

# 11-737 Multilingual NLP

Speech



**Carnegie Mellon University**

Language Technologies Institute

# Table of Contents

- What is speech?
- Speech applications
- Speech databases
- Speech hierarchy

# What is speech???

Watanabe's definition

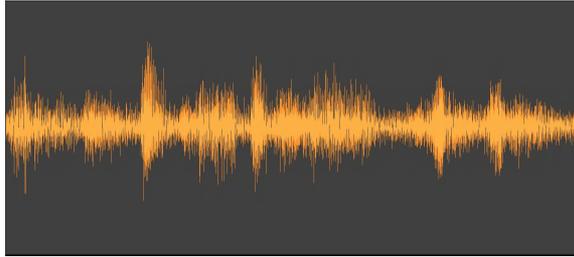
- **Sound produced by human for the communication**
- Is this speech?



- Freesound <https://freesound.org/>

# Sound

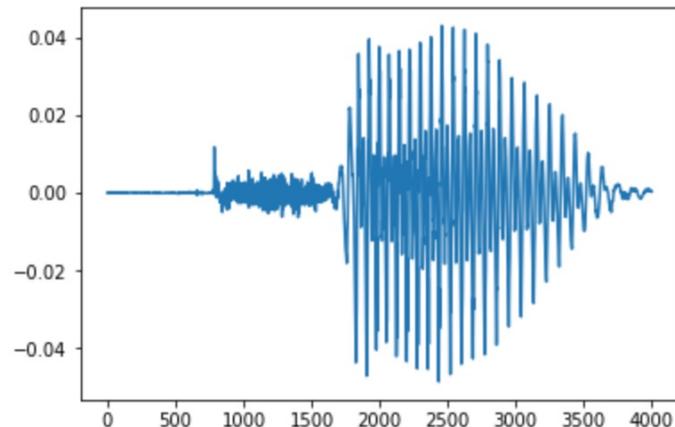
- Air pressure
- Captured by a microphone



- Governed by well known physical properties
  - Attenuation, refraction, reflection, diffraction, superposition

# Speech waveform?

- Waveform: Converting a sound pressure into a time series
- Usually **1-dimensional waveform** (mono) in this lecture
  - A lot of recording devices support stereo waveforms.
  - Then, it would be 2 (left and right) dimensional waveform
  - We also use a microphone array to capture N-dimensional waveform where N means the number of channels captured by N microphones (e.g., Alexa has 7 microphones, N=7)



# What kind of information does speech sound contain?

- Transcription
- Speaker identity
- ...
- ...

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- What is speech?
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# What kind of research topics in speech research?

- Speech recognition
- Speech synthesis
- Speech...
- Spoken...

# What kind of research topics in speech research?

- Speech recognition
- Speech synthesis
- Voice conversion
- Speaker recognition
- Language recognition
- Speech emotion recognition
- Speaker diarization
- Speech coding
- Speech perception
- Speech enhancement
- Microphone array processing
- Audio event classification and detection
- Speech separation
- Spoken language understanding
- Spoken dialogue systems
- Speech translation
- Multimodal processing
- Speech corpus

# Any others?

- Speech recognition
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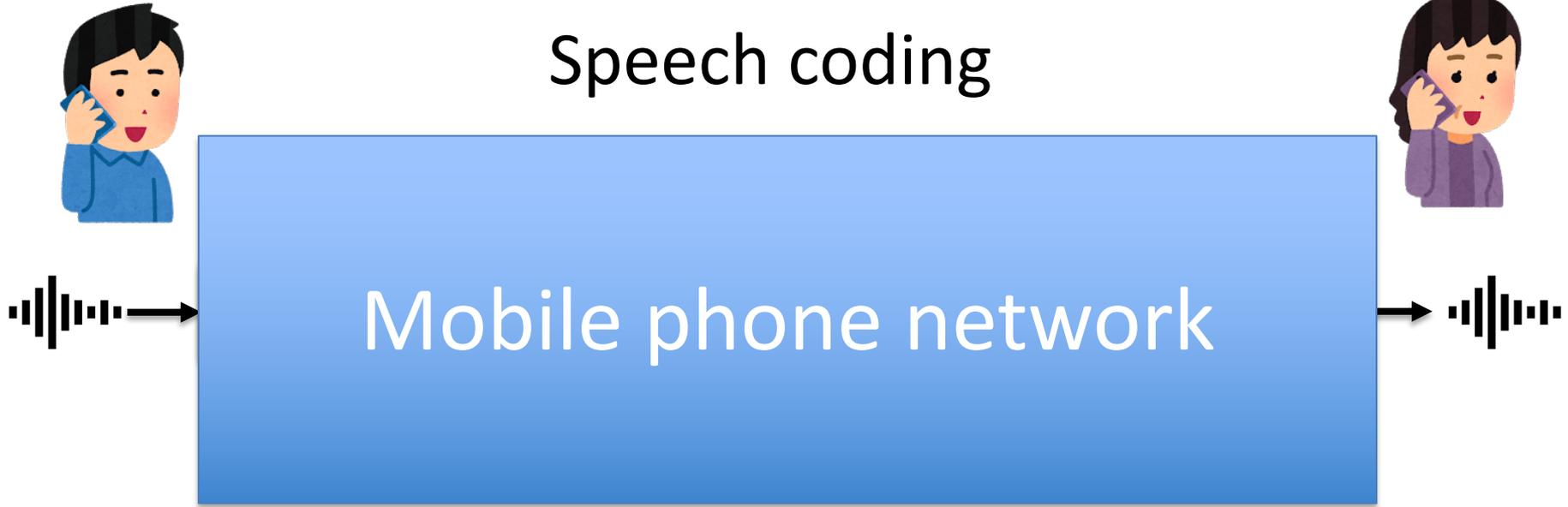
# What is the **most widely** used technique among them?

- Speech recognition
- Speech synthesis
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# What is the **most widely** used technique among them?

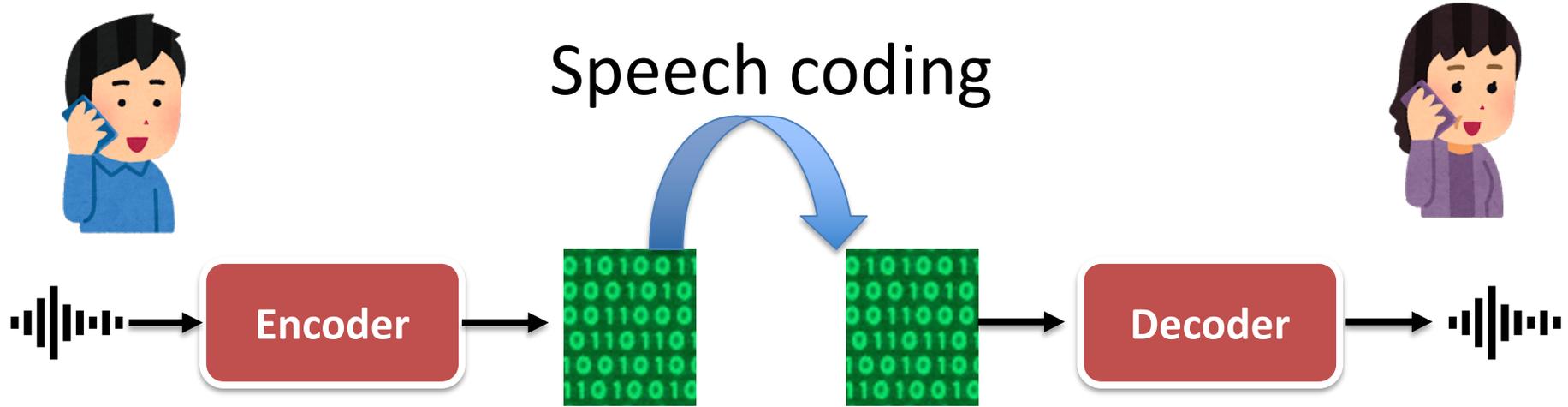
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# Speech coding



Our infrastructure

# Speech coding



Our infrastructure

Compress the data while keep the speech information

Statistical method based on linear prediction

# What kind of research topics in speech research?

- **Speech recognition**
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# Automatic Speech Recognition (ASR)



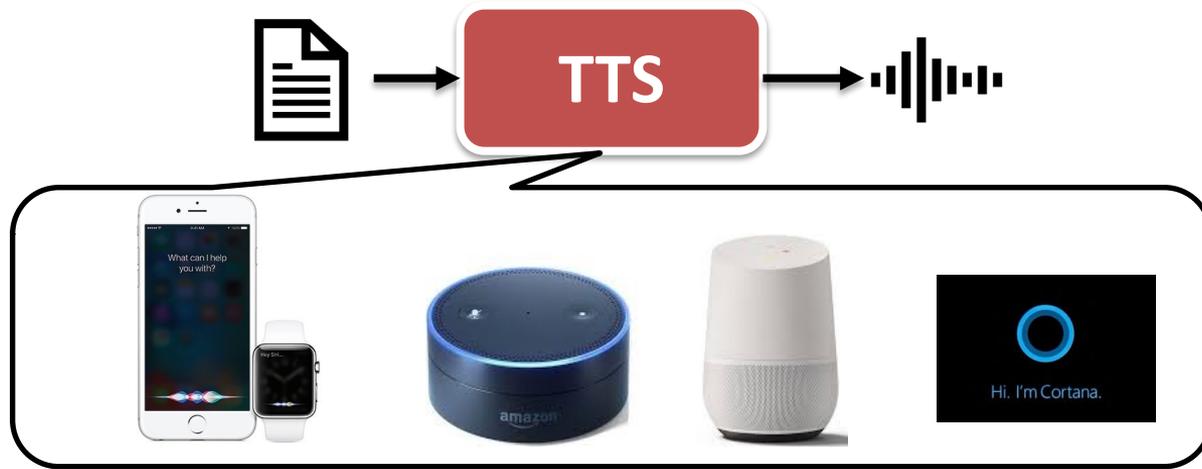
Widely used in many applications!

