GATE 2019

SIGNAL AND SYSTEM

ELECTRICAL ENGINEERING





A Unit of ENGINEERS CAREER GROUP

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GATE-2019: Signal and System | Detailed theory with GATE & ESE previous year papers and detailed solutions.

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First Edition: 2016

Price of Book: INR 715/-

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SIGNAL AND SYSTEM

CHAPTER - 1 INTRODUCTION

1.1 SIGNAL

Signal is a function of one or more independent variables contain information about some behavior or natural phenomenon.

Example. Speech, Video, Audio, T.V. Signal, Current, Voltage, RF signal etc.

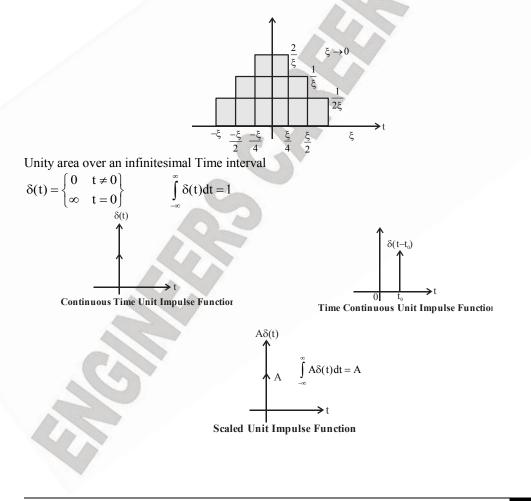
1.2 SYSTEM

Interconnection of various physical elements which are formed to get the desired response. **Example.** High Pass filter, Low pass filter, Automobile Car, Mobile, Tablet, etc.

1.3 IMPORTANT SIGNALS

1.3.1 Continuous Time Unit Impulse Signal

The unit Impulse function $\delta(t)$ is known as Dirac Delta function.





Example 1. What are even and odd fart of x(t) = $\delta(t)$ Solution. $\delta_e(t) = \frac{\delta(t) + \delta(-t)}{2}$ $\delta_0(t) = \frac{\delta(t) - \delta(-t)}{2}$ $\delta(t)$ is even function $\delta(t) = \delta(-t)$ $\delta_e(t) = \frac{2\delta(t)}{2} = \delta(t)$ $\delta_0(t) = \frac{\delta(t) - \delta(t)}{2} = 0$

Example 2. What are even and odd parts of x(t) = u(t)

Solution.

$$u_{e}(t) = \frac{u(t) + u(-t)}{2} = \frac{1}{2}$$
$$u_{0}(t) = \frac{u(t) + u(-t)}{2} = \frac{1}{2}Sgn t$$
$$u(t) = \frac{1}{2} + \frac{1}{2}Sgn t$$

Example 3. What is R.M.S value of $x(t) = A_1 \cos (\omega t + \phi_1) + A_2 \cos(\omega t + \phi_2)$ Solution. Power of signal x(t) will be

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$$P = \frac{A_1^2}{2} + \frac{A_2^2}{2}$$

R.M.S value = $\sqrt{\frac{A_1^2}{2} + \frac{A_2^2}{2}}$

Example 4. Calculate the energy and power of signal $x(n)=-(0.5)^n u(n)$ Solution.

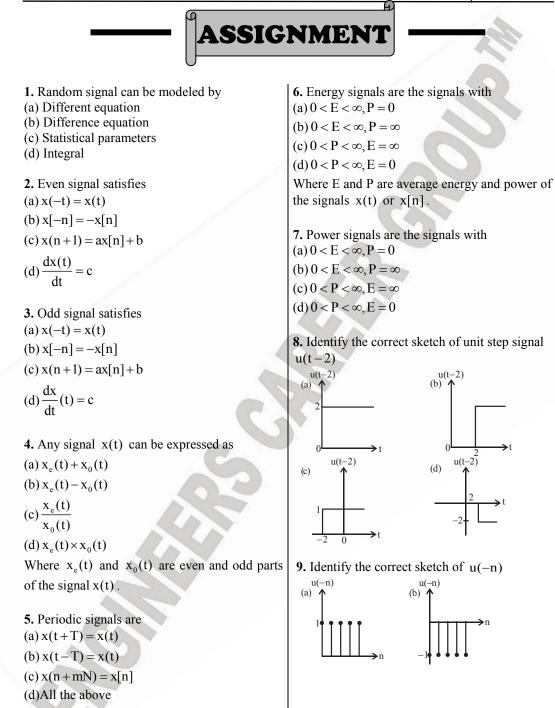
$$E = \sum_{n=-\infty}^{\infty} \left[-(0.5)^{n} u(n) \right]^{2}$$

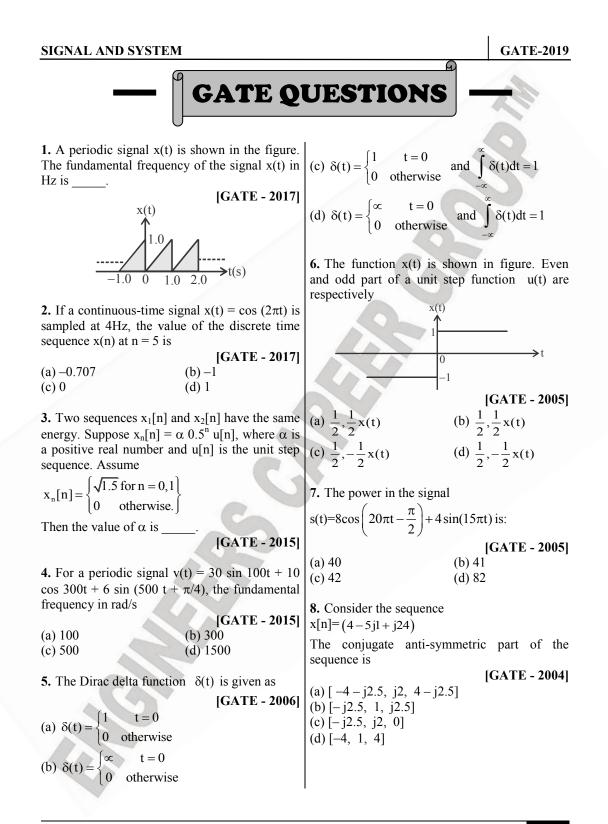
= $\sum_{n=0}^{\infty} (0.25)^{n} = \frac{1}{1-0.25} = \frac{4}{3}$
$$P = \lim_{N \to \infty} \frac{1}{2N+1} \sum_{n=-\infty}^{\infty} [x(n)]^{2}$$

= $\lim_{N \to \infty} \frac{1}{2N+1} \sum_{n=-0}^{\infty} \left(\frac{1}{4}\right)^{n}$

 $P = \frac{1}{2N+1} \cdot \frac{4}{3} = 0$

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CHAPTER - 2 SYSTEM

2.1 SYSTEM

A system is a mathematical model of physical process that relates the Inputs (or excitation) signal to the output (or response) of signal.

