

Techno India NJR Institute of Technology



Course File

Digital Signal Processing (7EX5-11)

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For Techno India NJR Institute of Technology
पंकज पोखवाल
Dr. Pankaj Kumar Porwal
(Principal)



RAJASTHAN TECHNICAL UNIVERSITY, KOTA

Scheme & Syllabus

IV Year- VII & VIII Semester: B. Tech. (Electrical and Electronics Engineering)

7EX5-11: DIGITAL SIGNAL PROCESSING

Credit: 3

Max. Marks: 150(IA:30, ETE:120)

3L+0T+0P

End Term Exam: 3 Hours

| SN | CONTENTS | Hours |
|----|---|-------|
| 1 | Introduction: Objective, scope and outcome of the course. | 1 |
| 2 | Discrete-time signals and systems Discrete time signals and systems: Sequences; representation of signals on orthogonal basis; Representation of discrete systems using difference equations, Sampling and reconstruction of signals - aliasing; Sampling theorem and Nyquist rate | 08 |
| 3 | Z-transform z-Transform, Region of Convergence, Analysis of Linear Shift Invariant systems using z-transform, Properties of z-transform for causal signals, Interpretation of stability in z-domain, Inverse z-transforms. | 06 |
| 4 | Discrete Fourier Transform Frequency Domain Analysis, Discrete Fourier Transform (DFT), Properties of DFT, Convolution of signals, Fast Fourier Transform Algorithm, Parseval's Identity, Implementation of Discrete Time Systems | 10 |
| 5 | Design of Digital filters Design of FIR Digital filters: Window method, Park-McClellan's method. Design of IIR Digital Filters: Butterworth, Chebyshev and Elliptic Approximations; Low-pass, Band-pass, Bandstop and High-pass filters. Effect of finite register length in FIR filter design. Parametric and non-parametric spectral estimation. Introduction to multi-rate signal processing | 11 |
| 6 | Applications of Digital Signal Processing Correlation Functions and Power Spectra, Stationary Processes, Optimal filtering using ARMA Model, Linear Mean-Square Estimation, Wiener Filter. | 06 |
| | TOTAL | |

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Course Overview:

The course covers theory and methods for digital signal processing including basic principles governing the analysis and design of discrete-time systems as signal processing devices. Review of discrete-time linear, time-invariant systems, Fourier transforms and z-transforms. Topics include sampling, impulse response, frequency response, finite and infinite impulse response systems, linear phase systems, digital filter design and implementation, discrete-time Fourier transforms, discrete Fourier transform, and the fast Fourier transform algorithms.

Course Outcomes:

| CO.NO. | Cognitive Level | Course Outcome |
|--------|----------------------|--|
| 1 | Comprehension | Classify different types of signals and system properties. |
| 2 | Application | Demonstrate continuous and discrete systems in time and frequency domain using different transforms. |
| 3 | Analysis | Analyze the output of IIR and FIR system . |
| 4 | Synthesis | Design and Develop Sampling and reconstruction circuit . |
| 5 | Evaluation | Evaluate the output of the MIMO systems. |

Prerequisites:

1. Fundamentals knowledge of differentiation and integration.
2. Fundamentals knowledge of partial fraction.
3. Fundamentals knowledge of Z transform and Basic signals.

Course Outcome Mapping with Program Outcome:

| Course Outcome | Program Outcomes (PO's) | | | | | | | | | | | |
|----------------|-------------------------|-----|-----|-----|-----|--------------------|-----|-----|-----|------|------|------|
| | Domain Specific | | | | | Domain Independent | | | | | | |
| CO. NO. | PO1 | PO2 | PO3 | PO4 | PO5 | PO6 | PO7 | PO8 | PO9 | PO10 | PO11 | PO12 |
| CO1 | 2 | 2 | 2 | 1 | 2 | 0 | 0 | 0 | 0 | 0 | 0 | 0 |
| CO2 | 2 | 2 | 2 | 2 | 2 | 0 | 0 | 0 | 0 | 0 | 0 | 0 |
| CO3 | 1 | 2 | 2 | 2 | 3 | 0 | 0 | 0 | 0 | 0 | 0 | 0 |
| CO4 | 3 | 2 | 3 | 3 | 1 | 0 | 0 | 0 | 0 | 0 | 0 | 0 |
| CO5 | 3 | 2 | 2 | 3 | 1 | 0 | 0 | 2 | 0 | 0 | 0 | 1 |

