

# Lecture 16: Forced Alignment, ASR

LING 1340/2340: Data Science for Linguists

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# Objectives

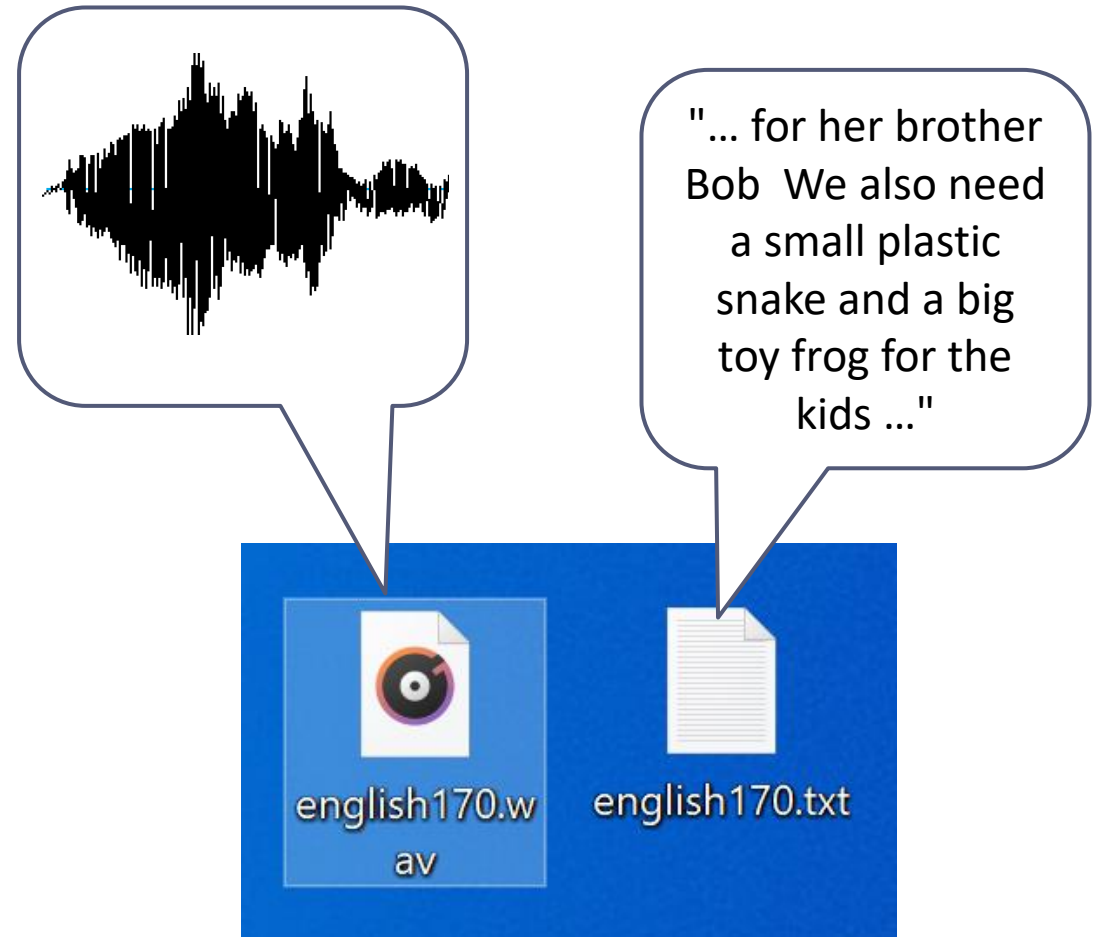
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- ▶ Forced alignment demo: Montreal Forced Aligner
- ▶ ASR demo: SpeechRecognition library
- ▶ ASR theory → Nope, next class

