

10. If a signal is compressed in time domain, then its spectrum in the frequency domain will be
- narrower and taller.
 - narrower but flatter.
 - wider but taller.
 - wider and flatter.
11. Using computer programs, we generally evaluate the following.
- Continuous Time Fourier Series (CTFS).
 - Continuous Time Fourier Transform (CTFT)
 - Discrete Time Fourier Transform (DTFT).
 - Fast Fourier Transform (FFT).
12. The discrete-time Fourier transform (DTFT) is
- continuous valued with period π .
 - continuous valued with period 2π .
 - discrete valued with period π .
 - discrete valued with period 2π .
13. A signal-sampler operates at 4 kHz clock frequency. An ideal anti-aliasing filter before the sampler can only have maximum cut-off frequency of less than
- 8 kHz.
 - 4 kHz.
 - 2 kHz.
 - 1 kHz.
14. Sampling rate can be lowered in digital signal processing by using a technique called
- convolution.
 - mixing.
 - interpolation.
 - decimation.
15. Autocorrelation may be used for
- filtering noise.
 - signal detection in noise.
 - signal modulation.
 - signal mixing.
16. Pink noise is characterized by a
- spectrum with similar amplitudes for the entire frequency range.
 - spectrum with random amplitudes for the entire frequency range.
 - spectrum with decreasing amplitudes at higher frequencies.
 - spectrum with increasing amplitudes at higher frequencies.
17. A signal mixer can be realized using a
- low pass filter.
 - high-pass filter.
 - divider.
 - multiplier.
18. A filter that blocks a range of mid-band frequencies is called
- low-pass filter.
 - band-pass filter.
 - notch filter.
 - high-pass filter.
19. An elliptic filter has ripple in its frequency response in
- both pass & stop bands.
 - in pass band but not in stop band.
 - neither in pass nor in stop band.
 - in stop band but not in pass band.
20. An input signal $x(t) = \cos(2\pi \times 1000t) + \sin(2\pi \times 2000t) + \cos(2\pi \times 4000t)$ was filtered and the output obtained was $y(t) = 0.01 \sin(2\pi \times 2000t) + \cos(2\pi \times 4000t)$. The filter is a
- low-pass filter.
 - high-pass filter
 - band-pass filter.
 - a band-stop filter.

KATHMANDU UNIVERSITY
End Semester Examination [C]
June/July 2024

Level : B.Sc.
Year : II
Time : 2 hrs. 30 mins.

Course : EEG 221
Semester : II
F.M. : 55

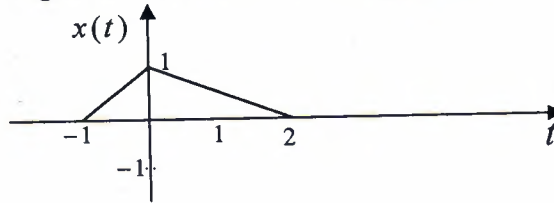
03 JUL 2024

SECTION "B"
[5 Q. × 8 = 40 marks]

Attempt ANY THREE questions.

1. a. Illustrate and state use or occurrence of following types of elementary signals in the real world. [4]
- Unit step signal
 - Rectangular pulse signal
 - Real exponential signal
 - Periodic signal

- b. A continuous time signal $x(t)$ is shown in the figure below. [4]



Sketch and label following signals derived from it.

- (i) $x(t - 1)$ (ii) $x\left(\frac{t}{2}\right)$ (iii) $x(3t)$ (iv) $\frac{dx(t)}{dt}$
2. a. What do you understand by the impulse response of a system? How can it be used to characterize a LTI system? Briefly describe the process of obtaining system response from its impulse response in case of discrete time and continuous time LTI systems. [4]
- b. Plot the given discrete time signal $x[n]$ and system impulse response $h[n]$. Determine and sketch the system output $y[n] = x[n] * h[n]$. [4]

$$x[n] = u[n] - u[n - 5] \quad \text{and} \quad h[n] = \left(\frac{3}{4}\right)^n \{u[n] - u[n - 3]\}$$

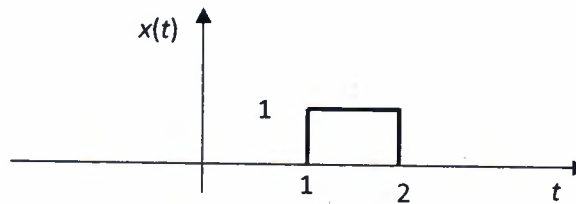
3. a. A continuous time system is represented by the following differential equation. [4]

$$\frac{dy(t)}{dt} + 2y(t) = x(t)$$

Find the output of the system $y(t)$ for the input $x(t) = 5e^{3t}u(t)$. Assume initial rest condition.

- b. How can Fourier Series coefficients of periodic signals be used in the analysis of electrical and electronic systems? Discuss the concept of line spectrum of a periodic signal and frequency response of an LTI system. [4]

4. a. Evaluate Fourier transform of the given signal and plot its magnitude spectrum. [4]



- b. Discuss the significance of the following properties of a continuous time Fourier transform in signal processing. [4]
- Time-shifting
 - Convolution
 - Multiplication or modulation
 - Time-scaling
5. a. Determine and plot the magnitude and phase spectrum of the following signal using the concept of discrete time Fourier series. [4]
- $$x[n] = 1 + 2 \sin\left(\frac{\pi}{4}\right)n + 3 \cos\left(\frac{\pi}{4}\right)n + \sin\left(\frac{\pi}{2}\right)n + \cos\left(\frac{\pi}{2}n + \frac{\pi}{4}\right)$$
- b. Describe briefly any modern electrical or electronic system that uses sampling, analog to digital conversion, and digital to analog conversion in its signal processing. Why do we need to avoid aliasing during sampling and what is the significance of Nyquist sampling rate? [4]
6. a. Distinguish between correlation, cross-correlation and autocorrelation. [2]
- b. List some of the important properties of a discrete-time random signal. [2]
- c. How does an ideal signal mixer work? [2]
- d. List basic filter types. Illustrate their ideal frequency responses. [2]