We will discuss it in more details in the next lecture

# What kind of research topics in speech research?

- Speech recognition
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# Speech Synthesis (TTS: Text to Speech)



Inverse problem of ASR

We will discuss it week after the next week

# What kind of research topics in speech research?

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- **Speech translation**
- Multimodal processing
- Speech corpus

# Speech Translation

source speech to target text



Combining ASR + machine translation

☹️ Complicated systems, Error Propagation

End-to-End modeling has been actively studied

# Speech Translation

source speech to target speech



ASR + machine translation + TTS

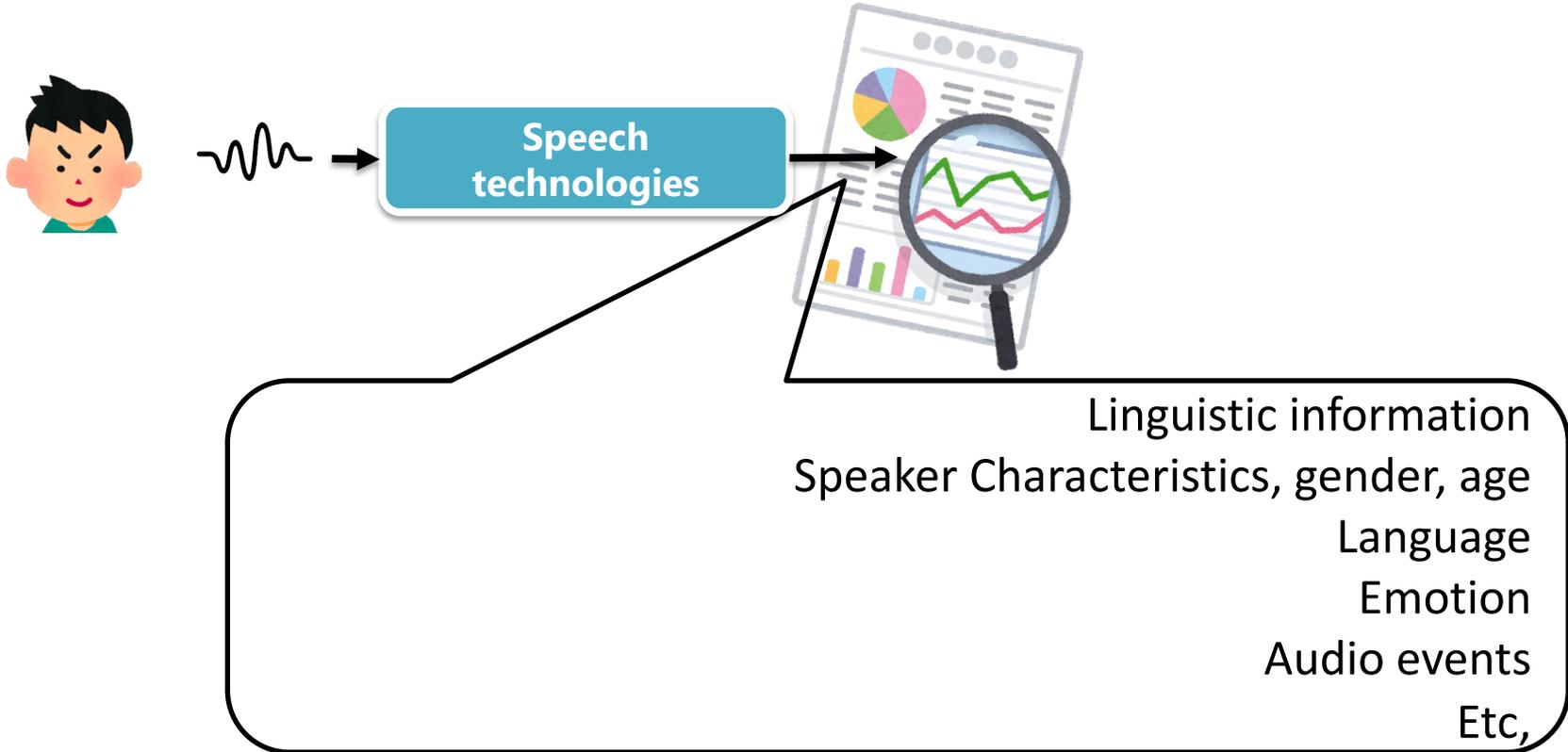
End-to-end?