# Forced alignment

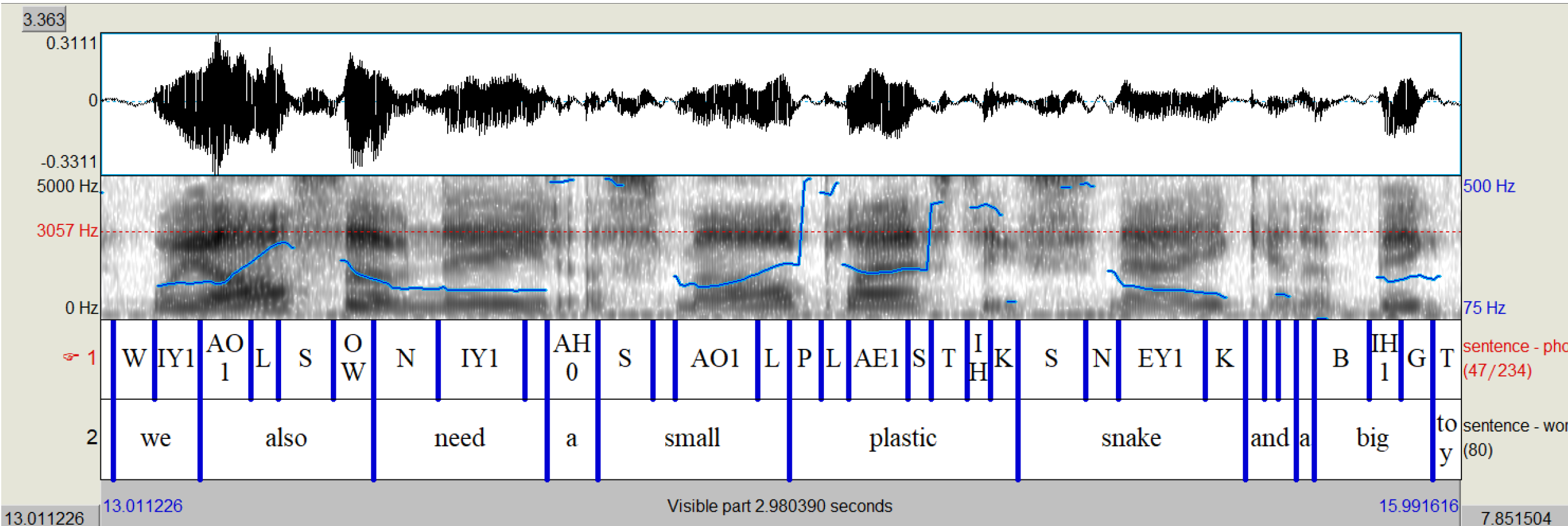
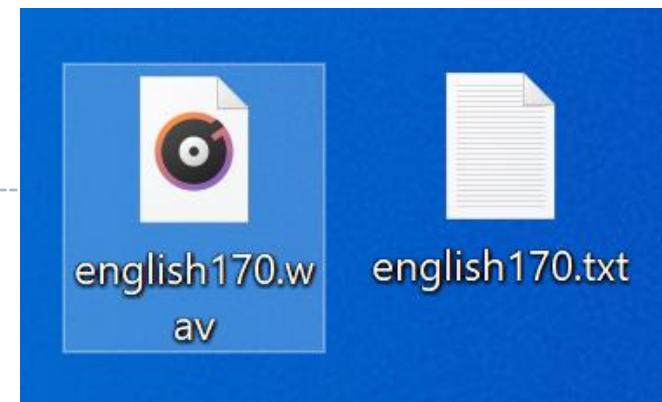
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- ▶ **"Forced alignment"**: automatic synchronization of a sequence of phones with an audio file.
- ▶ Purpose: speed up manual segmentation and annotation
  - ◆ Rather than manually creating phonetic transcription from scratch, correct output from forced aligner
  - ◆ Makes life easier for linguists doing speech-focused research!



# Forced alignment

- ▶ You have: a speech file (.wav), a transcript file (.txt) →
- ▶ You want:



# Sound wave, words, phones

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## ► What additional linguistic information is needed?

