

## DIGITAL SIGNAL PROCESSING LABORATORY

|   |  |                       |   |   |                   |               |     |       |
|---|--|-----------------------|---|---|-------------------|---------------|-----|-------|
| VI Semester: ECE  |  |                       |   |   |                   |               |     |       |
| Course Code   | Category                                   | Hours /Week           |   |   | Credits           | Maximum Marks |     |       |
| AECB25  | Core                                       | L                     | T | P | C                 | CIA           | SEE | Total |
|   |  | -                     | - | 2 | 1                 | 30            | 70  | 100   |
| Contact Classes: Nil  | Tutorial Classes: Nil                      | Practical Classes: 24 |   |   | Total Classes: 24 |               |     |       |
| I. COURSE OVERVIEW:   |  |                       |   |   |                   |               |     |       |
| This course is concerned with the implementation of digital signal processing algorithms using different computational platforms such as MATLAB and DSP tools that give core knowledge to develop the real time applications in the area of DSP. It focuses on the convolution, discrete Fourier transform, fast Fourier transform algorithms, digital filter design and multi rate signal processing. Digital signal processing applications are used in speech processing, image processing, audio and video data compression, communication systems. |  |                       |   |   |                   |               |     |       |
| II. OBJECTIVES:   |  |                       |   |   |                   |               |     |       |
| The course should enable the students to:   |  |                       |   |   |                   |               |     |       |
| I The behavior of discrete time signals and systems in time and frequency domain.   |  |                       |   |   |                   |               |     |       |
| II The analysis of IIR, FIR digital filters and multi rate signal processing systems.   |  |                       |   |   |                   |               |     |       |
| III The implementation of real time digital signal processing algorithms using MATLAB tool and TI TMS67XX target board.   |  |                       |   |   |                   |               |     |       |
| III. COURSE OUTCOMES:   |  |                       |   |   |                   |               |     |       |
| After successful completion of the course, students should be able to:  |  |                       |   |   |                   |               |     |       |
| CO 1 Apply discrete Fourier transforms for spectral analysis of discrete signals.   |  |                       |   |   |                   | Apply         |     |       |
| CO 2 Apply fast Fourier transform algorithms for reducing computational complexity of discrete Fourier transform.   |  |                       |   |   |                   | Apply         |     |       |
| CO 3 Compare IIR digital filter and FIR Digital filters using different methods.  |  |                       |   |   |                   | Evaluate      |     |       |
| CO 4 Analyze the Goertzel algorithm for the generation and detection of dual-tone multi-frequency (DTMF) signaling.   |  |                       |   |   |                   | Analyze       |     |       |
| CO 5 Apply multi-rate signal processing methods such as decimation and interpolation for interfacing the digital systems with different sampling rates.   |  |                       |   |   |                   | Apply         |     |       |
| CO 6 Apply the digital signal processing algorithms for designing real time embedded signal processing applications.  |  |                       |   |   |                   | Apply         |     |       |
| LIST OF EXPERIMENTS   |  |                       |   |   |                   |               |     |       |
| WEEK-1  | LINEAR CONVOLUTION VS CIRCULAR CONVOLUTION |                       |   |   |                   |               |     |       |
| Generation of linear convolution without using built in function and the function conv in MATLAB<br>Generation of circular convolution without using built in function in MATLAB  |  |                       |   |   |                   |               |     |       |
| WEEK -2   | DFT AND IDFT                               |                       |   |   |                   |               |     |       |
| Compute the Discrete Fourier Transform and IDFT with and without fft and ifft in MATLAB   |  |                       |   |   |                   |               |     |       |

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| <b>WEEK -3</b>   | <b>OVERLAPADD AND OVERLAP-SAVE METHODS</b>                                    |
| Implementation of Linear convolution using DFT (Overlapadd and Overlap-Save methods).                      |   |
| <b>WEEK - 4</b>  | <b>DIT-FFT ALGORITHM</b>  |
| Implementation of Decimation-in-time radix-2 FFT algorithm   |   |
| <b>WEEK -5</b>   | <b>DIF-FFT ALGORITHM</b>  |
| Implementation of Decimation-in-frequency radix-2 FFT algorithm  |   |
| <b>WEEK-6</b>  | <b>IIR DIGITAL FILTERUSING BUTTERWORTH METHOD AND BILINEAR TRANSFORMATION</b> |
| Implementation of IIR digital filter using Butterworth method and bilinear transformation                  |   |
| <b>WEEK -7</b>   | <b>IIR Digital Filter Using Chebyshev (Type I And II) Method</b>              |
| Implementation of IIR digital filter using Chebyshev (Type I and II) method                                |   |
| <b>WEEK -8</b>   | <b>FIR DIGITAL FILTER USING WINDOWS</b>                                       |
| Implementation of FIR digital filter using window (Rectangular, Hamming, Hanning, Bartlett) methods.       |   |
| <b>WEEK -9</b>   | <b>FIR DIGITAL FILTER USING FREQUENCY SAMPLING METHOD</b>                     |
| Implementation of FIR digital filter using frequency sampling method                                       |   |
| <b>WEEK 10</b>   | <b>OPTIMUM EQUIRIPPLE FIR DIGITAL FILTER</b>                                  |
| Implementation of optimum equiripple FIR digital filter using window methods                               |   |
| <b>WEEK 11</b>   | <b>DTMF TONE GENERATION AND DETECTION</b>                                     |
| DTMF Tone Generation and Detection Using Goertzel Algorithm  |   |
| <b>WEEK 12</b>   | <b>SAMPLING RATE CONVERSION</b>   |
| Implementation of sampling rate conversion by decimation, interpolation and a rational factor using MATLAB |   |
| <b>WEEK 13</b>   | <b>SINE WAVE GENERATION</b>   |
| a) Implementation of DFT b) Sine wave generation using lookup table with values generated from MATLAB      |   |
| <b>WEEK 14</b>   | <b>IIR AND FIR FILTERS USING DSP KITS</b>                                     |
| IIR and FIR Filter Implementation using DSP Kits   |   |
| <b>Reference Books:</b>  |   |

1. Robert J. Schilling, Sandra L. Harris, "Fundamentals of Digital Signal Processing using MATLAB", Thomson Engineering, 2<sup>nd</sup> Edition, 2005.
2. Vinay K. Ingle, John G. Proakis, "Digital Signal Processing Using MATLAB", Cengage 4<sup>th</sup> Edition, 2009.
3. DSK Donald Reay, Rulph Chassaing, "Digital Signal Processing and Applications with the TMS 320C6713 and TMS 320C6416" Wiley 2<sup>nd</sup> Edition.

#### Web References:

1. [http://www.ece.iit.edu/~biitcomm/Yarmouk/Digital%20Signal%20Processing%20Using%20Matlab%20v4.0%20\(John%20G%20Proakis\).pdf](http://www.ece.iit.edu/~biitcomm/Yarmouk/Digital%20Signal%20Processing%20Using%20Matlab%20v4.0%20(John%20G%20Proakis).pdf)
2. [http://web.mit.edu/acmath/matlab/course16/16.62x/16.62x\\_Matlab.pdf](http://web.mit.edu/acmath/matlab/course16/16.62x/16.62x_Matlab.pdf)
3. <https://www.mathworks.com/solutions/dsp.html>
4. <http://www.iare.ac.in>

#### **SOFTWARE AND HARDWARE REQUIREMENTS FOR A BATCH OF 36 STUDENTS**

**HARDWARE:** Desktop Computer Systems 18 nos and TMS 320C6713 DSP kits

**SOFTWARE :** MATLAB, CCStudio\_v3.1