The response or output of system depends upon transfer function system.



Mathematically, the functional Relationship between I/P and O/P may be written as:

 $\mathbf{y}(\mathbf{t}) = \mathbf{f} \{ \mathbf{x}(\mathbf{t}) \}$

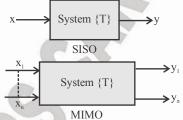
$$y(t) = T [x(f)]$$

Where T implies transformation and gives a mapping to be done on x(t) to get y(t)

2.1.1 Symbolically, we can write

 $x(t) \xrightarrow{s} y(t)$

Multiple input and/or output signals are possible. But we will restrict our attention for most part in this course to the single Input single output



Examples of system. Filters, amplifiers, communication channels, T.V. set are various example of electrical system.

2.2 TYPES OF SYSTEMS

- 1. Continuous-Time System
- 2. Discrete-Time System

1. Continuous-Time System

Continuous - Time system may be defined as also continuous. This means that Input and output of continuous time system are both continuous time signal.



Example. Audio, Video Amplifier, Power supplies etc. Simple Practical example of continuous time – system is Low Pass Filter

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GATE QUESTIONS 1. Consider a single input single output $H(f) = \begin{cases} 1, |f| \le 12kHz \\ 0, |f| > 12kHz \end{cases}$ discrete- time system with x[n] as input and y[n] as output, where the two are related as What is the number of sinusoids in the output $y[n] = \begin{cases} n \mid x[n] \mid, & \text{for } 0 \le n \le 10 \\ x[n] - x[n-1], & \text{otherwise} \end{cases}$ and their frequencies in kHz? [GATE - 2017] Which one of the following statements is true (a) Number = 1, frequency = 7 about the system? (b) Number = 3, frequencies = 2,7,11(c) Number = 2, frequencies = 2, 7[GATE - 2017] (d) Number = 2, frequencies = 7, 11(a) It is causal and stable (b) It is causal but not stable 5. An LTI system with unit sample response (c) It is not causal but stable (d) It is neither causal nor stable $h(n) = 5\delta[n] - 7\delta[n-1] + 7\delta[mn-3] - 5\delta[n-4]$ is a 2. Consider and LTI system with magnitude [GATE - 2017] response $|H(f)| = \begin{cases} 1 - \frac{|f|}{20}, & |f| \le 20\\ 0, & |f| > 20 \end{cases}$ and phase (a) Low pass filter (b) High pass filter (c) Band pass filter (d) Band stop filter 6. The input x(t) and the output y(t) of a response Arg[H(f)]=2f. continuous time system are related as If the input to the system is $y(t) = \int_{t-T}^{t} x(u) du$. The system is $x(t) = 8\cos\left(20\pi t + \frac{\pi}{4}\right) + 16\sin\left(40\pi t + \frac{\pi}{8}\right)$ [GATE - 2017] $+24\cos\left(80\pi t+\frac{\pi}{16}\right)$ (a) Linear and time variant (b) Linear and time invariant Then the average power of the output signal y(t)(c) Non Linear and time variant (d) Nonlinear and time invariant 15 [GATE - 2017] 7. Consider $g(t) = \begin{cases} t - \lfloor t \rfloor, t \ge 0 \\ t - \lceil t \rceil$, otherwise \lbrace , \rbrace 3. The transfer function of a causal LTI system is H9s) = 1/s. if the input to the system is x(t) =Where $t \in R$ $[\sin(t)/\pi t]u(t)$; where u(t) is a unit step function. Here, |t| represent the largest integer less than The system output y(t) as $t \to \infty$ is [GATE - 2017] or equal to t and [t] denotes the smallest integer greater than or equal to t. The coefficient 4. The signal $x(t) = sin(1400\pi t)$, where t is in of the second harmonic component of the seconds, is sampled at a rate of 9000 samples fourier series representing g(t) is per second. The sampled signal is the input to [GATE - 2017] an ideal lowpass filter with frequency response H(f) as follows: 8. Consider the signal $x(t) = \cos(6\pi t) + \sin(8\pi t)$, where t is in seconds. The Nyquist sampling

CHAPTER - 3 *LINEAR TIME-INVARIANT SYSTEM*

3.1 INTRODUCTION

A system is a mathematical model of physical process that relates the Inputs (or excitation) signal to the output (or response) of signal. We have discussed the Several Basic Properties of systems, two properties namely Linearity and Time - invariance plays a important role in analysis of signals and system. If a system has both linearity and time - invariance properties, then this system is called Linear -Time Invariant system (LTI system).



We study LTI system because of the fact that the most of practical and physical system can be modeled in form of Linear-Time (invariant) System

In this chapter we develop the fundamental Input-Output relationships for system having these properties and it will be shown that Input-output Relationship for LTI system is described of a convolution operation.

Importance of convolution operation if one knows the output of unit Impulse then output for general input can be calculated.

3.2 CHARACTERISTICS OF LINEAR TIME - INVARIANT(LTI) SYSTEM

Both continuous-time and discrete-time linear time invariant (LTI) system exhibit one important characteristics that the superposition theorem can be applied to find the response y(t) to a given input x(t).

3.2.1 Important steps to adopted to find response of LTI system using superposition

1. Resolve the input function x(t) in terms of simple or basic function like impulse function for which response can be easily evaluated.

2. Determine Response of LTI system for simple or Basic functional individually.

3. Using superposition theorem, find the sum of individual response which will become overall response y(t) of function x(t) from above, to find the response of LTI system to given function first we have to find the response of LTI system to an unit impulse called as unit impulse response of LTI system.

3.2.2 Unit Impulse Response [h(t)_n or h/n]

Impulse response of continuous time or discrete-time LTI system is output of system due to an unit impulse input applied at time t = 0 or n = 0.

[GATE - 2014]

[GATE - 2014]

[GATE - 2014]



(b) Differentiating the unit step response (c) Integrating the unit ramp response

6. For linear tune invariant systems, that are

Bounded Input Bounded Output stable, winch one of the following statements is TRUE?

(a) The impulse response will be integrable, but

(b) The unit impulse response will have finite

(c) The unit step response will be absolutely

If the input to the system is $\cos(3t)$ and the steady state output is A $sin(3t + \alpha)$, then the

(b) 1/15

(d) 4/3

(d) The unit step response will be bounded.