1: Slight (Low) , 2: Moderate (Medium), 3: Substantial (High)

Course Coverage Module Wise:

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| Lecture No. | Unit | Topic |
|-------------|------|--|
| 1 | 1 | INTRODUCTION: Objective, scope and outcome of the course. |
| 2 | 2 | DISCRETE TIME SIGNALS: Sequences |
| 3 | 2 | representation of signals on orthogonal basis |
| 4 | 2 | Sampling and reconstruction of signals |
| 5 | 2 | Sampling and reconstruction of signals |
| 6 | 2 | Discrete systems attributes |
| 7 | 2 | Discrete systems attributes |
| 8 | 2 | Z-Transform, Analysis of LSI systems |
| 9 | 2 | frequency Analysis of LTI systems |
| 10 | 2 | frequency Analysis of LTI systems |
| 11 | 2 | Inverse Systems |
| 12 | 3 | DISCRETE FOURIER TRANSFORM (DFT) |
| 13 | 3 | Discrete Fourier Transform (DFT) |
| 14 | 3 | Discrete Fourier Transform (DFT) |
| 15 | 3 | Fast Fourier Transform Algorithm |
| 16 | 3 | Fast Fourier Transform Algorithm |
| 17 | 3 | Fast Fourier Transform Algorithm |
| 18 | 3 | Fast Fourier Transform Algorithm |
| 19 | 3 | Implementation of Discrete Time Systems |
| 20 | 3 | Implementation of Discrete Time Systems |
| 21 | 4 | DESIGN OF FIR DIGITAL FILTERS |
| 22 | 4 | Window method |
| 23 | 4 | Park-McClellan's method |
| 24 | 4 | Design of IIR Digital Filters |
| 25 | 4 | Butterworth Approximation |
| 26 | 4 | Butterworth Approximation |
| 27 | 4 | Chebyshev and Elliptic Approximations |
| 28 | 4 | Chebyshev and Elliptic Approximations |
| 29 | 4 | Lowpass, Bandpass filter design |
| 30 | 4 | Band-Stop and High pass filters design |

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| | | |
|----|---|--|
| 31 | 5 | EFFECT OF FINITE REGISTER LENGTH IN FIR FILTER DESIGN |
| 32 | 5 | Effect of finite register length in FIR filter design |
| 33 | 5 | Parametric spectral estimation |
| 34 | 5 | Parametric spectral estimation |
| 35 | 5 | Nonparametric spectral estimation |
| 36 | 5 | Nonparametric spectral estimation |
| 37 | 5 | Introduction to mult-irate signal |
| 38 | 5 | Introduction to mult-irate signal |
| 39 | 5 | Application of DSP |
| 40 | 5 | Application of DSP |

TEXT/REFERENCE BOOKS

1. Digital Signal Processing: Principals, Algorithms And Applications”, Proakis, Manolakis, 4th ed., Pearson Education.
2. Discrete Time Signal Processing, Oppenheim, Schafer, 3rd ed. , PHI (2010).
3. Digital Signal Processing, Sanjit K Mitra, 4th ed., TMH.
4. Digital Signal Processing:A Modern Introduction, Ambardar, Cengage learning.

NPTEL COUSES LINK

1. <https://nptel.ac.in/courses/117/102/117102060/>

QUIZ Link

1. <https://www.javatpoint.com/digital-signal-processing-mcq>
2. <https://www.sanfoundry.com/1000-digital-signal-processing-questions/>

1. https://drive.google.com/drive/folders/1tz45RE6Fci2XRBp02k1rv8_2y9gp7vxj?usp=sharing

Assessment Methodology:

1. Practical exam using MATALB software.
2. Two Midterm exams where student have to showcase subjective learning.
3. Final Exam (subjective paper) at the end of the semester.

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5E1394

5E1394

B. Tech. V - Sem. (Main / Back) Exam., Feb.-March - 2021
PCC/PEC Electronics & Communication Engineering
5EC 4-04 Digital Signal Processing

Time: 2 Hours

[To be converted as per scheme]

Max. Marks: 82

Min. Marks: 29

Instructions to Candidates:

Attempt all ten questions from Part A, four questions out of eight questions from Part B and two questions out of five from Part C.