One of the goals of multilingual NLP

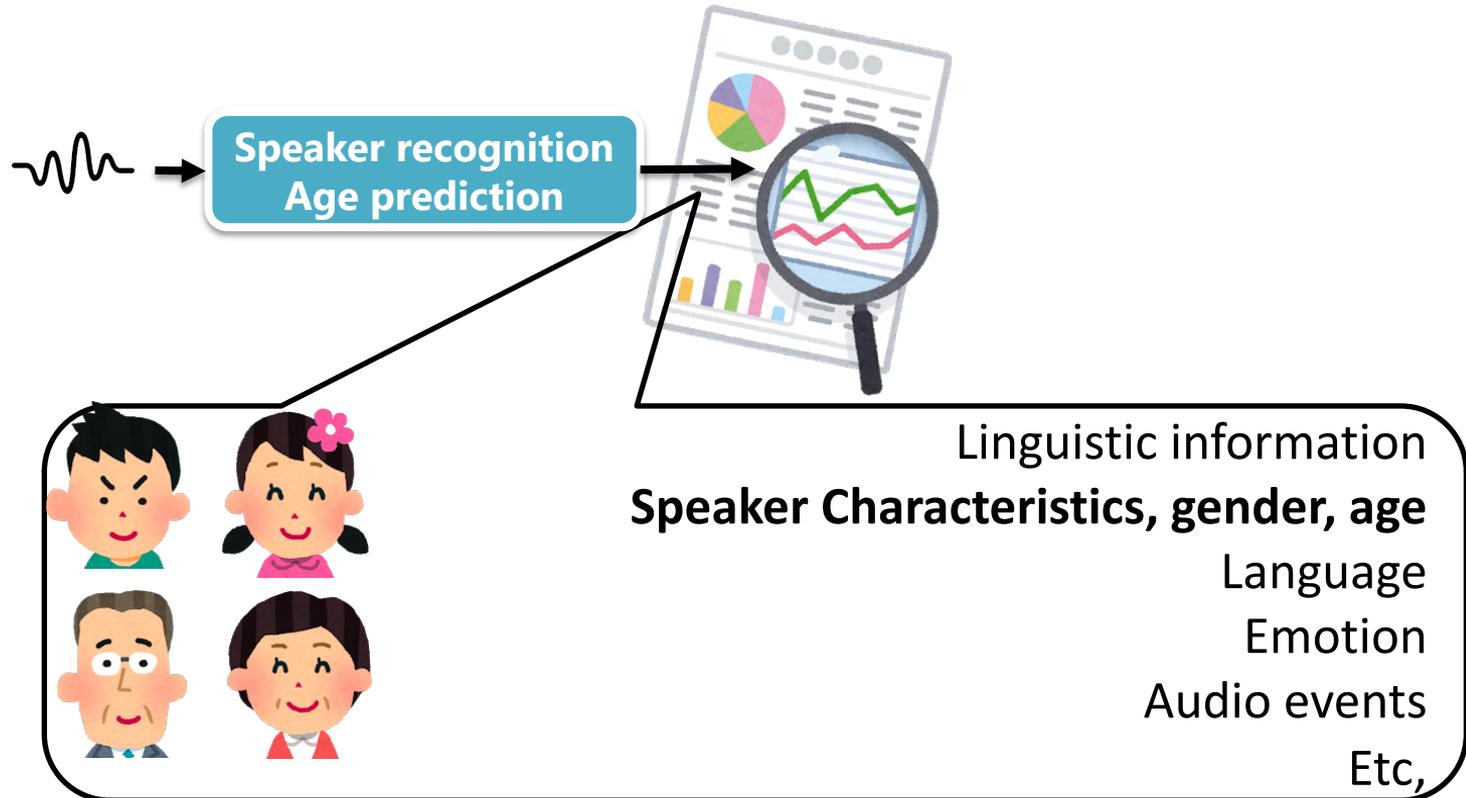
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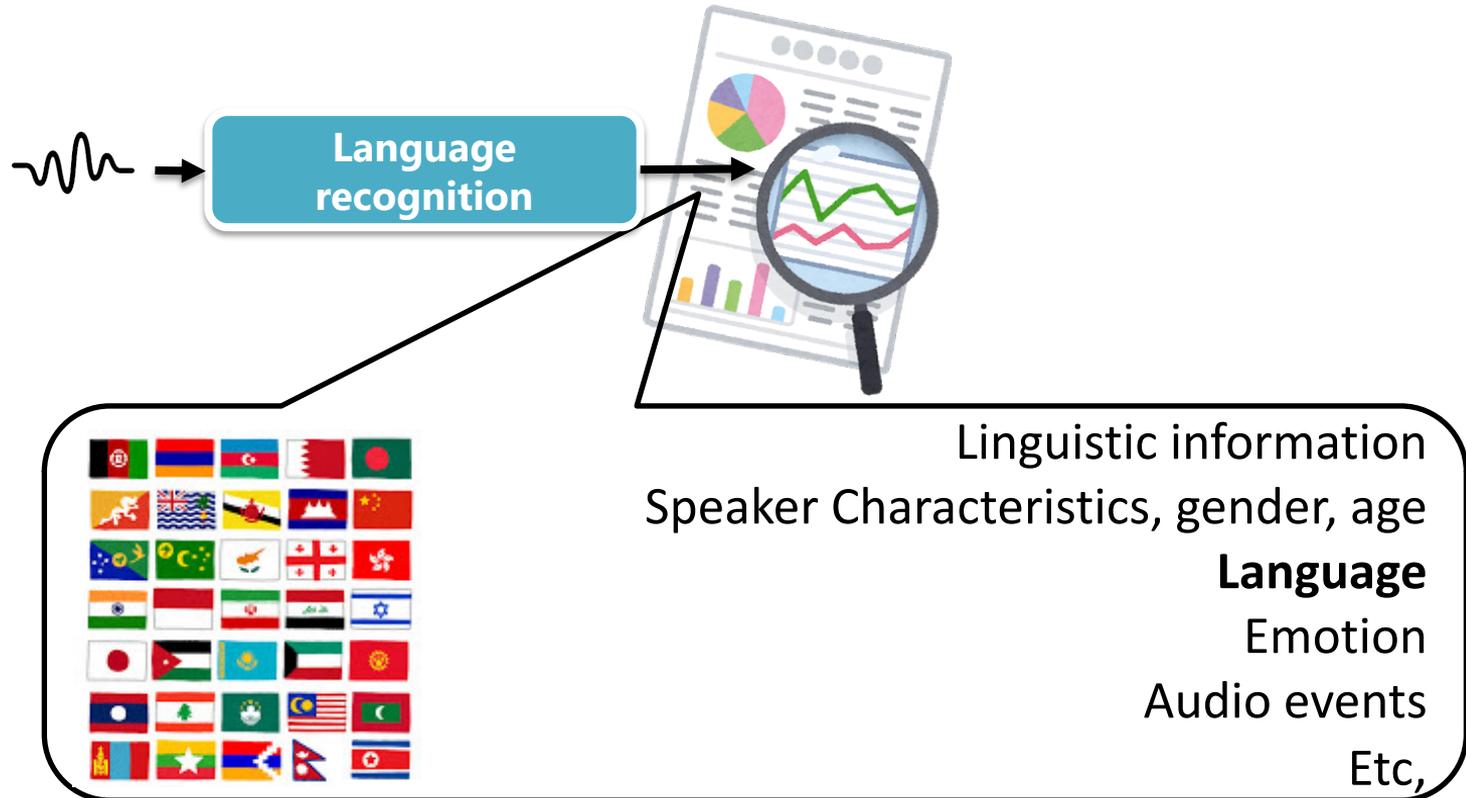
# Speaker Profiling/Audio Disentanglement



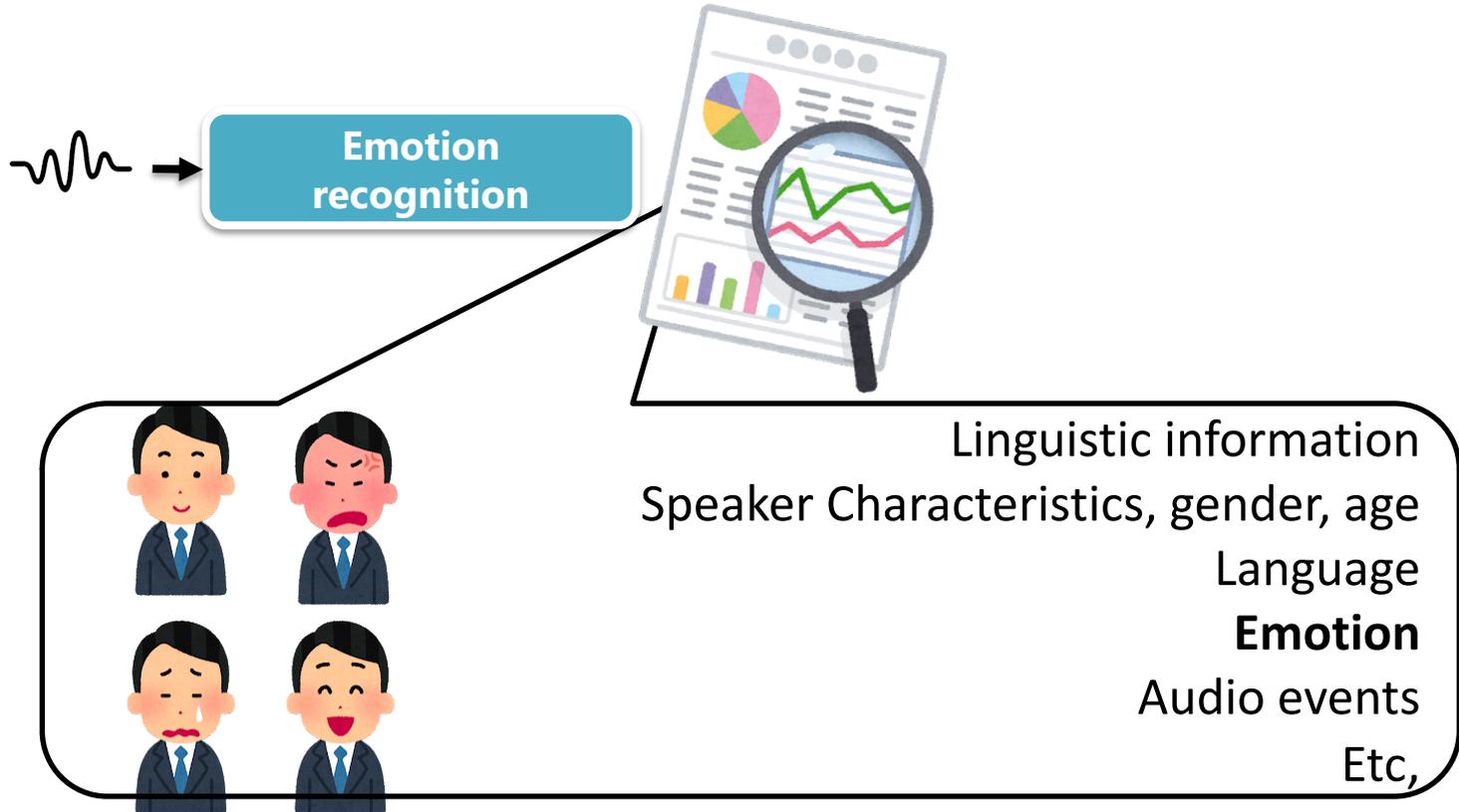
# Speaker Profiling/Audio Disentanglement