- ◆ Pronunciation dictionary
  - ◆ Phonemic representations for "brother", "we", "also"...
  - ◆ More broadly: **orthography** → **phone mapping** (G2P, "grapheme-to-phoneme")
- ◆ Acoustic model
  - ◆ How phonemic representation relates to sound wave



# Demo: Montreal Forced Aligner

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▶ Home page:

- ◆ <https://montreal-forced-aligner.readthedocs.io/en/latest/>

▶ GitHub project page:

- ◆ <https://github.com/MontrealCorpusTools/Montreal-Forced-Aligner>

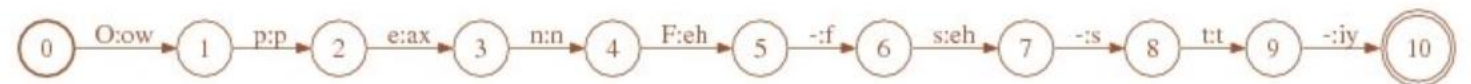
▶ Builds on popular/standard libraries:

- ◆ **Kaldi** ASR toolkit

- ◆ [\[home\]](#) [\[GitHub repo\]](#)

- ◆ which builds on **OpenFST**

- ◆ [\[home\]](#)



# Steps (latest MFA version 2.0)

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## ▶ Install Kaldi, MFA

- ◆ Windows users: For [ver 2.0, you need WSL](#) (Windows Subsystem for Linux, essentially Linux on Windows!) to use full G2P functionality. Alternatively: install [older ver 1.0.1 available here](#), which is Windows-native.

## ▶ Prepare data to align

- ◆ Speech files (WAV format, single-channel)
- ◆ Transcript files (.lab or .txt format; no punctuation)

We'll use TIMIT data for demo  
(pretend it came with audio files  
and .TXT transcripts only)

## ▶ Download language models (pre-trained, [MFA offers many](#))

- ◆ A pronunciation dictionary for the language
  - ◆ If not available: produce one by running language-specific G2P (grapheme-to-phoneme) on your transcript files
- ◆ An acoustic model for the language

## ▶ Run:

- ◆ `mfa align <input-dir> <pron-dict> <acoustic-model> <output-dir>`

## ▶ New TextGrid files in the output dir! Examine.

# Cleaning transcript files

MINGW64:/c/Users/narae/Desktop/true\_wav

narae@T480s MINGW64 ~/Desktop/FCJF0

```
$ cat *TXT
```

```
0 46797 She had your dark suit in greasy wash water all year.
0 34509 Don't ask me to carry an oily rag like that.
0 49460 Even then, if she took one step forward he could catch her.
0 45466 or borrow some money from someone and go home by bus?
0 57856 A sailboat may have a bone in her teeth one minute and lie becalmed the next.
0 24679 The emperor had a mean temper.
0 27751 How permanent are their records?
0 23143 The meeting is now adjourned.
0 36250 critical equipment needs proper maintenance.
0 39220 Tim takes Sheila to see movies twice a week.
```

Initial digits and  
punctuation need to go

narae@T480s MINGW64 ~/Desktop/FCJF0

```
$ perl -npe 's/^\d \d+ //' SA1.TXT
```

```
She had your dark suit in greasy wash water all year.
```

