

## Digital Signal Processing with MATLAB® (25 hours)

This course mainly deals with using MATLAB® Signal Processing toolbox for Digital signal processing, analysis, visualization, and algorithm development. The training covers various topics such as windowing techniques, filter design, transforms, multi-rate signal processing etc.

COURSE CONTENT :	
Introduction to DSP (3 hours)	<ul> <li>Introduction to DSP</li> <li>Sampled data systems</li> <li>Aliasing and antialiasing</li> <li>Reconstruction</li> <li>Practical limitations</li> <li>Frequency &amp; amplitude resolution</li> <li>Quantization and timing errors</li> <li>Correlation and convolution</li> <li>Frequency analysis</li> <li>Fourier transforms</li> <li>Frequency 'leakage'</li> <li>Windowing</li> <li>Multi-rate signal processing</li> </ul>
Transforms (2 hours)	<ul> <li>Fourier Transform</li> <li>Z – Transform</li> <li>DCT Transform</li> <li>Wavelet Transform</li> </ul>
Filters (5 hours)	<ul> <li>FIR Filter –</li> <li>FIR filter basics</li> <li>Analysis of FIR filters</li> <li>Frequency &amp; impulse responses</li> <li>The window design method</li> <li>Optimization design methods</li> <li>Practical limitations of FIR filters</li> <li>IIR filter -</li> <li>IIR filter basics</li> <li>Analysis of FIR filters</li> <li>Frequency &amp; impulse responses</li> <li>IIR filter design</li> <li>Poles, zeroes and filter response</li> </ul>



DSP with MATLAB®	<ul> <li>Introduction to DSP Toolbox</li> </ul>
(5 hours)	<ul> <li>Signal processing functions in MATLAB® (conv, conv2,</li> </ul>
	corrcoef, cov, cplxpair, deconv, fft, fft2, fftshift, filter2, freqspace,
	ifft, ifft2,unwrap)
	Ime domain analysis of a signal
	• Frequency domain analysis of a signal
Digital Filter Design in	<ul> <li>Discrete-Time Filters (Direct form I, Direct form II, lattice filters)</li> </ul>
MATLAB®	<ul> <li>1_D Median filtering</li> </ul>
(2 hours)	<ul> <li>Butterworth filter design</li> </ul>
	<ul> <li>Chebyshev Type I filter design (pass band ripple)</li> </ul>
	<ul> <li>Chebyshev Type II filter design (stop band ripple)</li> </ul>
	<ul> <li>Raised cosine FIR filter design</li> </ul>
	<ul> <li>Recursive digital filter design</li> </ul>
Analog Filter Design in	<ul> <li>Analog Lowpass Filter Prototypes</li> </ul>
MATLAB®	Analog Filter Transformation
(2 hours)	Bi-linesr transformation
	<ul> <li>Impulse-invariant Methods</li> </ul>
	<ul> <li>Stabilising a polynomial</li> </ul>
	<ul> <li>Z-Transform partial fraction expansion</li> </ul>
Window Design	<ul> <li>Rectangular window</li> </ul>
(2 hour)	<ul> <li>Hamming window</li> </ul>
	<ul> <li>Hanning window</li> </ul>
	• Bartlett window
	Kaiser window etc
Transforms	Discrete fourier transform
(2 hour)	Discrete cosine transform
	Hilbert transform
	<ul> <li>Discrete wavelet transform</li> </ul>
	<ul> <li>inverse transforms</li> </ul>
Multi-rate Signal Processing	Decimation
(2 hours)	<ul> <li>Interpolation</li> </ul>
	Up-Sampling
	• Down-Sampling
	Re-Sampling