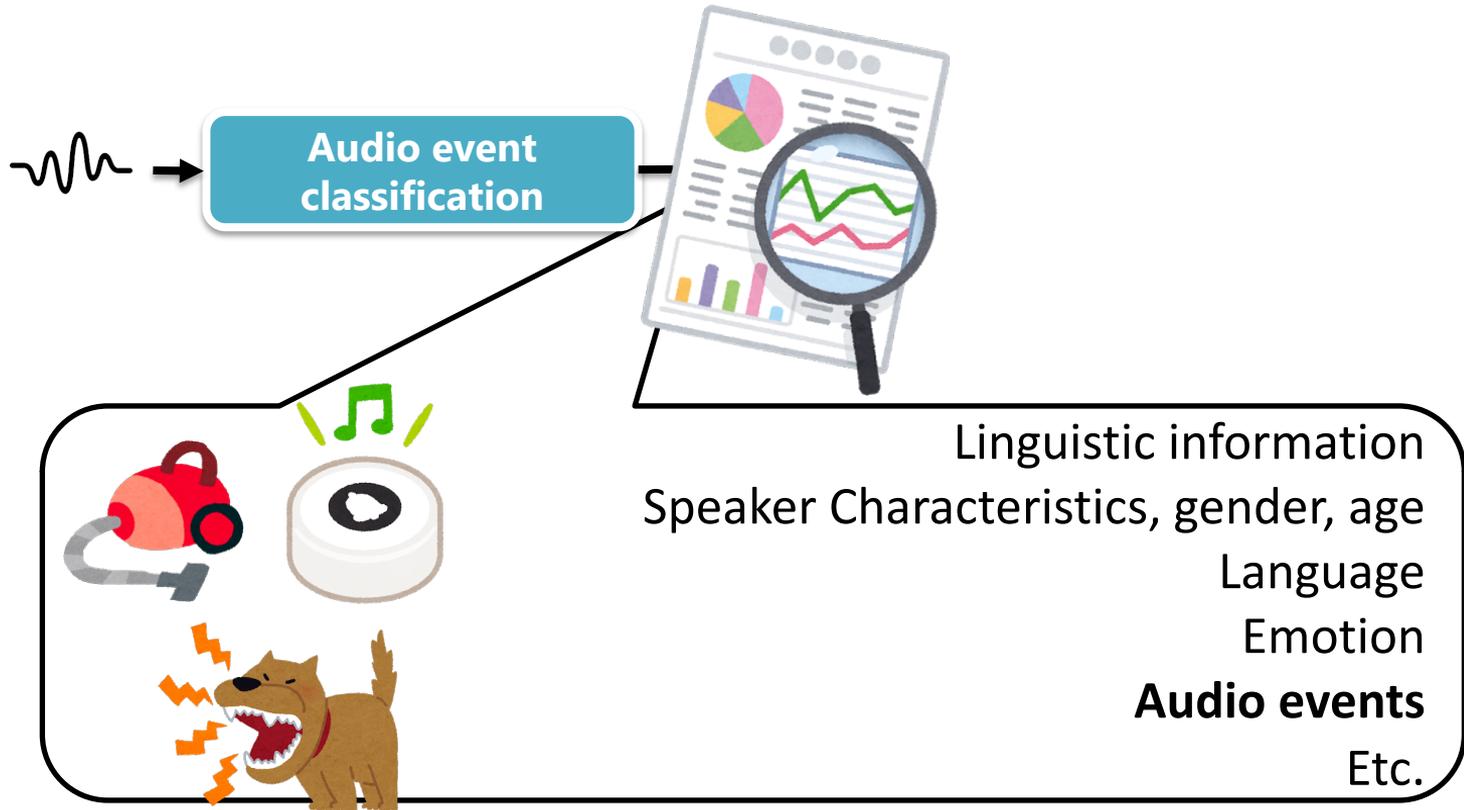
# Speaker Profiling/Audio Disentanglement



# Speaker Profiling/Audio Disentanglement



# Speaker Profiling/Audio Disentanglement



# Privacy in speech

- Speech contains various profiling information
- Current speech processing techniques require massive computations
  - Most computations at a server
  - Serious privacy issues
  - On device AI

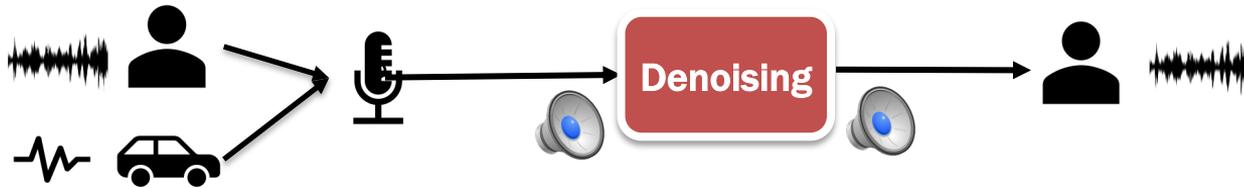
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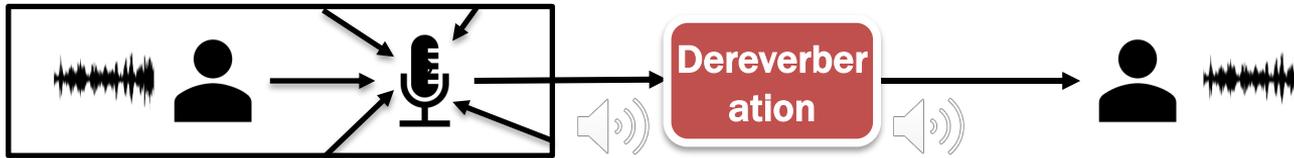
# Speech enhancement

## Several types of problems

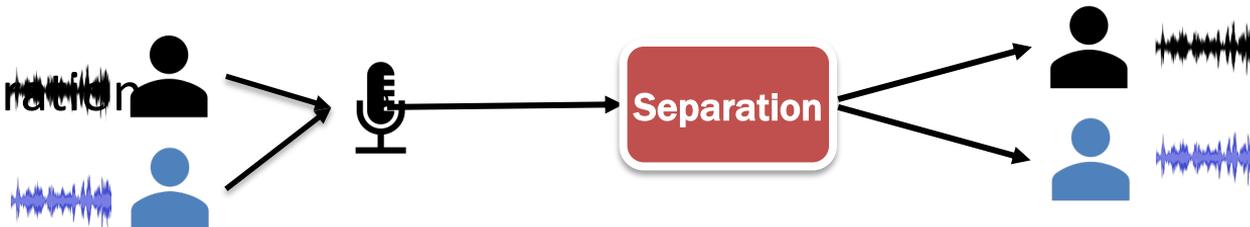
- Denoising (people mainly call it speech enhancement)



- Dereverberation



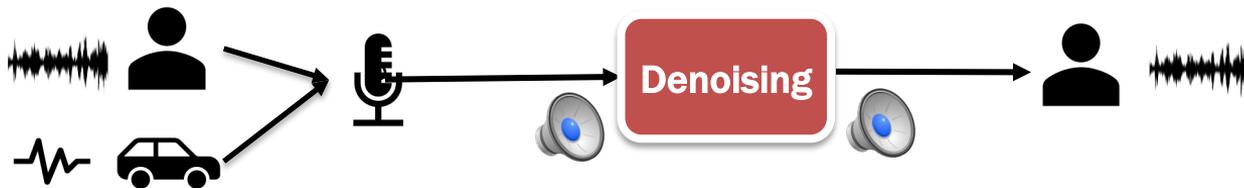
- Separation



# Speech enhancement

## Several types of problems

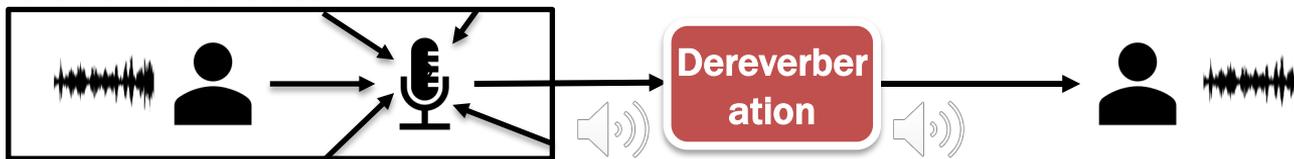
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# Speech enhancement