Perl + regular expressions  
to clean up

narae@T480s MINGW64 ~/Desktop/FCJF0

```
$ perl -npe 's/^\d \d+ //; s/\././g;' SA1.TXT
```

```
She had your dark suit in greasy wash water all year
```



```
narae@T480s MINGW64 ~/Desktop/FCJF0
```

```
$ perl -npe 's/^\d \d+ //; s/[\.\, \?]/ /g;' *.TXT
```

```
She had your dark suit in greasy wash water all year
```

```
Don't ask me to carry an oily rag like that
```

```
Even then if she took one step forward he could catch her
```

```
Or borrow some money from someone and go home by bus
```

```
A sailboat may have a bone in her teeth one minute and lie becalmed the next
```

```
The emperor had a mean temper
```

```
How permanent are their records
```

```
The meeting is now adjourned
```

```
Critical equipment needs proper maintenance
```

```
Tim takes Sheila to see movies twice a week
```

```
narae@T480s MINGW64 ~/Desktop/FCJF0
```

```
$ for x in *.TXT
```

```
> do
```

```
> perl -npe 's/^\d \d+ //; s/[\.\, \?]/ /g;' $x > ../true_wav/$x
```

```
> echo $x completed
```

```
> done
```

```
SA1.TXT completed
```

```
SA2.TXT completed
```

```
SI1027.TXT completed
```

```
SI1657.TXT completed
```

```
SI648.TXT completed
```

```
SX127.TXT completed
```

```
SX217.TXT completed
```

```
SX307.TXT completed
```

```
SX37.TXT completed
```

```
SX397.TXT completed
```

```
narae@T480s MINGW64 ~/Desktop/FCJF0
```

```
$ cd ../true_wav/
```

```
narae@T480s MINGW64 ~/Desktop/true_wav
```

```
$ ls
```

```
SA1.TXT SA2.TXT SI1027.TXT SI1657.TXT SI648.TXT SX127.TXT SX217.TXT SX307.TXT SX37.TXT SX397.TXT
```

```
SA1.WAV SA2.WAV SI1027.WAV SI1657.WAV SI648.WAV SX127.WAV SX217.WAV SX307.WAV SX37.WAV SX397.WAV
```

Use bash for-loop to  
create cleaned-up version  
of all .TXT files

.WAV and .TXT files are  
now ready...

# Download language models

## ► MFA's pre-trained models:

- ◆ [https://montreal-forced-aligner.readthedocs.io/en/latest/pretrained\\_models.html](https://montreal-forced-aligner.readthedocs.io/en/latest/pretrained_models.html)

### Pretrained acoustic models

As part of using the Montreal Forced Aligner in our own research, we have trained acoustic models for a number of languages. If you would like to use them, please download them below. Please note the dictionary that they were trained with to see more information about the phone set. When using these with a pronunciation dictionary, the phone sets must be compatible. If the orthography of the language is transparent, it is likely that we have a G2P model that can be used to generate the necessary pronunciation dictionary.

Any of the following acoustic models can be downloaded with the command `mfa download acoustic <language_id>`. You can get a full list of the currently available acoustic models via `mfa download acoustic`. New models contributed by users will be periodically added. If you would like to contribute your trained models, please contact Michael McAuliffe at [michael.e.mcauliffe@gmail.com](mailto:michael.e.mcauliffe@gmail.com).

Language	Link	Corpus	Number of speakers	Audio (hours)	Phone set
Arabic	<a href="#">Arabic acoustic model</a>	GlobalPhone	80	19.0	GlobalPhone
Bulgarian	<a href="#">Bulgarian acoustic model</a>	GlobalPhone	79	21.4	GlobalPhone
Croatian	<a href="#">Croatian acoustic model</a>	GlobalPhone	94	15.9	GlobalPhone
Czech	<a href="#">Czech acoustic model</a>	GlobalPhone	102	31.7	GlobalPhone
English	<a href="#">English acoustic model</a>	LibriSpeech	2484	982.3	Arpabet (stressed)
French (FR)	<a href="#">French (FR) acoustic model</a>	GlobalPhone	100	26.9	GlobalPhone

### Available pronunciation dictionaries

Any of the following pronunciation dictionaries can be downloaded with the command `mfa download dictionary <language_id>`. You can get a full list of the currently available dictionaries via `mfa download dictionary`. New dictionaries contributed by users will be periodically added. If you would like to contribute your dictionaries, please contact Michael McAuliffe at [michael.e.mcauliffe@gmail.com](mailto:michael.e.mcauliffe@gmail.com).

