

National Institute of Technology Meghalaya

An Institute of National Importance

CURRICULUM

Programme		nme	M.Tech/Ph.D Year of Regulat											egulation	2021					
Department			Electronics and Communication Engineering								Semester					I				
Course Code			Course Name								Credit Structure Marks Dis					ks Distrib	stribution			
										L	Т	Р	C	INT	MID	END	Total			
EC	C 525		Speech Signal Processing and Coding								0	0	3	50	50	100	200			
		Introducing of physiology of acoustic production and perception model.									CO1	Able to m	odel an el	ectrical e	al equivalent of speech production system					
Course Objectives		Introduc freque	cing ncy c	of concept lomains.	s for analy	ysis of the	speech in t	time and		Course	CO2	Able to apply time-frequency representation in processing of speech signals								
		Introducing of techniques for speech signal coding								Outcomes	CO3	Able to extract LPC coefficients for analysis of speech signals								
		muou	iemg	, or teening	fues for sp	Jeeen sign	ai counig				CO4	Able to apply speech coding techniques for compression and processing of speech signals								
No.	Cos		Mapping with Program Outcomes (POs)												Mapping with PSOs			3		
		PO	1	PO2	PO3	PO4	PO5	PO6	PO7	PO8	PO9	PO10	PO11	PO12	PSO1	PSO2	PSO3	PSO4		
1	CO1	2		1	0	0	1	0	0	0	0	0	0	0	2	0	1	0		
2	CO2	1		2	2	2	0	0	0	0	0	0	0	1	2	0	2	0		
3	CO3	0		2	2	1	2	0	0	0	0	0	0	2	2	2	2	0		
4	CO4	0		2	0	1	2	0	0	0	0	0	0	2	2	2	2	0		
									SYL	LABUS										
No.		Content														Hours COs				
Ι	Speech production and perception, information sources in speech, linguistic aspect of speech, acoustic and articulatory phonetics, nature of speech, models for speech analysis and perception; Short-term processing: need, approach, time, frequency and time-frequency analysis;																			
II	Short-t and filt Cesptru cepstru	Short-term Fourier transform (STFT): Overview of Fourier representation, non-stationary signals, development of STFT, transform and filter-bank views of STFT. Cesptrum analysis: Basis and development, delta, delta-delta and mel-cepstrum, homomorphic signal processing, real and complex10CO2Cepstrum.																		
III	Linear LP resi Sinuso	Linear Prediction (LP) analysis: Basis and development, Levinson-Durbin's method, normalized error, LP spectrum, LP cepstrum, LP residual. Sinusoidal analysis: Basis and development, phase unwrapping, sinusoidal analysis and synthesis of speech.09CO3																		
IV	Speech coders	Speech coding: Need and parameters, classification, waveform coders, speech-specific coders, GSM, CDMA and other mobile coders; Applications: Some applications like pitch extraction, spectral analysis and coding standard.															CO4			
	Total Hours														36					
Essential Readings																				
1. Lawrence Rabiner and Ronald Schafer, "Digital Processing of Speech Signals ", Pearson, 1st Edition, 1978.																				
2. John R. DellerJr., John H.L. Hansen and John G. Proakis, "Discrete-Time Processing of Speech Signals", Wiley-IEEE Press, 1999.																				
Supplementary Readings																				
1. Do	ouglas O	Shaugh	nessy	/, "Speech	Communi	cations: Hu	iman and M	/lachine'', V	wiley-IEE	² E Press, 2 nd	Edition, 1	2008								
Z . Th	iomas F.	Quatier	i, "Di	screte-11m	le Speech	Signal Pro	cessing: Pri	inciples and	a Practice'	, Pearson, F	- Edition	, 2008.								