(d) Integrating the unit step response

may not be absolutely integrable.

support.

integrable.

value of A is

input, x(t), is given by

(a) 1/30

(c) 3/4

1. Let the input be u and the output be y of a (a) Differentiating the unit ramp response system, and the other parameters are real constants. Identify which among the following systems is not a linear system: [GATE - 2018]

- (a) $\frac{d^3y}{dt^2} + a_1 \frac{d^2y}{dt^2} + a_2 \frac{dy}{dt} + a_3 y$ $= b_3 u + b_2 \frac{du}{dt} + b_1 \frac{d^2 u}{dt^2}$
- (b) $y(t) = \int_{0}^{t} e^{\alpha(t-\tau)} \beta u(\tau) d\tau$ (c) y = au + b, $b \neq 0$

(d) y = a u

(a) $x(t+t_0)$

(c) $x(-t+t_0)$

be obtained by

2. Let $z(t) = x(t)^*y(t)$. Where "*" denotes convolution. Let c be a positive real-valued 7. Consider an LTI system with transfer constant. function $H(s) = \frac{1}{s(s+4)}$ Choose the correct expression for z(ct).

$$\begin{array}{c} [GATE - 2017] \\ (a) c.x(ct)*y(ct) \\ (c) c.x(t)*y(ct) \\ \end{array} \\ \begin{array}{c} (b) x(ct)*y(ct) \\ (d) c.x(ct)*y(t) \\ \end{array} \\ \begin{array}{c} (d) c.x(ct)*y(t) \\ \end{array} \\ \end{array}$$

3. Consider the system with following inputoputput relation $y[n] = (1 + (-1)^n)x[n]$ Where x[n] is the input and y[n] is the output. The system is

8. Consider an LTI system with impulse **IGATE - 2017** response $h(t) = e^{-5t} u(t)$. If the output of the system is $y(t) = e^{-3t} u(t) = -e^{-5t} u(t)$ then the

(a) Invertible and time invariant (b) Invertible and time varying (c) Non-invertible and time invariant (d) Non-invertible and time varying (a) $e^{-3t}u(t)$ (b) $2e^{-3t}u(t)$ 4. The result of the convolution (c) $e^{-5t}u(t)$ $x(-t) * \delta(-t-t_0)$ is

5. The impulse response of an LTI system can

[GATE - 2015] (b) $x(t-t_0)$ (d) $x(-t-t_0)$

9. Two systems with impulse responses $h_1(t)$ and $h_2(t)$ are connected in cascade. Then the overall impulse response of the cascaded system

(d) $2e^{-5t}u(t)$

[GATE - 2013]

(a) Product of $h_1(t)$ and $h_2(t)$ [GATE - 2015]

is given by

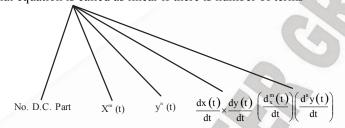
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CHAPTER - 4 *LINEAR-TIME INVARIANT-2 SYSTEM*

4.1 LINEAR CONSTANT CO-EFFICIENT DIFFERENTIAL EQUATIONS (LCC DE) A general nth -order linear - constant co-efficient differential equation is given by

 $\sum_{K=0}^{N} a_{k} \frac{d^{k} y(t)}{dt^{k}} = \sum_{K=0}^{M} b_{K} \frac{dk_{x}(t)}{dt^{k}}$

Above equation is representation of continuous system. 1. A differential equation is called as linear if there is number of terms



Example. $\frac{d^2y(t)}{dt^2} + 5\frac{dy(t)}{dt} + 5 = x(t)$

It is Non-Linear differential equation because a d.c. term present in it.

2. Differential equation is said to be time-invariant if all the co-efficient of differential equation are const.

So,
$$\frac{a_n d^n y(t)}{dt^n} + a_{n-1} \frac{d^{n-1} y(t)}{dt^{n-1}} + \dots y(t) = b_0 x(t)$$

Represents the linear constant co-efficient differential equation.



LCCDE equation is used to analyse the LTI system or we can say if any system is LTI system then it can be represented in differential equation by LCCDE

4.1.1 System described by Difference Equations

The role of differential equation in describing continuous-time system is played by difference equations for discrete-time system.

4.2 LINEAR CONSTANT COEFFICIENT DIFFERENCE EQUATIONS (LCCDE)

The discrete time counterpart of general differential equation is the nth order linear cost. Coefficient difference equation is given by

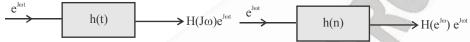
$$\sum_{k=0}^{N} a_{k} y(n-k) = \sum_{k=0}^{M} b_{k} x(n-k)$$

CHAPTER - 5 CONTINUOUS-TIME FOURIER SERIES

5.1 INTRODUCTION

In this chapter we explore an alternative representation of signals and LTI system. Here we represent signal as linear combination of complex exponentials.

The complex exponential in study of LTI system is important from the fact that the response of an LTI system to a complex exponential input is same complex exponential with only change in amplitude and phase.



Till now all signals are draw with respect to time.

That means 't' was considered as variable. The representation of signal w.r.t. to time is called time domain representation.

Since the time domain representation of signal is net sufficient for its analysis. For sake of analysis we will use the frequency domain representation.

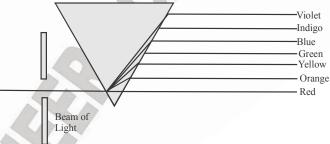
In frequency domain representation the variable is frequency 'f' rather than 't'.

The signal represented in frequency domain is called as "spectrum" the spectrum consists two graphs.

1. Amplitude spectrum occurs in $|M(j\omega)|Vs\omega$

2. Phase spectrum occurs in $|\phi(j\omega)|$ Vs ω

Example. Why frequency domain representations is important? **Solution.**



Now when we pass Beam of white light through a prism, we get band of colours produced by prism. It means that light Beam passing through a prism analysed it into its colour component without any change. The output colour can be normally specify as spectrum or analysis of light into color is nothing, but frequency analysis.

For analysis the signal, the various parameters are evaluated as

- 1. Amplitude
- 2. Frequency context
- 3. Power and energy densities

4. Periodicity

Periodicity can be obtained in time-domain but frequency content is not evaluated in time-do main. Hence these signal has to be transformed in frequency domain with the help of fourier series,

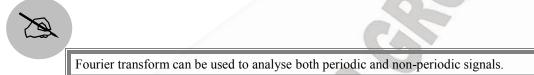
CHAPTER - 6 FOURIER TRANSFORM

6.1 INTRODUCTION

In previous chapter we have discussed the Fourier series which is tool used to analyse a periodictime signal in frequency domain.

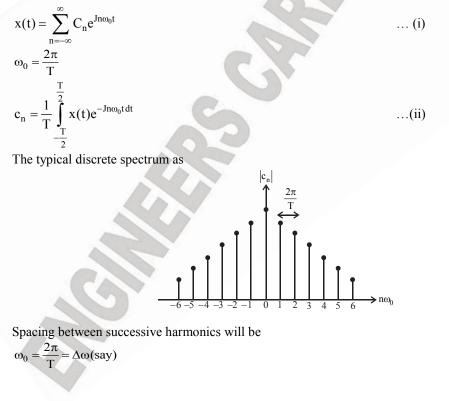
Disadvantage of Fourier series is it cannot analyse the non-periodic signals.