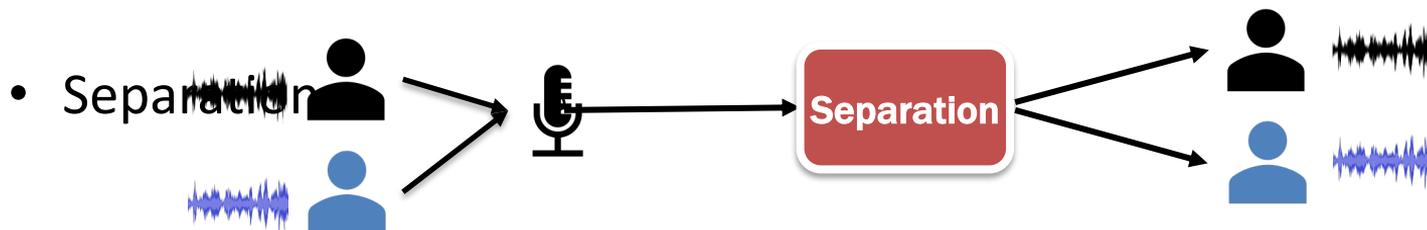
## Several types of problems

- Dereverberation

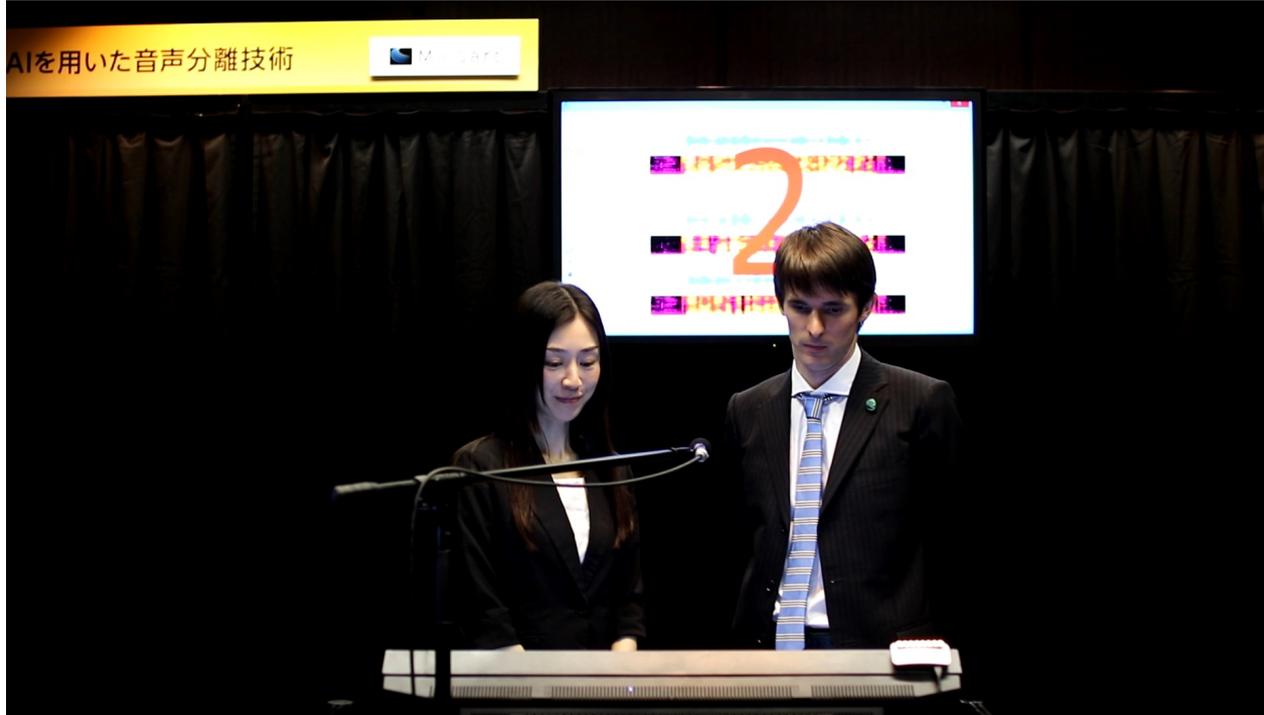


# Speech enhancement

## Several types of problems



# Deep clustering based speech separation [Hershey et al., 2016]



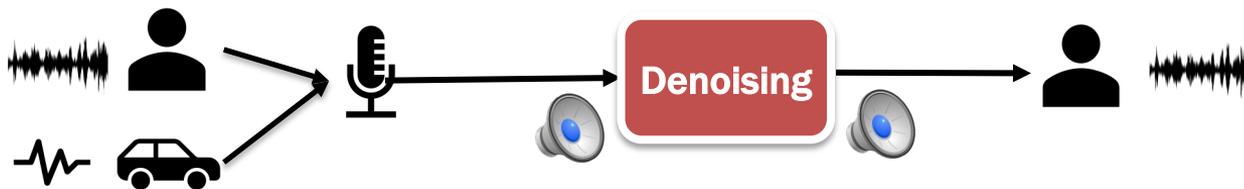
# How many microphones do we have?

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# Microphone array processing

## Single to multiple microphones

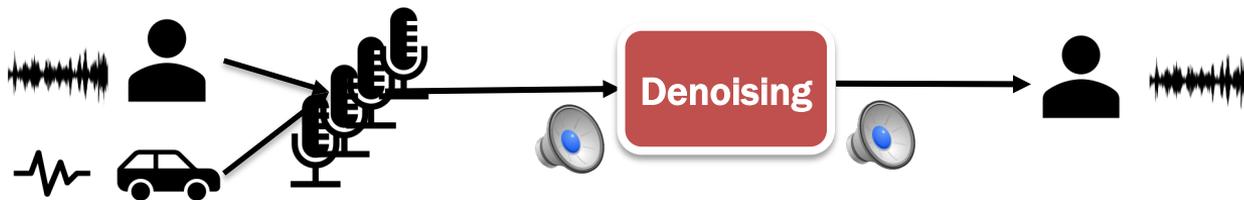
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# Microphone array processing

## Single to multiple microphones

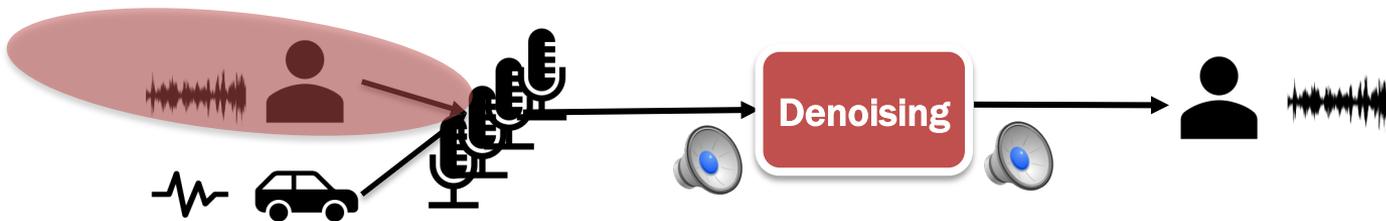
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# Microphone array processing

## Single to multiple microphones

- Denoising (people mainly call it speech enhancement)

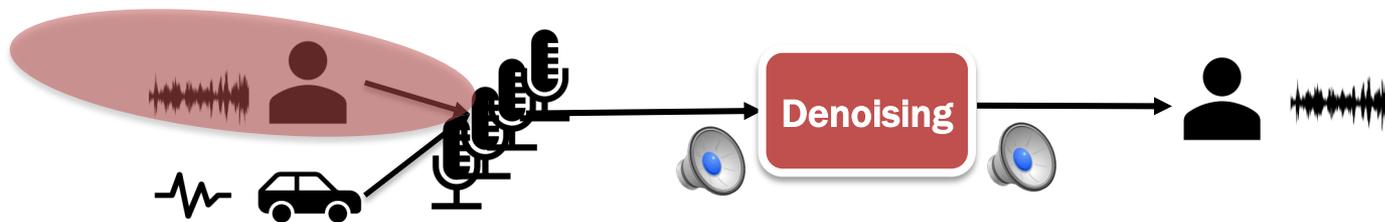


Make a spatial **beam** (beamforming)  
to only pick up desired signals

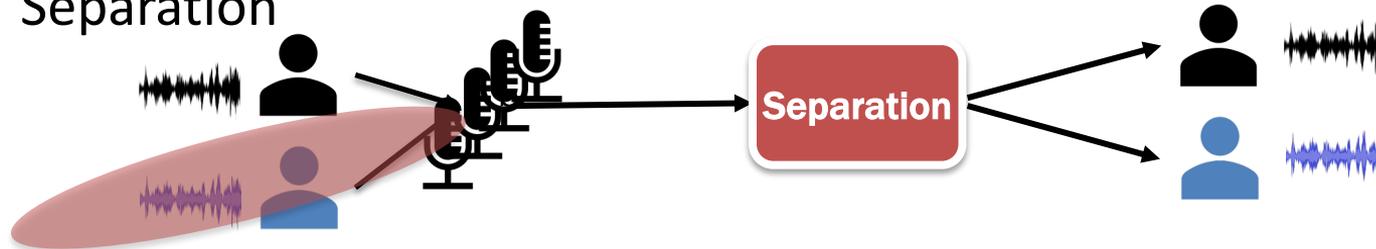
# Microphone array processing

## Single to multiple microphones

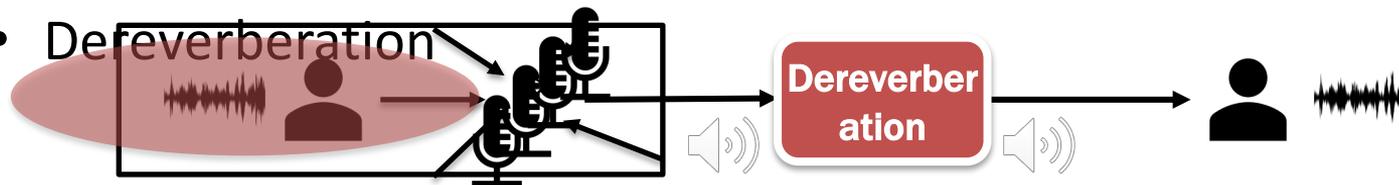
- Denoising (people mainly call it speech enhancement)



- Separation



- Dereverberation



# Cocktail party

- Many systems have more than one mic.
  - Alexa 7
  - Human 2
  - More microphones, easier to listen
- **Cocktail party**
  - Human can easily understand
  - One of the most difficult problem for a machine
- **One of the important speech research goal is to realize “who is speaking when what where how”**

