

21CSE21	DIGITAL SIGNAL PROCESSING	L	T	P	C
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<p><b><u>Course Objectives</u></b></p> <ul style="list-style-type: none"> <li>To introduce discrete Fourier transform and its applications.</li> <li>To teach the design of infinite and finite impulse response filters for filtering undesired signals.</li> <li>To introduce signal processing concepts in systems having more than one sampling frequency.</li> </ul>					
<b>UNIT I</b>	<b>SIGNALS AND SYSTEMS</b>	<b>9 Hours</b>			
Basic elements of DSP – concepts of frequency in Analog and Digital Signals – sampling theorem – Discrete – time signals, systems – Analysis of discrete time LTI systems – Z transform – Convolution – Correlation.					
<b>UNIT II</b>	<b>FREQUENCY TRANSFORMATIONS</b>	<b>9 Hours</b>			
Introduction to DFT – Properties of DFT – Circular Convolution – Filtering methods based on DFT – FFT Algorithms – Decimation – in – time Algorithms, Decimation – in – frequency Algorithms – Use of FFT in Linear Filtering – DCT – Use and Application of DCT.					
<b>UNIT III</b>	<b>IIR FILTER DESIGN</b>	<b>9 Hours</b>			
Structures of IIR – Analog filter design – Discrete time IIR filter from analog filter – IIR filter design by Impulse Invariance, Bilinear transformation, Approximation of derivatives – (LPF, HPF, BPF, BRF) filter design using frequency translation.					
<b>UNIT IV</b>	<b>FIR FILTER DESIGN</b>	<b>9 Hours</b>			
Structures of FIR – Linear phase FIR filter – Fourier Series – Filter design using windowing techniques (Rectangular Window, Hamming Window, Hanning Window), Frequency sampling techniques					
<b>UNIT V</b>	<b>FINITE WORD LENGTH EFFECTS IN DIGITAL FILTERS</b>	<b>9 Hours</b>			
Binary fixed point and floating point number representations – Comparison – Quantization noise – truncation and rounding – quantization noise power- input quantization error- coefficient quantization error – limit cycle oscillations-dead band- Overflow error-signal scaling.					
<b>UNIT VI</b>	<b>RECENT TRENDS</b>				
Recent trends in Filters					
<b>TOTAL PERIODS: 45</b>					
<p><b><u>Course Outcomes:</u></b></p> <p>At the end of the course, Students can able to</p> <ul style="list-style-type: none"> <li>Perform frequency transforms for the signals.</li> <li>Design IIR and FIR filters.</li> <li>Finite word length effects in digital filters</li> </ul>					

**Text books:**

1. John G. Proakis and Dimitris G. Manolakis, "Digital Signal Processing – Principles, Algorithms & Applications", Fourth Edition, Pearson Education, Prentice Hall, 2007.

**Reference Books:**

1. Emmanuel C. Ifeachor, and Barrie W. Jervis, "Digital Signal Processing", Second Edition, Pearson Education, Prentice Hall, 2002.
2. Sanjit K. Mitra, "Digital Signal Processing – A Computer Based Approach", Third Edition, Tata Mc Graw Hill, 2007.