So Fourier develops a new tool to analyse the non-periodic or a periodic signal in frequency domain known a Fourier transform.



6.2 ANALYSIS OF NON-PERIODIC FUNCTION OVER ENTIRE INTERVAL

A non-periodic signal may assume as limiting case of periodic signal where the period of signal approaches infinity. Such a signal form by replacing fundamental time period $T \rightarrow \infty$ let us consider a periodic function x(t) having period T. The complex Fourier series representation of function may be written.



ESE OBJ QUESTIONS 2. Finite Number of discontinuities within finite 1. The Fourier Transform of $e^{-\frac{x^2}{2}}$ is time interval t [EE ESE - 2018] 3. Infinite number of discontinuities (a) $\frac{1}{2} \cdot e^{-\frac{\omega^2}{2}}$ Select the correct answer using the codes given (b) $e^{-\frac{a}{2}}$ below: [EC ESE - 2017] (c) $\frac{\pi}{2}$ (a) 1, 2 and 3 (b) 1 and 3 only (d) $\sqrt{\pi}$ (c) 1 and 2 only (d) 2 and 3 only **2.** Fourier series of any periodic signal x(t) can 5. The fourier transform of a rectangular pulse be obtained if is [EC ESE - 2012] $\int |\mathbf{x}(t)| dt < \infty$ 1. (a) Another rectangular pulse (b) Triangular pulse 2. Finite number of discontinuities within finite (c) sinc function time interval t (d) Impulse function 3. Infinite number of discontinuities Select the correct answer using the codes given 6. Which one of the following is Dirichelt below: condition? [EE ESE - 2017] [EC ESE - 2010] (a) 1, 2 and 3 (b) 1, and 3 only (c) 1 and 2 only (d) 2 and 3 only (a) $\int |x(t)| < \infty$ 3. The laplace transform of the below function (b)Signal x(t) must have a finite number of is maxima and minima in the expansion interval x(t) (c)x(t) can have an infinite number of finite discontinuities in the expansion interval $(d)x^{2}(t)$ must be absolutely summable 7. If f(t) is an even function, then what is its [EE ESE - 2017] Fourier transform $F(j\omega)$? $2 \sin \omega$ [EC ESE - 2008] (a) $\omega \sin \omega$ (b)(a) $\int_{0}^{\infty} f(t) \cos(2\omega t) dt$ (c) $\frac{\omega}{\sin \omega}$ (d)(b) $2\int_{0}^{\infty} f(t)\cos(\omega t)dt$ (c) $2\int_0^{\infty} f(t)\sin(\omega t)dt$ 4. Fourier series of any periodic signal x(t) can be obtained if (d) $2\int_0^{\infty} f(t)\sin(2\omega t)dt$ $|\mathbf{x}(t)| dt < \infty$

SIGNAL AND SYSTEM

CHAPTER - 7 CORRELATION AND FILTERING ACTION

7.1 CORRELATION

Correlation is used to find similarity between two signals. There are two type of correlation

1. Auto-Correlation

It is used to find similarity between two same signals

2. Cross-Correlation

It is used to find similarity between different signal.

7.1.1 Auto-Correlation

7.1.1.1 Autocorrelation Function

It gives the measure of similarity, match or coherence between a signal and a delayed function. A signal may be energy signal or power signal.

7.1.1.2 Autocorrelation Function of Energy Signal

Autocorrelation function of this signal may be obtained by integrating the product of signal x(t) and delayed version of its.

$$R_{xx}(\tau) = \int_{-\infty}^{\infty} x(t) x(t-\tau) dt$$

Where τ is called is searching parameter.

7.1.1.3 Relationship between Auto Correlation and Convolution $P_{x_{1}}(x_{2}) = x_{1}(x_{2}) \otimes x_{2}(x_{2})$

 $\mathbf{R}_{xx}(\tau) = x(t) \otimes x(-t) |_{\text{Replace } t=\tau}$

7.1.1.4 Properties of Autocorrelation Function

1. Auto correlation is an even function.

 $\mathbf{R}_{xx}(\tau) = \mathbf{R}_{xx}(-\tau)$

2. It τ is increased in either direction, the auto correlation reduces, As τ reduces auto correlation increase and it maximum at $\tau = 0$ i.e. at origin mathematically,

$$R_{xx}(\tau) < R_{xx}(0)$$
 for all τ

and $\lim R_{xx}(\tau) = 0$

3. Autocorrelation function at $\tau = 0$ gives energy of signal

i.e.
$$R_{xx}(\tau) = \int x(t)x(t-\tau)dt$$

Substituting $\tau = 0$

$$R_{xx}(0) = \int x^2(t) dt$$

 $R_{xx}(0) = Energy of signal$

CHAPTER - 8 LAPLACE TRANSFORM

8.1 INTRODUCTION

In previous chapters, we have seen the tools such as Fourier series and Fourier Transform to analyse the signals. Now the Laplace Transform is another mathematical tool which is used for analysis of signals and system. Infect, Laplace Transform provides broader characterization of signal and systems compared to Fourier Transform.

The Laplace Transform can be used where Fourier Transform cannot be used.

Laplace Transform can be used for analysis of unstable systems whereas Fourier Transform has several limitation.

Example.

for given $x(t) = e^{3t.U(t)}$



8.1.1 Definition of Laplace Transform

For general continuous time signal x(t) The Laplace Transform x(s) is defined as.

$$\mathbf{x}(\mathbf{s}) = \int_{-\infty}^{\infty} \mathbf{x}(t) \mathbf{e}^{-\mathbf{s}t} dt$$

Where s is generally a complex variable and is expressed as

 $S=\sigma+J\omega$

Also known as complex frequency

Where σ is real part and ω is imaginary part.

For convenience we will sometime denotes the Laplace Transform in operator form al $\angle [x(t)]$ and denote the transform relationship between x(t) and x(s)

As $x(t) \xleftarrow{L.T} x(s)$

e^{-st} is kernel of function

It may be noted that integration is taken from 0 to ∞ . Therefore, this is called Bilateral Laplace Transform.