Language	Link	Orthography system	Phone set
English	<a href="#">English pronunciation dictionary</a>	Latin	Arpabet (stressed)
French	<a href="#">French Prosodylab dictionary</a>	Latin	Prosodylab French
German	<a href="#">German Prosodylab dictionary</a>	Latin	Prosodylab German

CMU pronouncing dictionary

```
narae@T480s: /mnt/c/Users/narae/Desktop
(base) narae@T480s:~$ conda activate aligner
(aligner) narae@T480s:~$ cd /mnt/c/Users/narae/Desktop/
(aligner) narae@T480s:/mnt/c/Users/narae/Desktop$ ls
FCJF0 desktop.ini english.dict english.zip tidy true_wav
(aligner) narae@T480s:/mnt/c/Users/narae/Desktop$ mkdir mfa out
(aligner) narae@T480s:/mnt/c/Users/narae/Desktop$ mfa align true_wav/ english.dict english.zip mfa_out/
All required kaldi binaries were found!
/home/narae/Documents/MFA/true_wav/align.log
INFO - Setting up corpus information...
loading from source
INFO - Number of speakers in corpus: 1, average number of utterances per speaker: 10.0
INFO - Number of speakers in corpus: 1, average number of utterances per speaker: 10.0
INFO - Parsing dictionary without pronunciation probabilities without silence probabilities
INFO - Creating dictionary information...
INFO - Setting up training data...
INFO - Generating base features (mfcc)...
INFO - Calculating CMVN...
INFO - Done with setup!
INFO - Performing first-pass alignment...
INFO - Calculating fMLLR for speaker adaptation...
INFO - Performing second-pass alignment...
INFO - All done!
(aligner) narae@T480s:/mnt/c/Users/narae/Desktop$ ls mfa_out/
true_wav_SA1.TextGrid      true_wav_SI1657.TextGrid  true_wav_SX217.TextGrid  true_wav_SX397.TextGrid
true_wav_SA2.TextGrid      true_wav_SI648.TextGrid   true_wav_SX307.TextGrid
true_wav_SI1027.TextGrid   true_wav_SX127.TextGrid   true_wav_SX37.TextGrid
(aligner) narae@T480s:/mnt/c/Users/narae/Desktop$
```

MFA is installed on WSL,  
need to bring out  
Ubuntu console



SUCCESS!  
New crop of TextGrid files

Praat Objects

Praat New Open Save Help

Objects:

