Speechrecognition

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9.11.2018

Seminar Robocup

Motivation

Why even bother?

- faster and more general way to give robots commands
- a necessity for casual users
- user does not need additional hardware

Content

What is Speechrec? What does it consist of?

Content[®]

- 1. Hardware
- 2. Localisation
- 3. Signal Enhancing
- 4. Voice Activation Detection
- 5. Speaker Recognition
- 6. Speech Recognition
- 7. Natural Language Processing

Quick example

content...

Hardware

Microphones

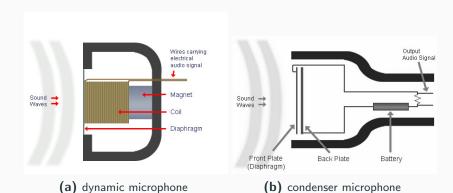


Figure 1: Diffrent kinds of microphones. Source: http://www.onlinetuner.co

Audio

Microphones provide audio defined by:

rate: resolution in time domain

bitrate: resolution in quality domain

endian: representation of signal

channel: amount of channels

interleaving: representation of signal by channel

Microphones: Polar Patterns

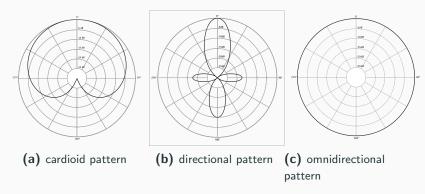


Figure 2: Diffrent kinds of microphone pattern. Source: Wikipedia

Microphone Arrays

Why?

- more sensors are always better
- several microphones can be linked together to reduce noise
- by analysing the distribution of signals and frequencies in time we can even detect the direction from where a sound was emitted

Microphone Arrays II

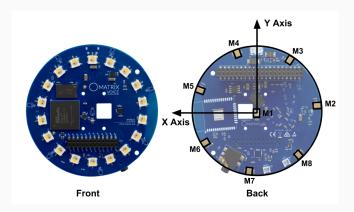


Figure 3: Matrix Voice, a microphone array. Source: https://matrix-io.github.io/matrix-documentation/matrix-voice/resources/microphone/

Tobi:

Tobi: One or two directional microphones, 2 additional for sound source localization.

Pepper:

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Pepper: An array of four omnidirectional microphones inside of the head.

Tiago:

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Tiago: Stereo microphones in the torso (black spots under his head).

Alsa

The Advanced Linux Sound Architecture manages all your physical and virtual microphones and speakers.

- It controls volume and gain
- It controls rate and bitrate of captured audio, as well as other properties
- Extensive configuration makes basically any imaginable microphone constellation possible

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- many extensions (called modules) to suit all kinds of needs
- eg. virtual devices over network, on the fly filtering, etc.

Localisation

Sound Source Localisation

content...

Signal Enhancing

Beamforming

content...

Voice Activation Detection

What we use

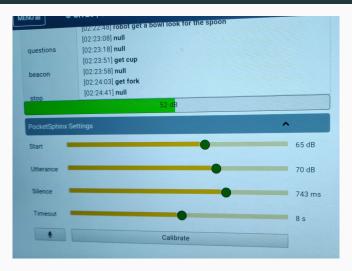


Figure 4: Double threshold voice activation detection

What we use II

A voice activation detection based on audio loudness with three states:

idle Start in this state

starting switch to this state if the audio > StartDb and stay here as long as audio > UtteranceDb

ending switch to this state if the audio < UtteranceDb and stay as long as specified via Silence, then return to idle

A maximum audio length can be specified via Timeout

Other approaches

Based on...

- loudness based on decibel calculation, it will only take into account the single most extreme value in an audio frame
 - energy in contrast to loudness-based approaches, energy calculation will take all values in an audio frame into consideration
- frequency will calculate frequencies and search for those typically used by human speech

Speaker Recognition

Speech Recognition

Approaches

There are three big approaches for consumer/ robotics speech recognition:

- Hidden Markov Models
- Deep Learing
- Online Services

Hidden Markov Models

Hidden Markov models are statistical models where a system is assumed to be a markov process with hidden states.

- imagine a statemachine where the transitions are modeled by probabilities
- states in this statemachine can be roughly understood as phonemes
- a most probable path through the model can be computed for a given sequence of signals or data
- this path is can then be mapped to a spoken word or sequence of words
- states are unknown (hidden) and assumed via probabilities

Sphinx

A HMM based group of application which use

- a speech model (models phenomes)
- a dictionary (maps phonemes to words)
- a language model (probabilities of word sequences)

Recent versions can provide quasi-free speech recognition.

Deep Learning & DeepSpeech

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- very big field
- computationally expensive concept to abstract high level information out of big chunks of data

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DeepSpeech

- ML architecture by Baidu
- several implementations, e.g. by mozilla using tensorflow
- detects free speech, but is somewhat phoneme based
- currently one of the best architectures in regards to word error rate

Google/Bing/... Online Speechrec

Commercial "Cloud" services provided by companies like Google, Amazon, Microsoft, etc.

- kind of a blackbox
- need fast internet connections
- typically better results than local speechrec
- not free, can be quite expensive if used extensively

Grammar vs Grammarless

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Corrected Spelling vs Phoneme based recognition

For everyday use:

For everyday use:

(Pocket-)Sphinx

For special cases:

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(Pocket-)Sphinx

For special cases:

Google Speechrec

Natural Language Processing

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- keywordspotting?

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- direct mapping?
- keywordspotting?
- how to do planning?

Grammar Analysis

Idea: Create the grammar tree of a sentence, thus making it easier to extract information.

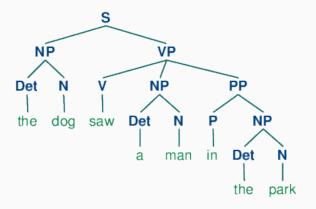


Figure 5: Example of a grammar tree. Source: https://www.nltk.org/book/ch08.html

Beyong Grammar

More intensive forms of analysis can involve:

- Statistical analysis of sentences/phrases
- tagging of phrases
- dependency parsing
- tokenization

With Pocketsphinx:

With Pocketsphinx:

Not that much :(

With Google Speechrec:

With Pocketsphinx:

Not that much :(

With Google Speechrec:

spaCy, a python library which can...

- create a grammar tree of a sentence
- classify phrases and words (in context)
- abstract information out of text
- analyse the similarity of two sentences

spaCy example

From Wikipedia:

"RoboCup is an annual international robotics competition proposed and founded in 1996 (Pre-RoboCup) by a group of university professors (among which Hiroaki Kitano, Manuela M. Veloso, and Minoru Asada). The aim of such a competition consists of promoting robotics and AI research, by offering a publicly appealing, but formidable challenge."

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spaCy:

RoboCup ORG, 1996 DATE, Hiroaki Kitano ORG, Manuela M. Veloso PERSON, Minoru Asada PERSON, Al GPE

Thanks for the Attention!

Discussion

Sources