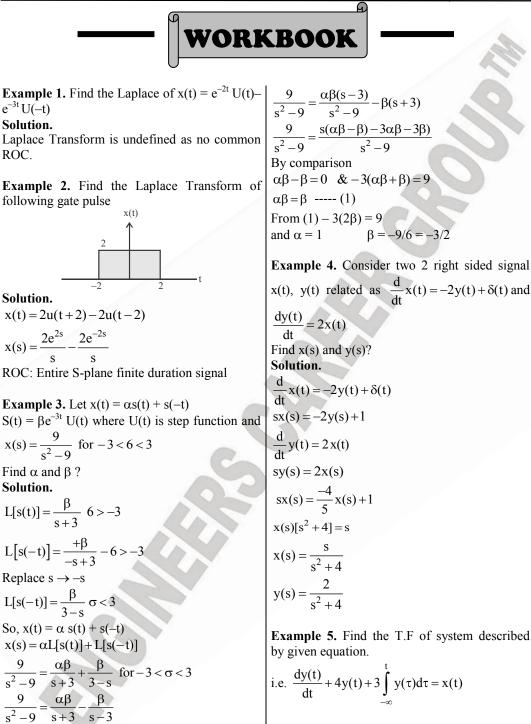
Similar if x(t) is zero for t < 0

Then Laplace may be defined as

$$\mathbf{x}(s) = \int_{0}^{\infty} \mathbf{x}(t) e^{-st} at$$

Where $s = \sigma + J\omega$

Integration is taken from 0 to ∞ . This is called as unilateral/one-sided Laplace Transform.



 Laplace transform ana (a) Solving integral differ (b) Converts differential equation (c) Converts integral execution (d) All the above 	rential equation. equation into algebraic	8. $x(t) = \cos\omega_0 t u(t)$ (a) $\frac{s}{s^2 + \omega_0^2}$ (c) $\frac{\omega_0}{s^2 + \omega_0^2}$	(b) $\frac{1}{s^2 + \omega_0^2}$ (d) $\frac{\omega_0}{s^2}$		
2. Laplace transform x(s	s) of signal x(t) is	9. Laplace transform of s with $x_2(t)$ is	ignal $x_1(t)$ convolving		
(a) $\int_{-\infty}^{\infty} x(t) e^{-st} dt$	$(b)\int_{-\infty}^{\infty} x(t)e^{-st}dt$	(a) $x_1(s) * x_2(s)$	(b) $x_1(s)x_2(s)$		
(c) $\int_{-\infty}^{\infty} e^{-st} dt$	$(d)\int_{-\infty}^{\infty}x(t)e^{-st}dt$	(c) $x_1(s) / x_2(s)$	(d) $x_1(t) x_2(t)$		
 3. Bilateral and unilate differs in terms of (a) Lower limit of integra (b) Upper limit of integra (c) They are same (d) Bilateral transform do 4. Laplace transform of 	ation ation bes not exist.	 10. For causal continuous-time LTI system, ROC is in the (a) Left of all system poles (b) Right of all system poles (c) Right of all zeros (d) Left of all zeros 11. If the system is causal and stable, the system poles must lie 			
(a) $\frac{1}{s}$ (c) 1	(b) s (d) s^2	 (a) On the jω axis (b) On the left half of s-pl (c) On the right half of s-j (d) Both (a) and (c) 			
5. Laplace transform of	$\mathbf{x}(\mathbf{t}) = \mathbf{t}$ is	(d) Doth (d) and (c)			
(a) $\frac{2}{s^2}$	(b) $\frac{1}{s^2}$	12. Laplace transform of			
(c) s^2	(d) $\frac{1}{s}$	(a) sx(s)	$(b)\frac{x(s)}{s}-x(0^{-})$		
6. Region of the converg (a) Zeros	(b) Poles	 (c) sx(s) - x(0⁻) 13. Laplace transform of 	$(d) x(s) - x(0^{-})$		
(c) No zeros	(d) No pole		$\int_0^{\infty} x(t) dt$ is		
7. Inverse Laplace transf	form of $\frac{1}{(s+a)^2}$ is	(a) $\frac{1}{s}x(s)$	(b) sx(s)		
(a) tu(t)	(b) $te^{-at}u(t)$	(c) $s^2 x(s)$	(d) $x(s)/s^2$		
$(c) e^{-at} u(t)$	(d) $ae^{-at}u(t)$				

GATE-2019

The transfer

[GATE - 2015]

1. The solution of the differential equation

$$\frac{d^2y}{dt^2} + 2\frac{dy}{dt} + y = 0 \text{ with } y(0) = y'(0) = 1 \text{ is}$$
[GATE - 2015]
(a) (2-t)e^t (b) (1+2t)e^{-t}
(c) (2+t)e^{-t} (d) (1-2t)e^t

(a) $2-e^{-0.2t}$ (c) $50-49e^{-0.2t}$

for a unit step input is given as

 $y(t) = 1 - \frac{2}{\sqrt{3}}e^{-t}\cos\left(\sqrt{3t} - \frac{1}{\sqrt{3}}e^{-t}\right)$

function of the system is

(c) $\frac{3}{s^2 + 2s + 3}$

GATE QUESTIONS

(b) $2 - e^{0.2t}$ (d) 50-49e

6. The output of a standard second-order system

2. The bilateral Laplace transform of a function 1if $a \le t \le b$

$$f(t) = \begin{cases} 0 \text{ otherwise} \end{cases}$$
 is

[GATE - 2015]
(a)
$$\frac{a-b}{s}$$
 (b) $\frac{e^{z}(a-b)}{s}$ (c) $\frac{e^{-az}-e^{-bz}}{s}$ (d) $\frac{e^{z(a-b)}}{s}$ (e) $\frac{e^{z(a-b)}}{s}$ (f) $\frac{e^{z(a-b)}}{s}$ (f) $\frac{3}{s^{2}+2s+3}$

3. Let the signal f(t) = 0 outside the interval $[T_1, T_2]$, where T_1 and T_2 are finite. Furthermore, $|f(t)| < \infty$. The region of convergence (ROC) of the signal's bilateral Laplace transform F(s) is

(d)
$$\frac{4}{1}$$

7. Input x(t) and output y(t) of an LTI system are related by the differential equation y''(t)-y'(t)-6y(t) = x(t)/. If the system is neither causal nor stable, the impulse response h(t) of the system is

[GATE - 2015]

(b) A parallel strip no containing the j Ω axis (c) The entire s-plane

(a) A parallel strip containing the j Ω axis

(d) A half plane containing the j Ω axis

4. Let $x(t) = \alpha s(t) + s(-t)$ with $s(t) = \beta e^{-4t} u(t)$. where u(t) is unit step function. If the bilateral Laplace transform of x(t) is

$$X(s) = \frac{16}{s^2 - 16} - 4 < \operatorname{Re}\{s\} < 4;$$

Then the value of β is

[GATE - 2015]

5. Consider the differential equation $\frac{dx}{dt} = 10 - 0.2x$ with initial condition x(0) = 1. The response x(t) for t > 0 is [GATE - 2015]

