

Lecture 15: Speech Data

LING 1340/2340: Data Science for Linguists

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Objectives

- ▶ Grid search, parallel computing on CRC
 - ◆ Not going over in class!
- ▶ Speech data
 - ◆ Speech corpora, datasets
 - ◆ PRAAT
 - ◆ Command-line utilities, conversion
 - ◆ For loop in BASH!!

Grid search and parallel computing on CRC

- ▶ Grid search
 - ◆ How to discover best-performing parameters for your ML pipeline
- ▶ Parallel processing on CRC
 - ◆ How to utilize multiple computing nodes on CRC, build multiple ML models in parallel
- ▶ I have a Jupyter Notebook (to run on CRC) posted on GitHub:
 - ◆ https://github.com/Data-Science-for-Linguists-2022/Class-Exercise-Repo/blob/main/activity5_crc/gridsearch.ipynb
- ▶ Will not go over in class!
- ▶ Review only if it interests you or helps your project!

What to do with speech data?

- ▶ Analyze it directly.
 - ◆ Language identification
 - ◆ Phonetic research
 - ◆ Informing models (example below)
- ▶ Convert it to text, then text-process for downstream tasks
 - ◆ ASR (Automatic Speech Recognition) and ASU (... Understanding)
 - ◆ Automatic closed-captioning
- ▶ The other direction:
 - ◆ Speech Synthesis / Text-to-Speech (TTS)
 - ◆ Conversational Agents

Speech sounds: how to encode/represent?

- ▶ IPA, `ɒbvɪəsli...`

- ◆ But IPA chars are Unicode characters, difficult to use directly

- ▶ Do you remember CMU Pronouncing Dictionary?

```
>>> from nltk.corpus import cmudict
>>> prondict = cmudict.dict()
>>> prondict['anxious']
[['AE1', 'NG', 'K', 'SH', 'AH0', 'S'], ['AE1', 'NG', 'SH', 'AH0', 'S']]
>>>
```

- ◆ Uses **ARPABET**: <https://en.wikipedia.org/wiki/ARPABET>
 - ◆ ASCII-based representation of English speech sounds
- ◆ CMU pronouncing dict is used in all sorts of English speech technologies...
- ◆ Also: <https://heardle.glitch.me/>

Well-known speech datasets, corpora

- ▶ [Buckeye Corpus](#) (Pitt et al. 2005)
 - ◆ Python interface! <https://github.com/scjs/buckeye/blob/master/Quickstart.ipynb>
- ▶ [TIMIT](#) (Garofolo et al. 1993)
 - ◆ 10 sentences read by 630 speakers from 10 US dialect regions
 - ◆ Orthographic transcription and phonetic annotation
- ▶ [Switchboard corpus](#) (Godfrey et al. 1993, 1997)
 - ◆ Phone conversations between strangers on assigned topic, 2400 conversations by 543 speakers, many US dialects represented
- ▶ [TalkBank](#) corpora (MacWhinney, at CMU!)
 - ◆ Multiple research focus areas: L1 acquisition, multilingualism, etc.
 - ◆ Data contributed by many researchers
- ▶ [CORAAL](#) (Corpus of Regional African American Language)
 - ◆ Recorded speech from regional varieties of AAL, includes audio recordings along with time-aligned orthographic transcription, all downloadable

What do *linguists* do with speech data?

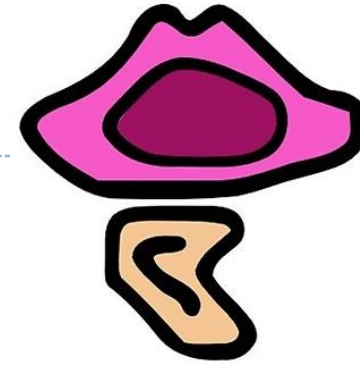
- ▶ Measuring duration: VOT (Voice Onset Time), etc.
- ▶ Measuring formants, F0/pitch
- ▶ Measuring amplitude, frequency
- ▶ Audio format conversion
 - ◆ WAV, MP3, FLAC
 - ◆ Channels, sampling rates, etc.
- ▶ Edit and manipulate sound
 - ◆ Crop, copy, slice, paste...
 - ◆ Manipulate pitch, duration...