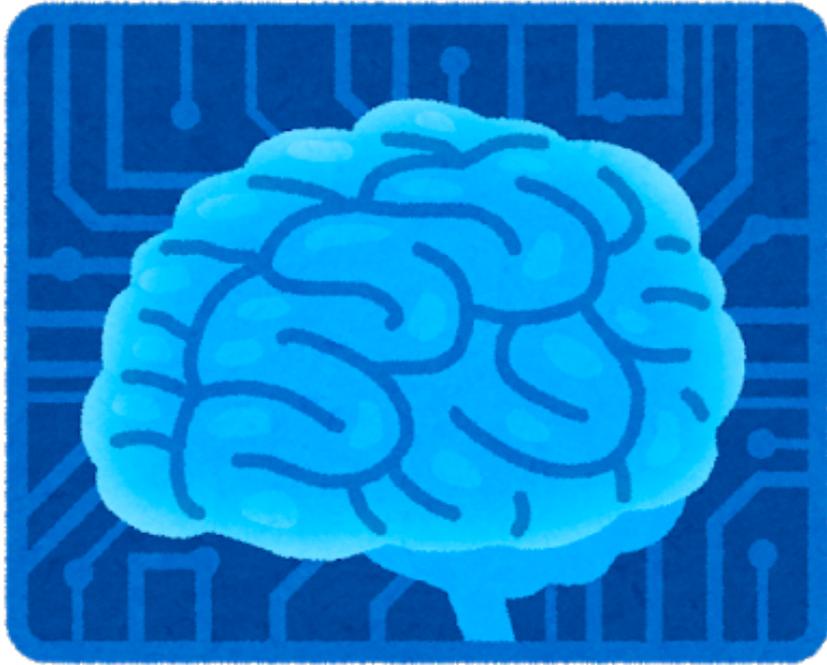


# What kind of research topics in speech research?

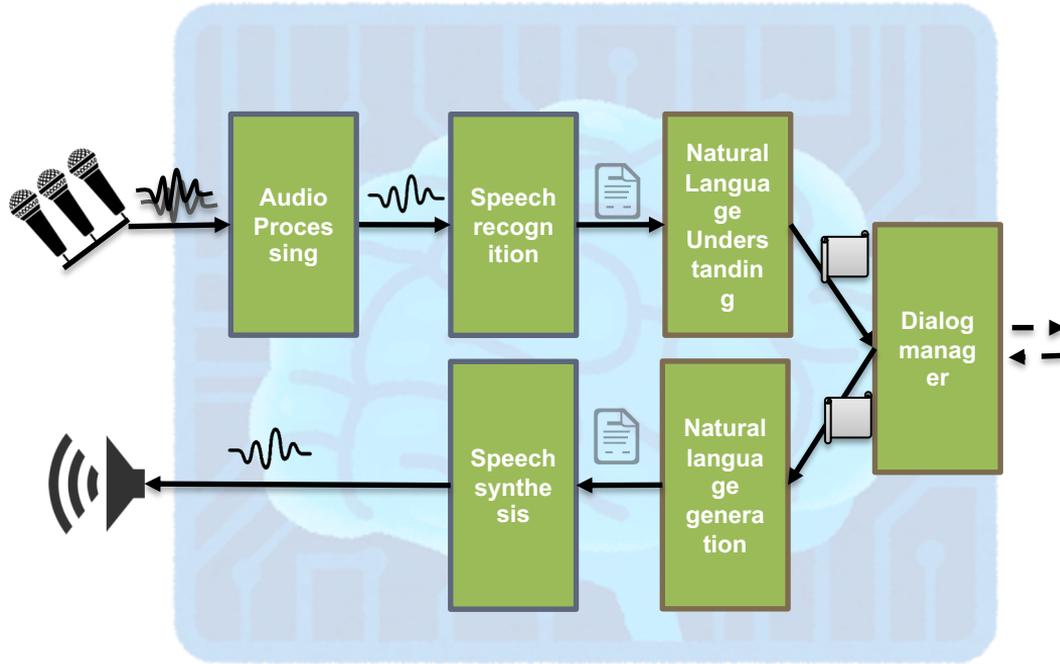
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- **Spoken dialogue systems**
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# My long-term research topic

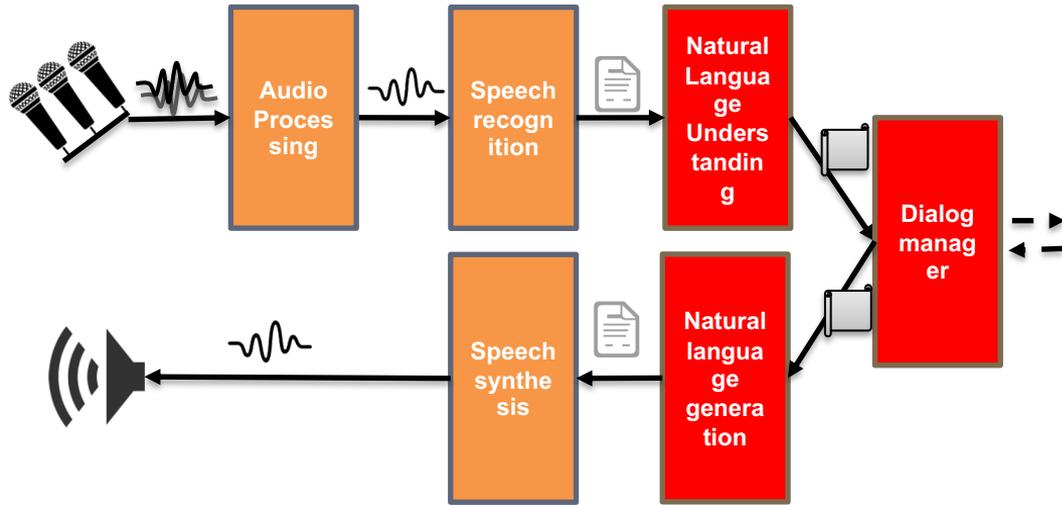
## Conversational AI



# Spoken dialog systems

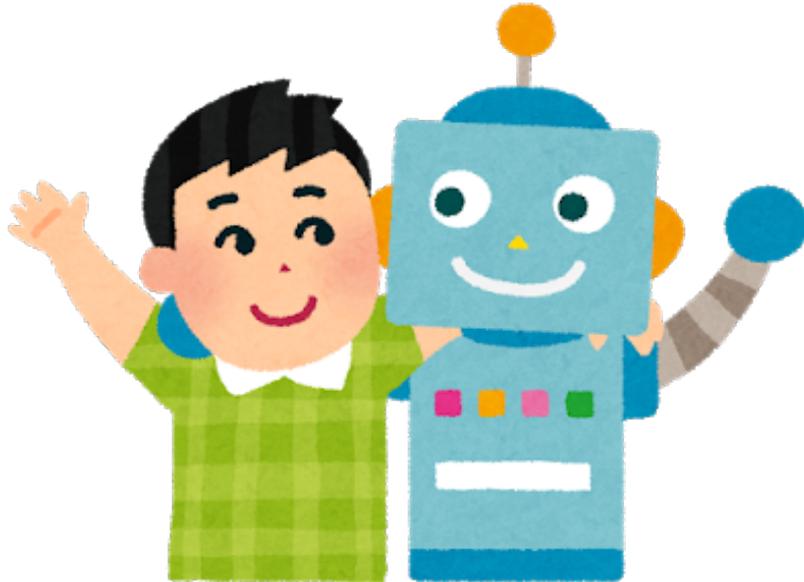


# Speech + Language!



# One of the ultimate speech research goals

- Human-level spoken dialog systems



# What kind of research topics in speech research?

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- What is speech?
- Speech applications
- **Speech databases**
- Speech hierarchy

# Speech variations

## Speaking styles and environments

	Style	Hours	Environment	Transcriber
Wall Street Journal (WSJ)	Read speech	~80	Clean/Close talk	Just confirm
Switchboard	Spontaneous	~300	Clean/Close talk	Have to transcribe
Librispeech	Read speech	~1,000	Clean/Close talk	Just confirm
<a href="#">CHiME-3</a>	Read Speech	~20	Noisy/Distant talk	Just confirm
<a href="#">CHiME-6</a>	Spontaneous	~50	Noisy/Distant talk	Have to transcribe



- Read speech: we prepare sentences in advance, and ask people to read them
  - Easy to obtain the reference
- Non-read speech (spontaneous): we have to transcribe by listening the audio, expensive

# Read speech examples

- Read a prompt
- We can make a pair data of a prompt and corresponding audio

Ex) common voice: <https://commonvoice.mozilla.org/en>

- Easy to collect
  - We still need to check whether the person can correctly utter a prompt
- Easy to anonymize
- Not a real conversation

# Spontaneous speech

- Transcribe actual recording
- Real, real, real
- Takes very long time to transcribe it
  - 2 minutes of the switchboard audio sample takes 30 minutes (for the beginner)
  - Need some postprocessing (anonymization, filler handling, etc.)