(a)
$$\frac{1}{5}e^{3t}u(-t) + \frac{1}{5}e^{-2t}u(-t)$$

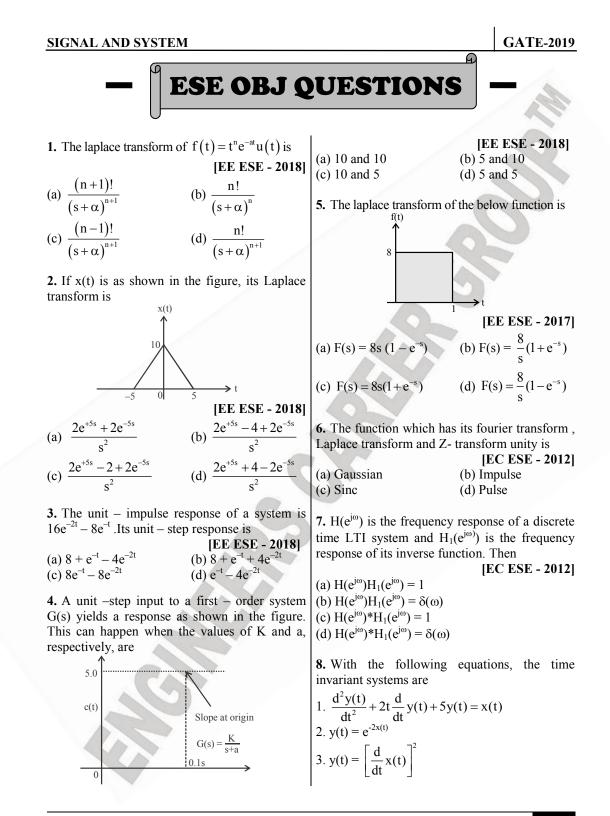
(b) $-\frac{1}{5}e^{3t}u(-t) + \frac{1}{5}e^{-2t}u(-t)$
(c) $\frac{1}{5}e^{3t}u(-t) - \frac{1}{5}e^{-2t}u(t)$
(d) $-\frac{1}{5}e^{3t}u(-t) - \frac{1}{5}e^{-2t}u(t)$

8. Consider the function $g(t) = e^{-t} \sin(2\pi t)u(t)$ where u(t) is the unit step function. The area under g(t) is _

[GATE - 2015]

9. The stable linear time invariant (LTI) system has a transfer function H(s) = $\frac{1}{s^2 + s - 6}$

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CHAPTER - 9 Z-TRANSFORM

9.1 INTRODUCTION

Z-Transform which is discrete-time counterpart of Laplace Transform.

It may be observed that Laplace Transform is an extension of continuous-time Fourier. Transform because of fact that Laplace Transform may be applied to broader class of signals than Fourier Transform. Just for instances, there are several signal s for which the Fourier transform does not converges but Laplace Transform converses.

Similarly Z-Transform is introduced to represent discrete-time sequences in Z-domain (Z is complex variable). Also to analyse the difference equations that describes the linear time-invariant (LTI systems) and converts into algebraic equation. Thus simplifying further analysis.

In general Z-Transform of discrete signal x(n) is expressed as

$$x(z) = \sum_{n=-\infty}^{\infty} x(n) z^{-n}$$

Generally denoted a

$$\mathbf{x}(\mathbf{z}) = \mathbf{z}[\mathbf{x}(\mathbf{n})]$$

x(n) & x(z) forma Z-Transform pair

$$x(n) \leftarrow 7 \rightarrow x(z)$$

It may be noted that Z-Transform is an infinite power series. It may exists only for those values of Z for which series converges.

x(z) is a complex number and a function of complex variable Z.

In polar form, $Z = re^{J\omega}$

With r gives magnitude of z, |z|

 ω gives phase of z, $\angle z$

So,
$$x(re^{J\omega}) = \sum_{n=-\infty}^{\infty} x(n) \left(re^{-J\omega}\right)^n$$

$$\mathbf{x}(\mathbf{r}\mathbf{e}^{\mathbf{J}\omega}) = \sum_{n=-\infty}^{\infty} \left[\mathbf{x}(n)\mathbf{r}^{-n}\right] \mathbf{e}^{-\mathbf{J}\omega n}$$

We see that $x(re^{J\omega})$ is discrete time Fourier Transform of sequence x(n) multiplied by real exponential r^{-n}

i.e.,
$$x(re^{j\omega}) = F |x(n)r^{-n}|$$

The exponential weighting r^{-n} may decaying or growing with increasing n depended on whether it is greater than or less than unity.

Now if r = 1 or |z| = 1

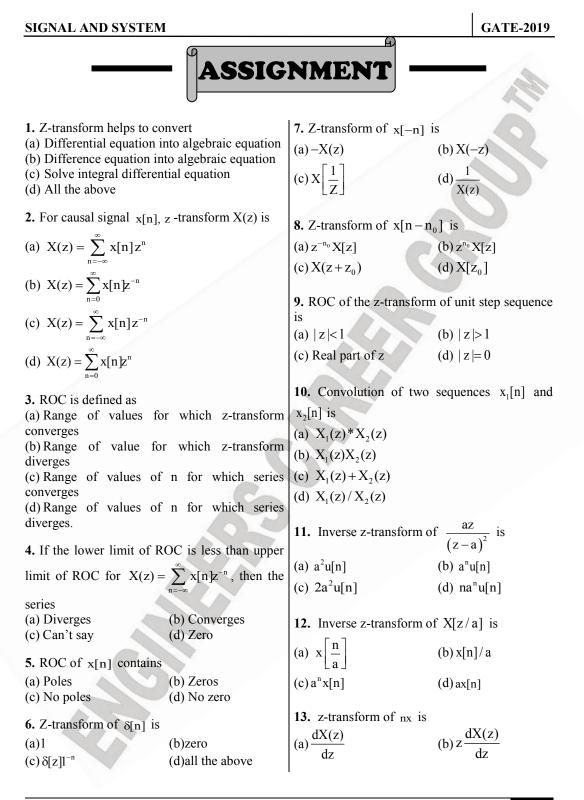
The expression thus reduces to discrete Fourier Transform of input sequence.

