COURSE DESCRIPTION

| COURSE NAME SPEECH TECHNOLOGY AND ELEMENTS OF FUZY LOGIC CODE: ML21 | | | | | | CODE: ML2103 | |
|---|--|--|---|----------|--|--|--|
| STUDY YEAR MA | STER II | TER II SEMESTER 2 | | E STATUS | cultative) C | | |
| | | | | | | | |
| HOURS PER WEEK TOTAL HOURS PEI | | TOTAL HOUR | S CREDITS | (P-duri | EVALUATION ng the semester, C -oral examination, | TEACHING LANGUAGE | |
| 2 - 2 - | SEMESTE 56 | | 8 | | M | English | |
| 2 2 | | | | | | | |
| COURSE TEACHING AND SCIENTIFIC DEGREE, FIRST NAME, LAST NAME DEPARTMENT | | | | | | | |
| TEACHER PROF. HORIA-NICOLAI TEODORESCU Computer Science | | | | | | | |
| | | | | | | | |
| PREVIOUS COURSES REQUESTED Natural Languages; Formal languages and Automata | | | | | | | |
| OBJECTIVES | Acquiring background knowledge in acoustics, phonetics, and signal processing as related to speech technology Acquiring basic knowledge Introduction in general, acoustical, and articulatory phonetics | | | | | | |
| Acquiring abilities to design applications for speech processing, synthesis and recognit | | | | | | is and recognition | |
| Elements of acoustics, phonetics, and signal processing as related to speech technolog Introduction to signal processing: signals and their temporal and frequential chara sampling and sampling theorem, Fourier transform, filtering, digital filters, harmoni noises | | | | | | | |
| Basic design of elementary signal processing modules Voice signal features: pitch, formants, number of zero crossings, instantaneous power Spectrum and spectrogram – "reading" and interpreting voice sounds Design of basic modules for elementary signal features extraction Prosody estimation; fuzzy logic approach Statistical characterization of voice signals; HMM models | | | | | | aneous power | |
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| | | | | | | | |
| | Speech synthesis principles | | | | | | |
| Speech recognition principles; fuzzy logic approach | | | | | | | |
| Applications of speech technology: robotics, appliances, communications, gaming, medicin | | | | | | | |
| DESCRIPTION OF • Performing voice recordings with various technical parameters and analyzing spectrog | | | | | | | |
| SEMIIVARY / and sonograms | | | | | | | |
| | • Use of applications like WASP, GOLDWAVE, PRAAT for voice analysis | | | | | | |
| Interpretation of the waveforms of the vowel and consonant sounds (time domain Vowels characterization – vowel triangle for the Romanian language – mini-program | | | | | | e domain analysis) | |
| | | | | | | nini-project | |
| | Enouonal voice analysis Speech surthesis – various methods | | | | | | |
| | Speech synthesis – various methods Subject expedition on the heard - theoretical issues formula hands on demonstrations | | | | | | |
| | Subject exposition on the board – theoretical issues, formula, nands-on demonstrations, exercises, free discussions in class | | | | | | |
| | | | | | | | |
| BIBLIOGRAPHY (SELECTION)Lawrence R. Rabiner, Biing-Hwang Juang, Fundamentals of speech recognition. If Hall, 1993 - 507 pages Frederick Jelinek, Statistical methods for speech recognition. MIT Press, 1997. (fr Books) Sadaoki Furui, Digital speech processing, synthesis, and recognition. CRC Press, 200 Li Deng, Douglas O'Shaughnessy, Speech processing: a dynamic and optimiza | | | | | | gnition. PTR Prentice 1997. (free at Google ress. 2001 | |
| | | | | | | optimization-oriented | |
| approach. CRC Press, 2003 - 626 pages | | | | | | | |
| ELALUATION conditions participation in cominger and alogges | | | | | | | |
| EVALUATION | con | - active participa | - active participation in seminars and classes. | | | | |
| | | criteria - micro-pr | - micro-project evaluation | | | | |
| | evaluation m | - design re - micro-pr nethods - active cl | ojects (uP): 60% ass participation (| Ac): 10% | | | |
| | $\begin{array}{c} - \text{ final examination: } 30\% \\ \hline \text{final result - formula} \text{NF} = 0.6 \text{*uP} + 0.1 \text{*Ac} + 0.3 \text{ F} \\ \end{array}$ | | | 3 E | | | |
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