$=\sum_{n=-\infty}^{\infty}x(nT_s)\delta(t-nT_s)$$
 (2.0.9)

Also, from (2.0.8)

$$X_s(f) = X(f) * X_i(f)$$
 (2.0.10)

$$= \frac{1}{T_s} \sum_{k=-\infty}^{\infty} X(f - kf_s)$$
 (2.0.11)

 $X_s(f)$  consists periodically repeated copies of X(f), shifted by integer multiples of  $f_s$ . For our example, X(f) is the **truncated normalised sinc pulse**. Let W be the maximum frequency component.

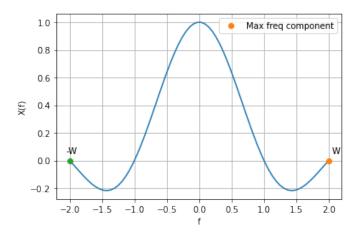


Fig. 4: Plot of X(f)

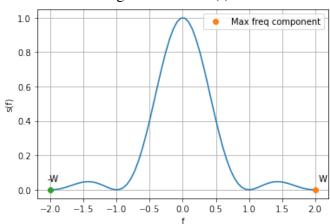
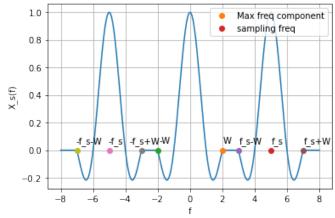
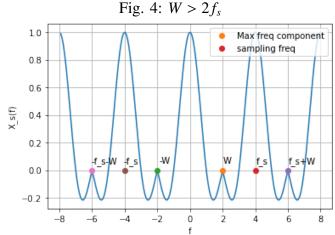


Fig. 4: Plot of s(f)

- Case 1:  $W > 2f_s$ : The copies don't overlap. Hence, x(t) can be recovered from  $x_s(t)$  using a low pass filter.
- Case 2:  $W = 2f_s$ : The copies just touch each other, but don't overlap. So, x(t) can be recovered from  $x_s(t)$  using a low pass filter.
- Case 3:  $W < 2f_s$ : As the copies overlap, they get added. So, we cannot reconstruct the original signal x(t). This gives rise to situation called aliasing.





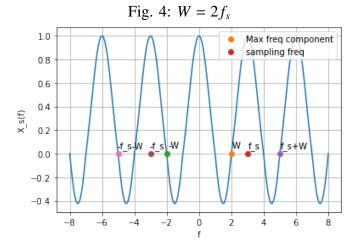


Fig. 4:  $W < 2f_s$ 

Hence, we need to sample at a rate higher than Nyquist rate, inorder to be able to recover the original signal.