

Voice recognition

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0.1 Context

0.2 Learned in this study

0.3 Things to explore

- MFC/MFCC
- Recognize speakers using per speaker models
 - Start with a single model for all speakers, and slowly figure out when a given speaker speaks, then retrain their individual model to become more and more accurate

0.4 To try

- Determine vocal tract size

1 Overview

The goal of this project is to recognize a person based on a record of his/her voice.

- Record audio
- Convert a certain window size (e.g., 20ms long) of the signal into the frequency domain using a fast Fourier transform

2 From the Deep Learning Book

- Factors of variation
 - Age
 - Sex
 - Accent
 - Words spoken

3 From the Bay Area Deep Learning School

- Application matters
 - Styles of speech
 - * Read
 - * Conversational
 - * Spontaneous
 - * Command/control
 - Issues
 - * Disfluency/Stuttering
 - * Noise

- * Microphone quality/Number of channels
- * Far field
- * Reverb/Echo
- * Lombard effect
- * Speaker accents

4 From PHO121 - Speech Analysis

- Vowels can be classified by their two first formants (F1 and F2)
 - Resonant frequencies that can roughly be associated with the size of specific cavities in the vocal tract
 - F1: Pharyngeal cavity
 - F2: Front cavity

5 See also

6 References

- <https://www.youtube.com/watch?v=9dXiAecyJrY&feature=youtu.be&t=13874>
- <https://www.youtube.com/watch?v=MyNrmiJQ4dI>