- 3. Sound SA1
- 4. TextGrid true\_wav\_SA1

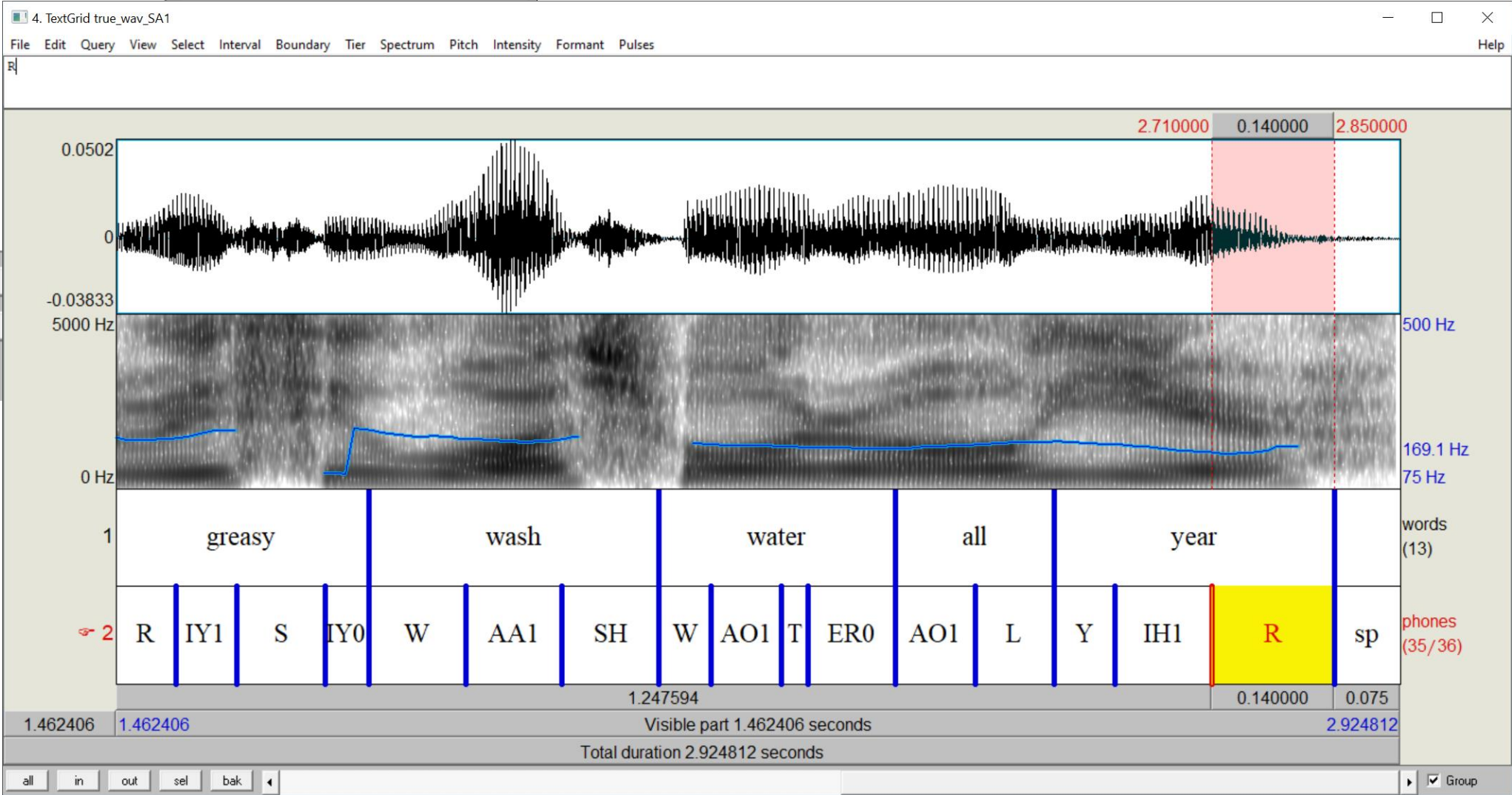
View & Edit

Draw...

