

B.E ELECTRICAL AND ELECTRONICS ENGINEERING(EEE) CHOICE BASED CREDIT SYSTEM (CBCS) SEMESTER -VI			
DIGITAL SIGNAL PROCESSING (Core Subject)			
Subject Code	15EE63	IA Marks	20
Number of Lecture Hours/Week	04	Exam Hours	03
Total Number of Lecture Hours	50	Exam Marks	80
Credits - 04			
Course objectives:			
<ul style="list-style-type: none"> • To define Discrete Fourier transform and its properties. • To evaluate DFT of various signals using properties of DFT. • To explain different linear filtering techniques. • To explain the evaluation of DFT and inverse DFT using fast and efficient algorithms • To discuss impulse invariant transformation, bilinear transformation techniques and their properties. • To design infinite impulse response Butterworth digital filters using impulse invariant and bilinear transformation techniques. • To design infinite impulse response Chebyshev digital filters using impulse invariant and bilinear transformation techniques. • To discuss direct, cascade, parallel and ladder methods of realizing a digital IIR filter. • To discuss window functions used for the design of FIR filters. • To discuss windowing technique of designing FIR filter. • To discuss frequency sampling technique of designing FIR filter. • To discuss direct, cascade and linear phase form of realizing a digital FIR filter. ■ 			
Module-1			Teaching Hours
Discrete Fourier Transforms: Definitions, properties-linearity, shift, symmetry Properties- circular convolution – periodic convolution, use of tabular arrays, circular arrays, Stock ham’s method, linear convolution – two finite duration sequence, one finite & one infinite duration, overlap add and save methods. ■			10
Revised Bloom’s Taxonomy Level	L ₁ – Remembering, L ₂ – Understanding, L ₃ – Applying, L ₄ – Analysing. L ₅ – Evaluating		
Module-2			
Fast Fourier Transforms Algorithms: Introduction, decimation in time algorithm, first decomposition, number of computations, continuation of decomposition, number of multiplications, computational efficiency, decimation in frequency algorithms, Inverse radix – 2 algorithms. ■			10
Revised Bloom’s Taxonomy Level	L ₁ – Remembering, L ₂ – Understanding, L ₃ – Applying, L ₄ – Analysing. L ₅ – Evaluating		
Module-3			
Design of IIR Digital Filters: Introduction, impulse invariant transformation, bilinear transformations, All pole analog filters- Butterworth & Chebyshev filters, design of digital Butterworth filter by impulse invariant transformation and bilinear transformation, Frequency transformations. ■			10
Revised Bloom’s Taxonomy Level	L ₁ – Remembering, L ₂ – Understanding, L ₃ – Applying, L ₄ – Analysing. L ₅ – Evaluating		
Module-4			
Design of IIR Digital Filters (Continued): Design of digital Chebyshev –type I filter by impulse invariant transformation and bilinear transformation, Frequency transformations. Realization of IIR digital systems: direct form, cascade form and parallel form, Ladder structures for equal degree polynomial. ■			10
Revised Bloom’s Taxonomy Level	L ₁ – Remembering, L ₂ – Understanding, L ₃ – Applying, L ₄ – Analysing, L ₅ – Evaluating		

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15EE63 DIGITAL SIGNAL PROCESSING (Core Subject) (continued)				
Module-5				Teaching Hours
Design of FIR Digital Filters: Introduction, windowing, rectangular, modified rectangular. Hamming, Hanning, Blackman window, design of FIR digital filters by use of windows, Design of FIR digital filters -frequency sampling techniques. Realization of FIR systems: direct form, cascade form, linear phase form ■				10
Revised Bloom's Taxonomy Level	L ₁ – Remembering, L ₂ – Understanding, L ₃ – Applying, L ₄ – Analysing, L ₅ – Evaluating			
Course outcomes: At the end of the course the student will be able to: <ul style="list-style-type: none"> • Compute the DFT of various signals using its properties and linear filtering of two sequences. • Apply fast and efficient algorithms for computing DFT and inverse DFT of a given sequence • Design infinite impulse response Butterworth digital filters using impulse invariant / bilinear transformation technique. • Design infinite impulse response Chebyshev digital filters using impulse invariant or bilinear transformation technique. • Realize a digital IIR filter by direct, cascade, parallel and ladder methods of realization. • Discuss different window functions and frequency sampling method used for design of FIR filters. • Design FIR filters by use of window function or by frequency sampling method. • Realize a digital FIR filter by direct, cascade, and linear phase form. ■ 				
Graduate Attributes (As per NBA) Engineering Knowledge, Problem analysis, Design/ Development of Solutions, Modern Tool Usage.				
Question paper pattern: <ul style="list-style-type: none"> • The question paper will have ten full questions carrying equal marks. Each full question consisting of 16 marks. • There will be two full questions (with a maximum of four sub questions) from each module. • Each full question will have sub question covering all the topics under a module. • The students will have to answer five full questions, selecting one full question from each module. ■ 				
Textbook				
1	Introduction to Digital Signal Processing	Jhonny R. Jhonson	Pearson	1 st Edition, 2016
Reference Books				
1.	Digital Signal Processing – Principles, Algorithms, and Applications	Jhon G. Proakis Dimitris G. Manolakis	Pearson	4 th Edition, 2007.
2.	Digital Signal Processing	A.NagoorKani	McGraw Hill	2 nd Edition, 2012
3	Digital Signal Processing	Shaila D. Apte	Wiley	2 nd Edition, 2009
4	Digital Signal Processing	Ashok Amberdar	Cengage	1 st Edition, 2007
5	Digital Signal Processing	Tarun Kumar Rawat	Oxford	1 st Edition, 2015