Schematic diagrams must be shown wherever necessary. Any data you feel missing may suitably be assumed and stated clearly. Units of quantities used /calculated must be stated clearly.

Use of following supporting material is permitted during examination. (Mentioned in form No. 205)

1. NIL

2. NIL

PART – A

(Answer should be given up to 25 words only)

[10×2=20]

All questions are compulsory

- Q.1 Explain about zero – order hold sampling.
- Q.2 Explain the Sampling theorem for band pass signal.
- Q.3 Write the advantages of representing the digital filter in the block diagram form.
- Q.4 Compare Canonic structure and Non-Canonic structure.
- Q.5 Explain Parseval's theorem for Discrete time sequence.
- Q.6 Explain the relation between DFT and Z – transform.
- Q.7 Explain Picket-Fence effect.
- Q.8 Explain the application of DFT in linear filtering and spectrum analysis.
- Q.9 Compare FIR filters and IIR filters.
- Q.10 What do you mean by linear phase response?

PART – B

(Analytical/Problem solving questions)

[4×8=32]

Attempt any four questions

- Q.1 Find the Nyquist rate for the continuous time signal given below and find $x[n]$.

$$x(t) = \frac{\sin(4 \times 10^3 \pi t)}{\pi t}$$

- Q.2 DFT of a sequence $x(n)$ is given by-

$$X(k) = \{6, 0, -2, 0\}$$

Determine $x(n)$.

- Q.3 Find direct forms – II realizations for the second order filter given by-

$$y(n) = 2b \cos \omega_0 y(n-1) - b^2 y(n-2) + x(n) - b \cos \omega_0 x(n-1)$$

- Q.4 Find the N – Point DFT of the following sequence-

$$h(n) = \begin{cases} \frac{1}{3} & \text{for } 0 \leq n \leq 2 \\ 0 & \text{elsewhere} \end{cases}$$

- Q.5 Show that $z\left(\frac{1}{n+1}\right) = z \log\left(\frac{z}{z+1}\right)$, $n > 0$

- Q.6 Write the short notes of design of IIR digital filter.

- Q.7 A causal discrete – time LTI system is described by-

$$y(n) - \frac{3}{4}y(n-1) + \frac{1}{8}y(n-2) = x(n)$$

Where $x(n)$ and $y(n)$ are the input and output of the system, respectively. Determine the system function $H(z)$ and impulse response $h(n)$ of the system.

- Q.8 Explain the concept of multirate signal processing and different application of DSP.

PART – C

(Descriptive/Analytical/Problem Solving/Design Questions) [2×15=30]

Attempt any two questions

- Q.1 Determine the 8-Point DFT of the following sequence-
 $x[n] = \left\{ \frac{1}{2}, \frac{1}{2}, \frac{1}{2}, \frac{1}{2}, 0, 0, 0, 0 \right\}$ use in place radix – 2 decimation in time FFT Algorithm.
- Q.2 Draw the cascade and parallel realizations for the following system function-
$$H(z) = \frac{1 + \frac{1}{4}z^{-1}}{(1 + \frac{1}{2}z^{-1})(1 + \frac{1}{2}z^{-1} + \frac{1}{4}z^{-2})}$$
- Q.3 Using bilinear transformation, design Butterworth filter which satisfy the following condition-
$$0.8 \leq |H/e^{j\omega}| \leq 1 \quad 0 \leq \omega \leq 0.2 \pi$$
$$|H/e^{j\omega}| \leq 0.2 \quad 0.6 \pi < \omega < \pi$$
- Q.4 Compute the 8-point circular convolution for following sequence-
 $x_1(n) = \{1, 1, 1, 1, 0, 0, 0, 0\}$
 $x_2(n) = \sin\left(\frac{3\pi n}{8}\right) \quad 0 \leq n \leq 7$
- Q.5 Determine the impulse response of $h(n)$ for the system described by the second order difference equation-
 $y(n) - 4y(n-1) + 4y(n-2) = x(n) - x(n-1)$
Where $y(-1) = y(-2) = 0$
-