Extract -

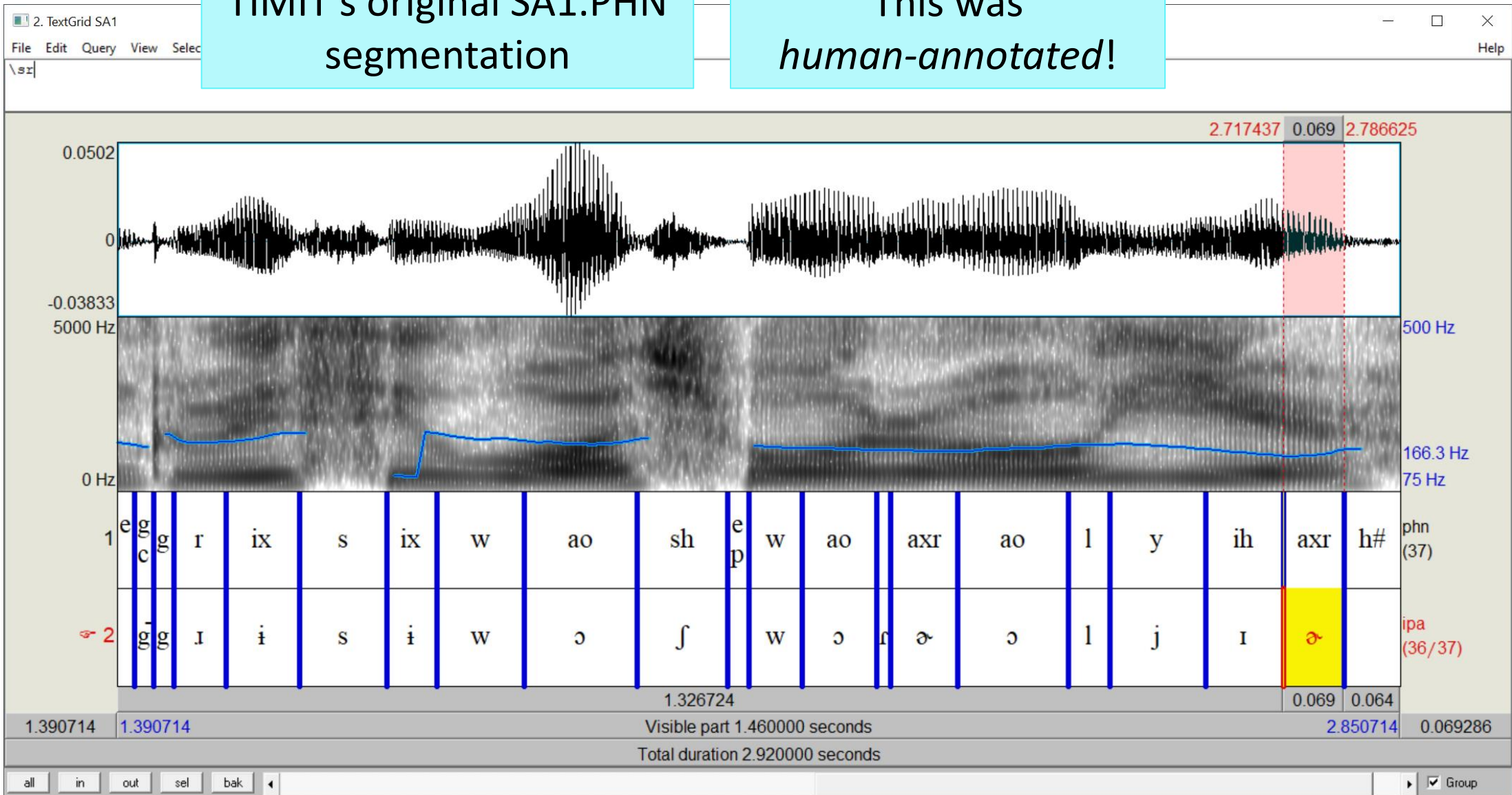
Modify TextGrid

Inspect the result in PRAAT.  
How did MFA do?