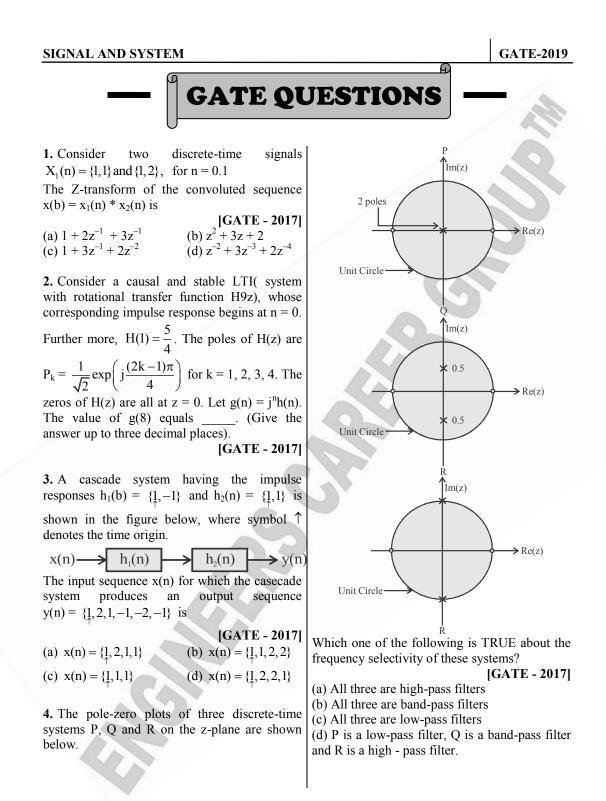
 $\mathbf{x}(\mathbf{z})\Big|_{\mathbf{z}=e^{j\omega}} = \mathbf{x}(e^{j\omega}) = \text{DTFT}[\mathbf{x}(n)]$

...(i)

9.2 Z-PLANE OR Z-DOMAIN

Here we transforming discrete time sequence x(n) into x(z)Where $Z = re^{J\omega}$





CHAPTER - 10 DISCRETE TIME FOURIER TRANSFORM

10.1 INTRODUCTION

Basically the Fourier Transform of periodic finite energy signal is called DTFT mathematical.

$$x(e^{J\omega}) = \sum_{n=-\infty}^{\infty} x(n)e^{-J\omega n}$$

or also denoted as $x(\Omega)$ Where Ω is discrete frequency

10.1.1 Periodic Nature of DTFT

Since Ω is discrete frequency. Then Substituting $\omega = \omega + 2\pi k$

$$\begin{split} x \Big[e^{J(\omega + 2\pi k)} \Big] &= \sum_{n = -\infty}^{\infty} x(n) e^{-J(\omega + 2\pi k)n} \\ &= \sum_{n = -\infty}^{\infty} x(n) e^{-J\omega n} e^{-J2\pi kn} \end{split}$$

By using Euler's identity

$$e^{-J2\pi kn} = \cos(2\pi kn) - J\sin(2\pi kn)$$

$$= 1 - J0$$

n=-~

So,
$$x\left(e^{J(\omega+2\pi k)}\right) = \sum_{n=-\infty}^{\infty} x(n)e^{-J\omega n}$$

So, $x\left(e^{J(\omega+2\pi k)}\right) = x\left(e^{J\omega}\right)$

or

 $x(\Omega + 2\pi k) = x(\Omega)$

Thus DTFT is periodic nature with a period of 2π . We DTFT is restricted to 0 to 2π or $-\pi$ to π .

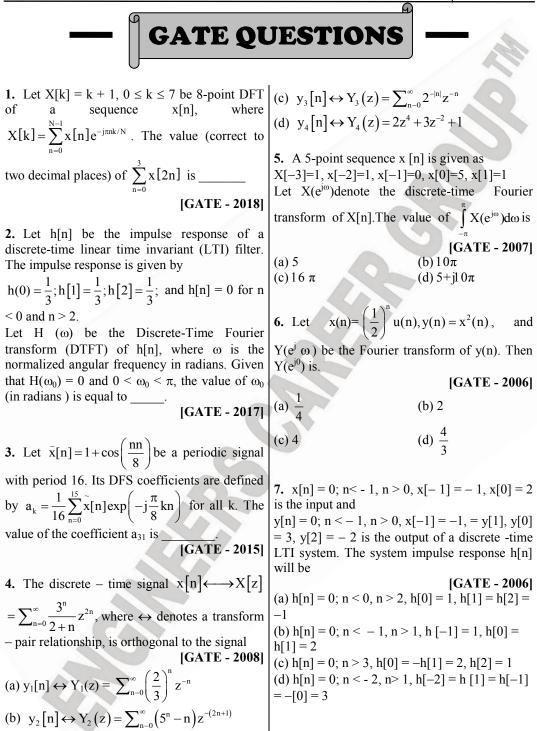
Z

DTFT is continuous frequency Ranging from - ∞ to ∞ because of a periodic time function.

Inverse discrete time Fourier Transform:

$$\mathbf{x}(\mathbf{n}) = \frac{1}{2\pi} \int_{-\pi}^{\pi} \mathbf{x}(\mathbf{e}^{\mathbf{J}\omega}) \mathbf{e}^{\mathbf{J}\omega\mathbf{n}} d\omega$$

Thus we can say x(n) and $x(e^{J\omega})$ form a Fourier Transform pair.



CHAPTER - 11 DISCRETE TIME FOURIER SERIES

11.1 INTRODUCTION

Discrete time Fourier series is used to analyses the discrete periodic signals.

A discrete time x(n) is said to periodic is there is smallest positive integer 'N' for which it is satisfied.

x(n + n) = x(n) for all 'n'

11.1.1 Discrete Fourier Series Representation

The discrete Fourier series Representation of periodic sequence x(n) with fundamental time period N is given by

$$x(n) = \sum_{K=0}^{N-1} c_k e^{\frac{J2\pi nk}{N}}$$

 $(K = 0, 1, 2, \dots, N-1)$

The FS representation of x(n) consists of N harmonically related exponential function. $_{J2\pi kn}$

 e^{N} K = 0, 1, 2 N - 1

Where c_k is the Fourier Series co-efficient. It is given by

$$c_{k} = \frac{1}{N} \sum_{n=0}^{N-1} x(n) e^{\frac{-2\pi kn}{N}}$$

Here input sequence x(n), FS coefficient c_k both are periodic.

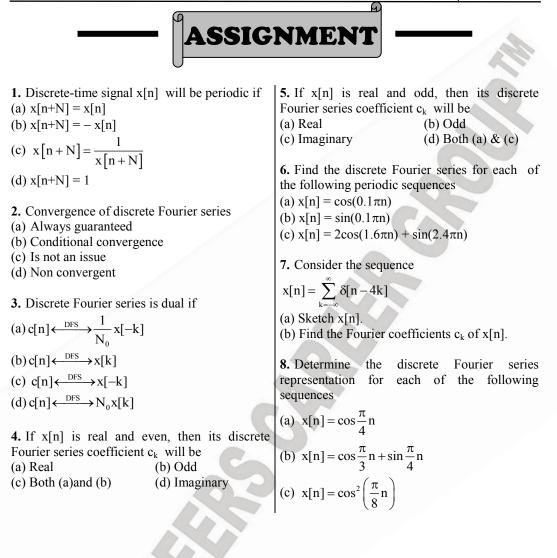
11.1.2 Comparison between Continuous Time Fourier Series and Discrete Time Fourier Series

CTFS	DTFS
n=	$x(n) = \sum_{K=0}^{N-1} c_k e^{\frac{J2\pi kn}{N}}$
$c_{n} = \frac{1}{T_{0}} \int_{0}^{T_{0}} x(t) e^{-Jn\omega_{0}t}$	$c_{k} = \frac{1}{N} \sum_{n=0}^{N-1} x(n) e^{\frac{-J2\pi nk}{N}}$
C _n is discrete and periodic	C _n is Periodic and Discrete

11.1.3 Convergence of Discrete Fourier Series

Since x(n) is discrete Fourier Series is a Finite series because summation limits are from K = 0 to N - 1. So, I_n compassion to continuous-time case, there is no convergence issue with discrete Fourier series.

