Advanced Topics in Speech Processing CS60116

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Introduction

- Objective
- What is speech?
- What is speech processing?
- Applications of speech processing
- How speech processing applications are developed?
- Scope of the course
- Assignments Hands-on
- Text books & References

Objective

- Gross-level understanding of how machines process speech
- Contribution of multiple disciplines in speech processing
 - Computer Science
 - Electronics & Communications
 - Electrical Engineering
 - Mechanical & Industrial Engg
 - * Healthcare
 - Financial & Banking
 - Audio & Music
 - Linguistics & Phonetics

What is Speech?

- Produced from human's vocal apparatus (Mouth)
- Natural mode of communication among humans
- Best communication aid for humans

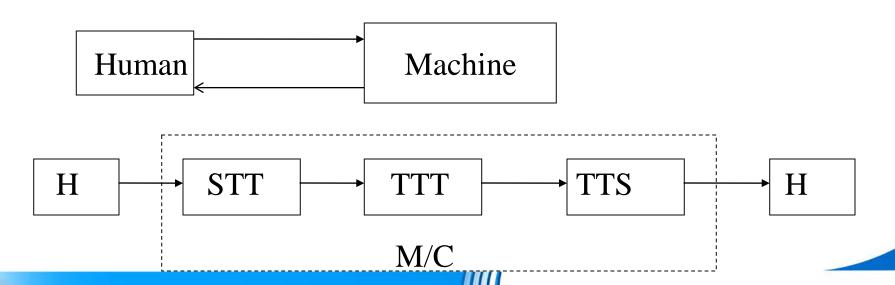
Need for speech processing

Speech: Natural mode of communication among human beings Message, Speaker information and Language information

Language constraints: Legal sound units

Legal sequence of legal sound units

Why speech processing?



Speech Processing: Applications

Speech Recognition

- Speaking interface with machines
- Automatic dictation system
- √ Healthcare

Speech Synthesis

- Speaking interface with machines
- ✓ Voice response system
- Screen readers, reading story books, etc..
- Aid for visually challenged people

• Speaker Recognition/Speaker Verification

- Voice based person authentication system
- Forensic investigation application

Speech Processing: Applications

- Language identification
- Information (data) retrieval (voice-based)
- Speech enhancement
- Pathological Speech Analysis: Detection & Classification of Voice Disorders
- Speech coding
- Paralinguistic analysis
 - Emotion recognition
 - Speaker style modeling (speaking rate, pronunciation, etc..)

How to Develop Speech Applications?

- Speech Recognition : Features to characterize sound units
- Speech Synthesis: Parameters of Vocal-tract and Excitation
- Speaker Recognition : Features to characterize speaker
- Language Identification : Features to characterize language
- Information (data) Retrieval (voice-based): Pattern discovery
- Speech Coding: Features for efficient coding and reproduction
- Speech Enhancement: Features to characterize speech, non-speech, noise, reverberation, etc...

Scope of the Course

- Introduction & Acoustic Phonetics
- Speech Signal Processing & Analysis Methods
- Speech Modelling Techniques
- Speech Systems

Scope of the Course (in detail)

Module	Topics
1	Introduction
2	Acoustic Phonetics
3	Representation and processing of speech
4	Speech analysis using time and frequency domain techniques
5	Linear prediction analysis of speech
6	Vector quantization and spectral distortion measures
7	Statistical pattern recognition
8	Probability density estimation & Gaussian mixture models
9	Hidden Markov models
10	MLP, SVM & DL
11	Speech coding & Speech recognition
12	Speech synthesis, speaker/language recognition and speech enhancement

Assignments: Hands-on

- 1. Familiarity with speech recording, playback and editing software
- 2. Effect of sampling and quantization
- 3. Recording and analysis of speech sounds
- 4. Time domain analysis of speech
- 5. Spectral analysis of speech using STFT
- 6. Spectral analysis using different windows
- 7. Sinusoidal analysis/synthesis of speech
- 8.Linear predication analysis/synthesis of speech
- 9. Cepstral analysis of speech
- 10. Estimation of pitch and formants from speech
- 11. Synthesis of vowels

Assignments: Hands-on

- 12. Voiced/unvoiced (speech/non-speech) detection (Energy, ZCR, ACF, AMDF)
- 13. Vowel recognition using Filter-bank approach
- 14. Vowel recognition using ZCR (realization of filter-bank using ZCR)
- 15. Vowel recognition using: VQ, GMM, HMM, KNN, ANN, SVM, DNN, CNN
- 16. Prototype speech recognizer
- 17. Speaker recognition
- 18. Language Identification
- 19. Text-to-Speech synthesis
- 20. Speech Enhancement (Noise subtraction)

Text Books

- 1. L. R. Rabiner and B. H. Juang, "Fundamentals of speech recognition", Pearson Education, LPE, New Delhi, 2003.
- 2. L. R. Rabiner and R. W. Schafer, "Digital processing of speech signals", Pearson Education, LPE, New Delhi, 2005.
- 3. D O'Shaughnessy, "Speech communication: Human and Machine", Second Edition, IEEE Press, NY, USA, 2012.
- 4. J. R. Deller, Jr., J.H.L. Hansen and J.G. Praokis, "Discrete-time processing of speech signals" IEEE Press, NY, USA, 1999.
- 5. T. F. Quateri, "Discrete-time speech signal processing: Principles and practice", Pear Education, LPE, New Delhi, 2004.
- 6. B. Gold and N. Morgan, *Speech and Audio Signal Processing*, Wiley Student Edition, Singapore, 2004.
- 7. J. Benesty, M. M. Sondhi and Y. Huang, "Springer Handbook on Speech Processing Springer publishers, 2008.
- 8. X. Huang, A. Acero and H. W. Hon, "Spoken Language Processing", PHI, 2001.

Course Evaluation Details

- Mid-Sem: 30%
- End-Sem: 50%
- Class Test I: 10%
- Class Test II: 10%

References

- 1.IEEE Trans. Audio, Speech and Language Processing
- 2. Speech Communication (Elsivier)
- 3. Computer Speech and Language (Elsivier)
- 4. Journal of acoustical society of America (JASA)
- 5.IEEE Int. Conf. Acoust., Speech, Signal Processing (ICASSP)
- 6.Int. Conf Speech Processing (INTERSPEECH)