Compare with  
TIMIT's original SA1.PHN  
segmentation

This was  
*human-annotated!*

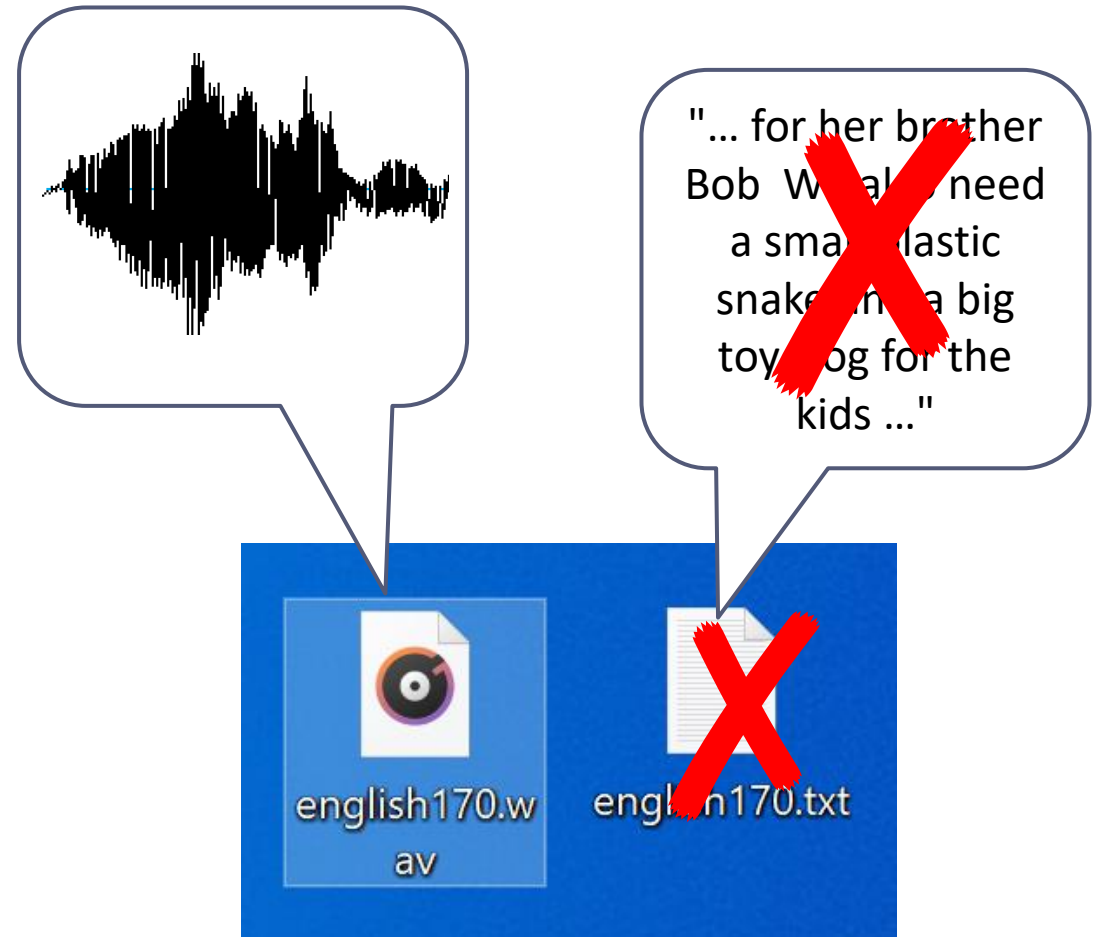




# But, what if we don't have a transcript?

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- ▶ Suppose all we have is audio files...
- ▶ Hire people to manually transcribe
- ▶ ... Or, go for ASR (automatic speech recognition)
  - ◆ Forced alignment is based on ASR.



# ARS demo

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- ▶ With SpeechRecognition library
- ▶ In Jupyter Notebook!

# Spoken language + Python

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## ▶ Praat in Python

- ◆ Libraries: [Praat-textgrids](#), [ParseImouth](#)

## ▶ [SpeechRecognition](#) library

- ◆ [https://github.com/Uberi/speech\\_recognition](https://github.com/Uberi/speech_recognition)
- ◆ Speech recognition module for Python, supporting several engines and APIs, online and offline.

## ▶ DataCamp course: Spoken Language Processing in Python

- ◆ <https://learn.datacamp.com/courses/spoken-language-processing-in-python>
- ◆ Libraries covered: [wave](#), [SpeechRecognition](#), [PyDub](#)



DataCamp subscription  
until Jul 10!



# Wrapping up

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## ▶ Next class:

- ◆ ELAN demo by Lindsey
- ◆ Project presentation: Misha
- ◆ Intro to ASR

## ▶ Final project submission:

- ◆ May 1 (Sun) 6pm
- ◆ If using 2-day late pass, email and LET ME KNOW before SUNDAY! (Final grade is due on Wed)