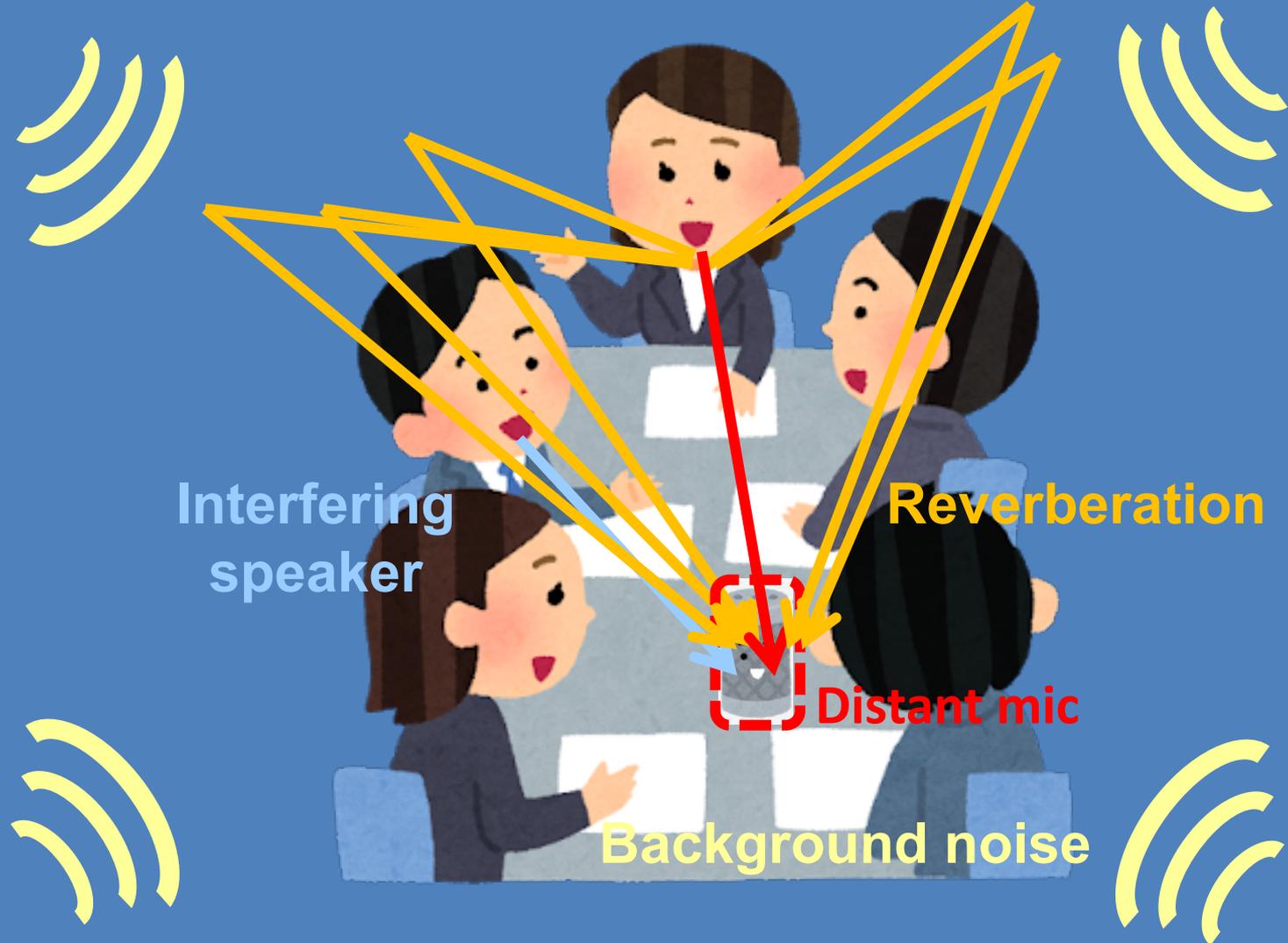
# Single speaker processing to conversation processing



Single speaker  
Close-talking microphone  
Error rate <5 %



Conversation analysis  
Distant microphone  
Error rate ~40%



Interfering speaker

Reverberation

Distant mic

Background noise

# CHiME-3

[http://spandh.dcs.shef.ac.uk/chime\\_challenge/chime2015/](http://spandh.dcs.shef.ac.uk/chime_challenge/chime2015/)



Cafe



Street



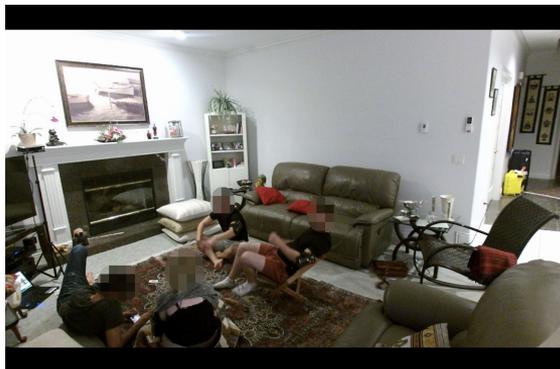
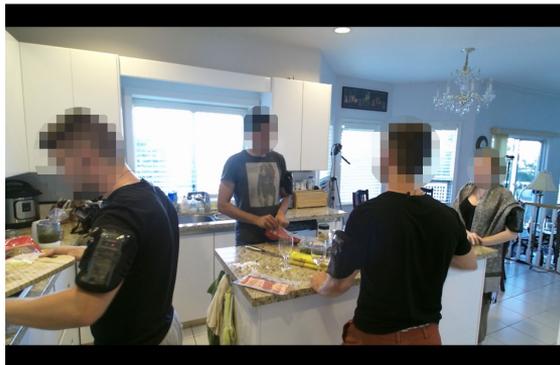
Bus



Pedestrian area

# CHiME-6 examples

<https://chimechallenge.github.io/chime6/>



# The CHiME-6 recording setup

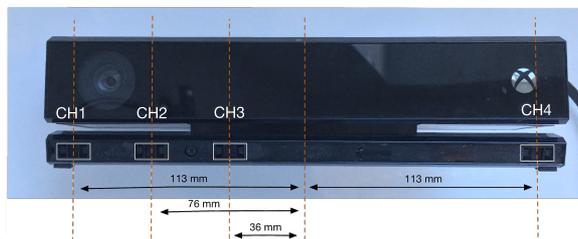
Data has been captured with 32 audio channels and 6 video channels

- Participants' microphones

- Binaural in-ear microphones recorded onto stereo digital recorders
- Primarily for transcription but also uniquely interesting data
- Channels: 4 x 2

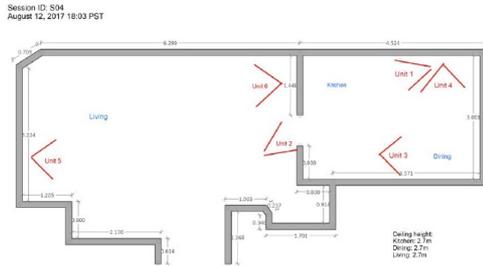
- Distant microphones

- Six separate Microsoft Kinect devices
- Two Kinects per living area (kitchen, dining, sitting)
- Arranged so that video captures most of the living space
- Channel: 6 x 4 audio and 6 video