11.1.4 Discrete Fourier Series of Arbitrary Periodic Sequence x(n) W define Twiddle factor) Phase factor



CHAPTER - 12 DISCRETE AND FAST FOURIER TRANSFORM

...(i)

12.1 INTRODUCTION

In previous chapter we have mainly studied about signal analysis in frequency domain by Fourier Tools and for a discrete time sequence the Fourier Tool used is generally a discrete time Fourier Transform.

$$x\left(e^{J\omega}\right) = \sum_{n=-\infty}^{\infty} x\left(n\right)e^{-J\omega n}$$

We know that $x(e^{J\omega})$ is Fourier Transform of discrete time signal x(n).

The frequency analysis of discrete-time signals are usually and most conventional performed on digital computer. To perform frequency analysis of discrete time x(n). We convert time-domain sequence to an equivalent frequency-domain representation such a representation is given by Fourier Transform $x(e^{J\omega})$ or $H(\omega)$.

Since $x(\omega)$ is continuous function of frequency ' ω '. The range of ω is from 0 to 2π or $-\pi$ to π .

Since this calculation is not possible to computer $x(\omega)$ on digital computer because range of summation (Equation (i)) is from " $-\infty$ to ∞ ".

So, if we make Range finite then it is possible to do these calculation on digital computed.

12.2 FREQUENCY DOMAIN SAMPLING AND RECONSTRUCTION OF DISCRETE-TIME SIGNALS

We recall that a periodic signals have continuous spectrum. If we consider such an a periodic discrete-time sequence x(n) with Fourier Transform.

$$x\left(e^{J\omega}\right) = \sum_{n=-\infty}^{\infty} x\left(n\right)e^{-J\omega n}$$

But DTFT is continuous in nature and periodic with period of 2π and unique information in frequency domain in only one period i.e. "0 to 2π " and DTFT range of from " $-\infty$ to ∞ ".

A finite range of sequence is obtained by extracting a particular portion from infinite sequence [x(n)] now $x(\omega)$ is continuous and periodic.

A discrete finite sequence is obtained by sampling $x(\omega)$ periodically in frequency at specing $D\omega$ radians between two samples.

For uniqueness of information only samples in fundamental period is necessary.

For convenience, we take N equidistant samples in interval $0 < \omega \le 2\pi$.

Total Range = 2π (one period) Total samples = N

 $D\omega$ (spacing between samples) = $\frac{2}{3}$

GATE QUESTIONS

1. The DFT coefficient out of five DFT Let $X_1(k)$ and $X_2(k)$ be 4-point DFTs of $x_1[n]$ coefficients of a five - point real sequence are and $x_2[n]$, respectively. given as: X(0) = 4, X(1) = 1 - j1 and X(3) = 2 + j1Another sequence x₃[n] is derived by taking 4j2. The zero - the value of the sequence point inverse DFT of $X_3(k)=X_1(k)X_2(k)$. x(n)x(0) is. The value of $x_3[2]$ is _____. [GATE - 2015] [GATE - 2017] (a) 1 (b) 2 (c) 3 (d) 4 5. The N – point DFT X of sequence $x[n], 0 \le n$ \leq N – 1 is given by 2. The Discrete Fourier Transform (DFT) of the $X[k] = \frac{1}{\sqrt{N}} \sum_{n=0}^{N-1} x[n] e^{-j\frac{2\pi}{N}nk}, \quad 0 \le k \le N-1.$ 4-point sequence $X[n] = {x[0], x[1], x[2], x[3]} = {3, 2, 3, 4}$ is Denote this relation as X = DFT(x). For N = 4, $X[k] = {X[0], X[1], X[2], X[3]} = {12, 2j, 0,$ which one of the following sequence satisfies 2j} DFT (DFT(x)) = x?If $X_1[k]$ is the DFT of the 12-point sequence [GATE - 2014] $x_1[n] = \{3, 0, 0, 2, 0, 0, 3, 0, 0, 4, 0, 0\},\$ (a) x = [1 2 3 4](b) x = [1 2 3 2]The value of $\left| \frac{\mathbf{x}_1[8]}{\mathbf{X}_1[11]} \right|$ is _____. (d) x = [1 2 2 3](c) x = [1 3 2 2]6. The DFT of a vector [a b c d] is the vector $[\alpha]$ [GATE - 2016] $\beta \lambda \delta$]. Consider the product **3.** Two sequence $[pqrs] = [abcd] \begin{vmatrix} a & b & c & d \\ d & a & b & c \\ c & d & a & b \end{vmatrix}.$ [a, b, c] and [A, B, C] are related as, [A]B = 1 $W_3^{-1} W_3^{-2}$ b where $W_3 = e^{j\frac{27}{3}}$ The DFT of the vector [p q r s] is scaled version $1 W_{3}^{-2} W_{3}^{-4}$ C of If another sequence [p.q.r] is derived as, [GATE - 2013] $\begin{bmatrix} 1 & 1 \end{bmatrix}$ 1 1 0 0 | A/3 |p | (a) $\left[\alpha^2 \beta^2 \gamma^2 \delta^2\right]$ $W_3^2 = 0 \quad W_3^2$ $q = 1 W_3^1$ (b) $\left[\sqrt{\alpha}\sqrt{\beta}\sqrt{\gamma}\sqrt{\delta}\right]$ $W_{2}^{4} \| 0 \| 0$ $| r | | 1 W_{2}^{2}$ (c) $\left[\alpha + \beta \beta + \delta \delta + \gamma \gamma + \alpha\right]$ then the relationship between the sequences (d) $\left[\alpha \beta \gamma \delta\right]$ [p.q.r] and [a.b.c] is [GATE - 2015] (b) [p.q.r] = [b.c.a](a) [p.q.r] = [b.a.c]7. The first six points of the 8-point DFT of a (c) [p.q.r] = [c.a.b](d) [p.q.r] = [c.b.a]real valued sequence are 5,1-j3,0,3-j4,0 and 4. Consider two real sequences with time-origin 3+j4.. The last two points of the DFT are marked by the bold value, respectively $x_1[n] = \{1,2,3,0\}, x_2[n] = \{1,3,2,1\}$ [GATE - 2011] (a) 0, 1-j3(10) 0.1+i3

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