

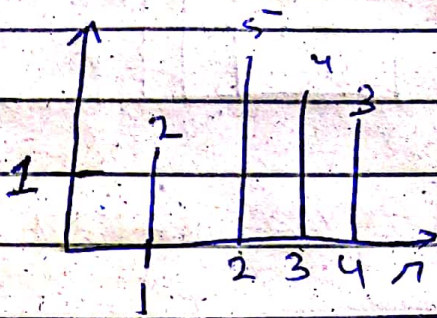


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Module	4 th
Subject	Electronic Circuit Desing
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Assingment	Sessional

Q No 1

Evaluate the even and odd.

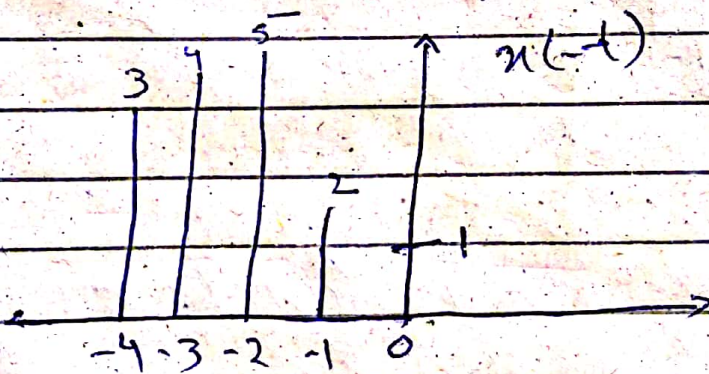
Ans:



$$\text{even} \rightarrow x_e(t) = \frac{x(t) + x(-t)}{2} = \frac{1}{2} [x(t) + x(-t)]$$

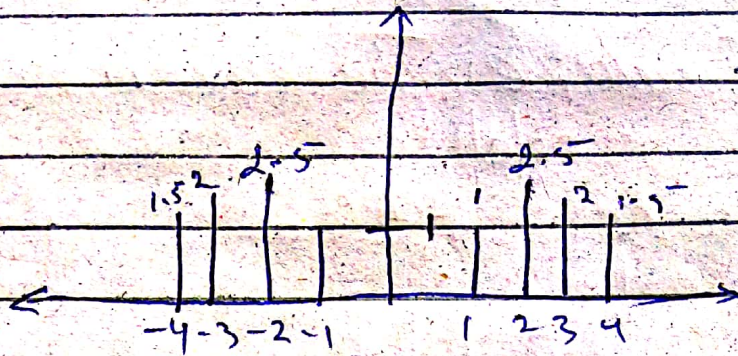
$$\text{odd} \rightarrow x_o(t) = \frac{x(t) - x(-t)}{2} = \frac{1}{2} [x(t) - x(-t)]$$

so find $x(-t)$



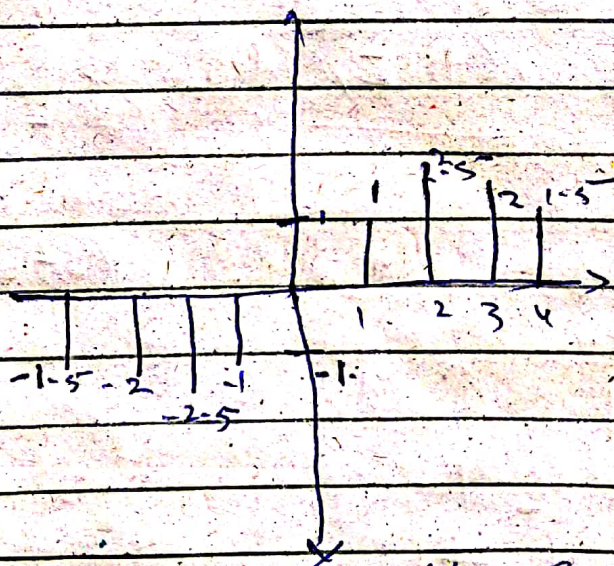
Now even function is

even function is



So this is the even function

for odd function



This is odd function of
give signal.

In both even and odd
The Amplitude is half of
The given signal because
it's divided by (2) two.

Q No 2 :-

Ans.

$$Y(s) = \frac{s+4}{s^2+4s-12}$$

$$Y(s) = \frac{s+4}{s^2-2s+6s-12}$$

$$Y(s) = \frac{s+4}{s(s-2)+6(s-2)}$$

$$Y(s) = \frac{s+4}{(s-2)(s+6)}$$

The form for the partial fraction expansion is

$$\frac{s+4}{(s-2)(s+6)} = \frac{A}{s-2} + \frac{B}{s+6}$$

multiply both side $(s-2)(s+6)$

then

$$Y(s) = \frac{s+4}{s+4} = A(s+6) + (s-2)B$$

so

find

$$s+4 = A(s+6) + (s-2)B$$

if $s = 2$ so

$$2+4 = A(2+6) + (2-2)B$$

$$6 = A8 \Rightarrow A = \frac{6}{8} = \frac{3}{4}$$

$$\boxed{A = \frac{3}{4}}$$

if $s = -6$ Then

$$s+4 = A(s+6) + (s-2)B$$

$$-6+4 = A(-6+6) + (-6-2)B$$

$$-2 = -8B \Rightarrow B = \frac{-2}{-8} = \frac{1}{4}$$

$$\boxed{B = \frac{1}{4}}$$

Then

$$Y(s) = \frac{\frac{3}{4}}{s-2} + \frac{\frac{1}{4}}{s+6}$$

$$\mathcal{L}^{-1}(Y) = \frac{3}{4} \mathcal{L}^{-1}\left(\frac{1}{s-2}\right) + \frac{1}{4} \mathcal{L}^{-1}\left(\frac{1}{s+6}\right)$$

$$\boxed{\mathcal{L}^{-1}(Y) = \frac{3}{4} e^{2t} + \frac{1}{4} e^{-6t}}$$

Q no 3 part (i) :-

Ans :-

Analog signal is converted to a digital signal using two step process.

(1) sampling :-

sampling converts a continuous time continuous amplitude (real value) signal to discrete time continuous amplitude (still real valued) signal. Remember only time axis is discretized and not the amplitude axis.

(2) Quantization.

Quantization converts the discrete time continuous amplitude signal to discrete time and discrete valued (from a set of finite values) so that it can be represented by finite bits and can be stored on a computer.

(Part B)

ANS:

The Nyquist sampling rate is the lowest sampling rate that can be used without having aliasing. The sampling rate for an analog signal must be at least two times the bandwidth of the signal. So, for example, an audio signal with a bandwidth of 60 Hz must be sampled at least at 120 Hz to avoid aliasing.

Q No 4:ANS:

As we are given.

$$x[n] * [h_1[n] * h_2[n]] = [x[n] * h_1[n]] * h_2[n]$$

consider

$$y[n] = [x[n] * h_1[n]] * h_2[n]$$

$$x[n] * h_1[n] = w_1[n]$$

now

$$y[n] = [x[n] * h_1[n]] * h_2[n]$$

$$y[n] = w_1[n] * h_2[n]$$

$$x[n] \rightarrow [h_1[n]] \rightarrow [h_2[n]] \rightarrow y[n]$$

now consider x that

$$w_2[n] = h_1[n] * h_2[n]$$

$$y[n] = x[n] * [h_1[n] * h_2[n]]$$

$$y[n] = x[n] * w_2[n]$$

$$x[n] \rightarrow \boxed{w_2[n]} \rightarrow y[n]$$

As both block diagrams give
the same response
so

$$L.H.S = R.H.S$$