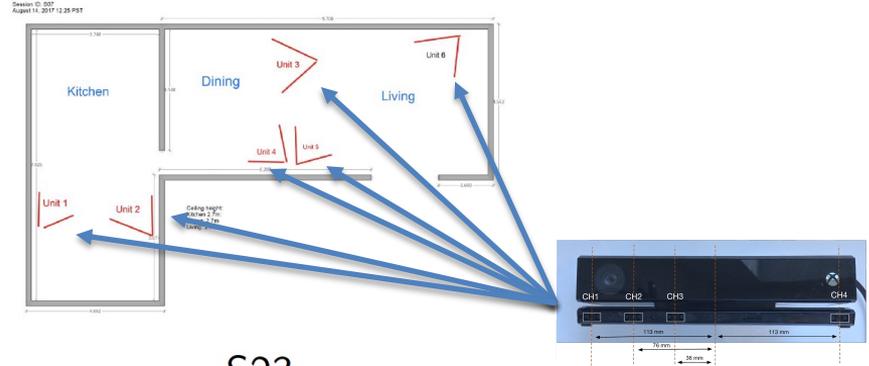


# Example recording setups

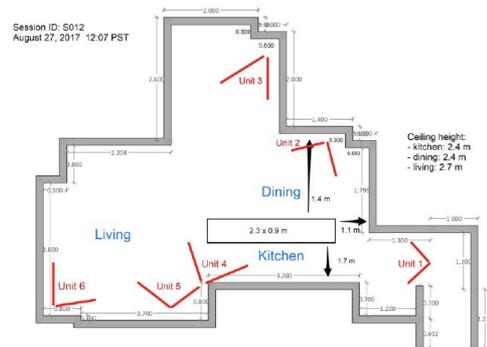
## S04



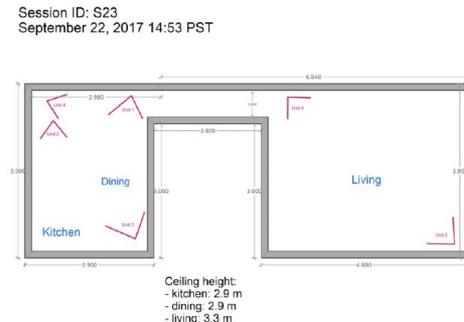
## S07



## S12



## S23



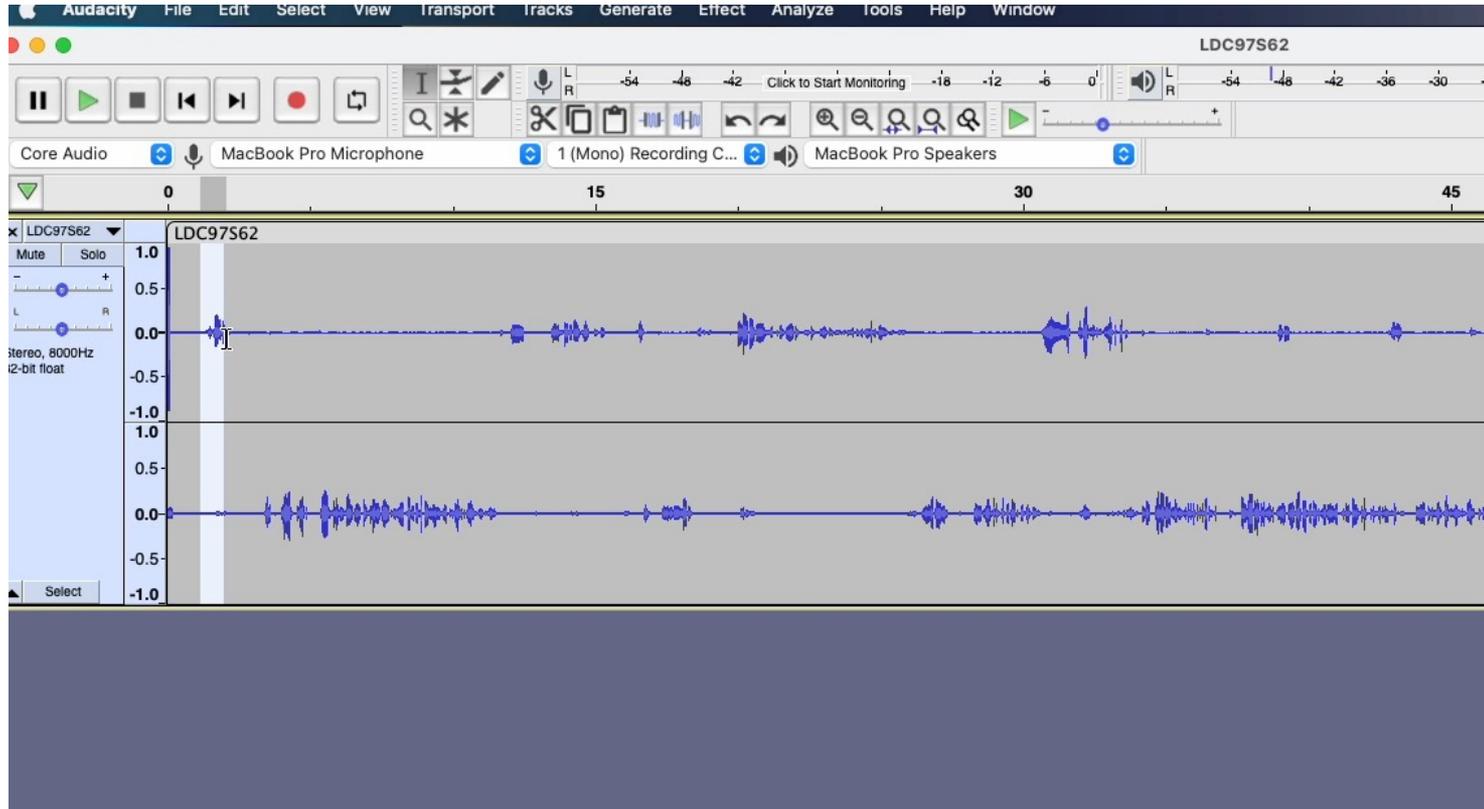
# Spontaneous speech

- Transcribe actual recording
  - Example based on Audacity (developed at CMU!)



- Real, real, real
- Takes very long time to transcribe it
  - 2 minutes of the switchboard audio sample takes 30 minutes (for the beginner)
- Need some postprocessing (anonymization, filler handling, etc.)

# How to transcribe an audio with Audacity?



# Where we found the speech data?

- LDC, ELRA, other university or government institution
  - <https://www ldc.upenn.edu/>
  - Well managed, license restricted
  - Famous ASR benchmarks (e.g., TIMIT, WSJ, Switchboard)
- Voxforge, openslr, commonvoice, zenodo
  - We can find less license restricted data (e.g., Creative Commons)
- Audio books, public recordings with captions (e.g., YouTube, Podcast, TED talk, Parliament or other government recordings, Bible)
  - Need some cares for the license and post processing
  - The data will be updated very frequently (deletion, modification, API change, etc.)
  - CMU Wilderness has **700(!)** languages (20 hours each)

# How many hours of training data do we need?

- We often use “**hour**” as a unit
- Commercial products: **More than thousand hours**
  - Very limited languages as public data, e.g., English, Mandarin, Japanese, German, Russian
- Do some ASR research experiments: **~100 hours**
- Less than 100 hours: Low-resource language in ASR
  - Pre-training/fine-tuning is changing the game

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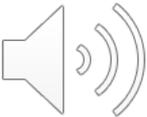
# Speech <-> Text

Speech sound: 



Text: I want to go to the CMU campus

# Speech <-> Phonem <-> Text

Speech sound: 



Phoneme: AY W AA N T T UW G OW T UW DH AH S IY EH M Y UW K AE M P AH S



Text: I want to go to the CMU campus

# What is phone and phoneme???

## GO TO: “g oʊ t u” or “G OW T UW”

- Phone: g oʊ t u
  - Devised by International Phonetic Association
  - Not applicable to all languages, needs special characters, too many variations, use of them depending on linguists
- Phoneme: one of the units of that distinguish one word from another in a **particular language**
  - /r/ and /l/ are degenerated in some languages (e.g., “rice” and “lice” sounds same for me!)
  - ARPAbet vs. International Phonetic Alphabet (IPA)
  - ARPAbet: G OW T UW
    - Proposed by ARPA for the development of speech recognition of only “American English”
    - Represented by ASCII characters

# Pronunciation dictionary

- CMU dictionary
  - <http://www.speech.cs.cmu.edu/cgi-bin/cmudict>

”I want to go to the CMU campus”

→ AY W AA N T T UW G OW T UW DH AH S IY EH M Y UW K AE M P AH  
S

- Powerful, but limited
- Out of vocabulary issue, especially new word
  - Grapheme2Phoneme mapping based on machine learning

# Let's play the CMU dictionary!

- Access: <http://www.speech.cs.cmu.edu/cgi-bin/cmudict>
- Find some in-vocabulary words
- Find five out-of-vocabulary words

# Multilingual phone dictionary

- [https://en.wiktionary.org/wiki/Wikt:Main\\_Page](https://en.wiktionary.org/wiki/Wikt:Main_Page)

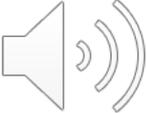
# Multilingual speech recognition (phone based)

- Try to split the problem from speech to phoneme and phoneme to text
- Speech to phone: **language independent (acoustic model)**
- Phone to phoneme, phoneme to word: **language dependent (lexicon model)**



- Build speech to phone based on universal acoustic model
- Linguistic knowledge to make a lexicon model

# Speech <-> Phonem <-> Text

Speech sound: 



Language  
**independent**

Phoneme: AY W AA N T T UW G OW T UW DH AH S IY EH M Y UW K AE M P AH S



Language  
**dependent**

Text: I want to go to the CMU campus

# Multilingual speech recognition (phone based)

- Try to split the problem from speech to phoneme and phoneme to text
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- Build speech to phone based on universal acoustic model
- Linguistic knowledge to make a lexicon model

# Other units?

- Syllable {C\*} V {C\*}
- Allophone: /k/ can be different depending on the context (/a/-)/k/(-/a/), (/a/-)/k/(-/i/)
- Pinyin
- Etc.

# Summary of today's talk

- Speech: sound waveform but used by human for the communication
- Speech applications: many applications
- Speech data: read vs. spontaneous, various sources
- Speech hierarchy: introduction of phone and phoneme
  
- The next lectures will introduce two main applications, ASR and TTS

# Assignment 3