What tool do we
use for these,
I wonder...?

PRAAT

<https://www.fon.hum.uva.nl/praat/>

- ▶ Everyone's favorite phonetics data analysis tool
- ▶ Venerable, powerful, versatile... and idiosyncratic
- ▶ Logo change was very much celebrated (or not...):
 - ◆ <https://blogs.umass.edu/linguist/2020/10/19/umass-redesign-of-praat-logo/>
- ▶ Using Praat for Linguistic Research, by Will Styler:
 - ◆ <https://wstyler.ucsd.edu/praat/>
- ▶ Paat Scripting Tutorial, by Eleanor Chodroff:
 - ◆ <https://eleanorchodroff.com/tutorial/PraatScripting.pdf>



Praat + TIMIT

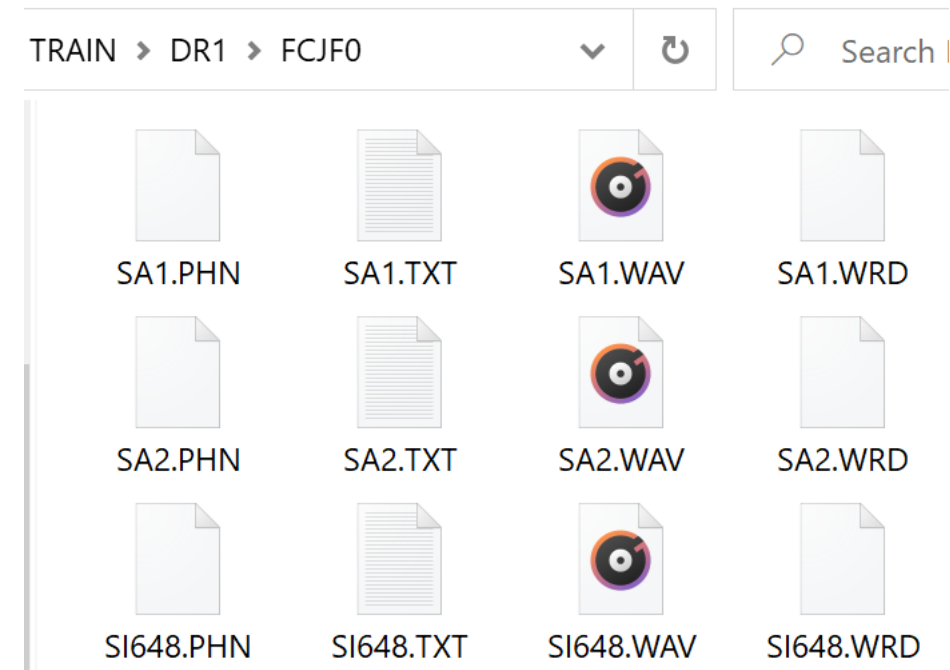
Activity
7 minutes



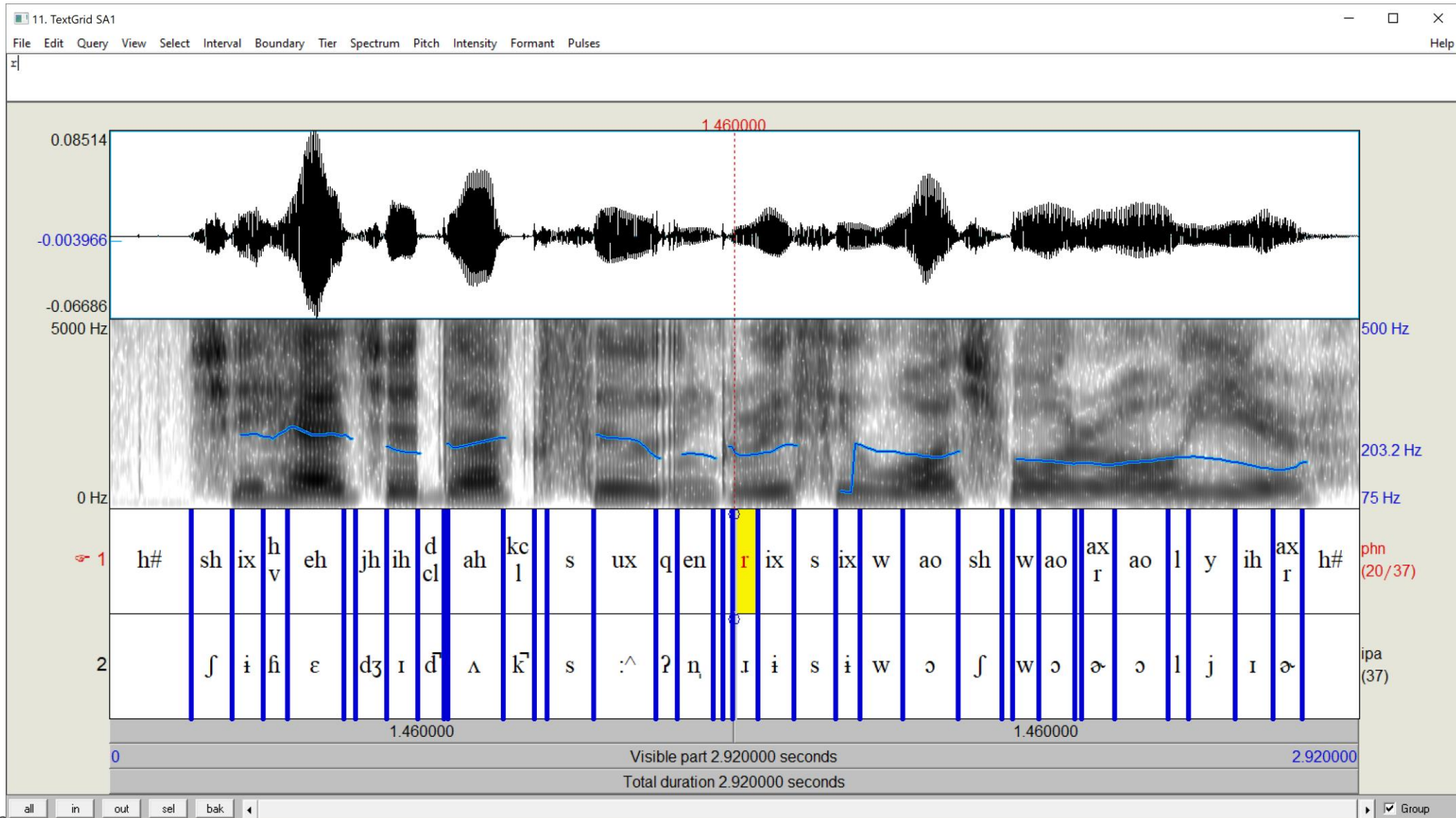
- ▶ An excerpt of TIMIT dataset is available on our GitHub org, in "Licensed-Datasets"
 - ◆ Get it by pulling from the repo.
- ▶ You probably have Praat on your laptop already
 - ◆ Pair up, open up "SA1.*" files in Praat, explore, see what you can do!
 - ◆ Also encouraged: command-line exploration

Open .WAV file first, and
then the rest after

Your will get warnings with
some txt files



TIMIT data in Praat



```
narae@T480s MINGW64 ~/Desktop/speech/TRAIN-DR1-FCJF0
$ ls
SA1.PHN SA2.PHN SI1027.PHN SI1657.PHN SI648.PHN SX127.PHN SX217.PHN SX307.PHN SX37.PHN SX397.PHN
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
SA1.WRD SA2.WRD SI1027.WRD SI1657.WRD SI648.WRD SX127.WRD SX217.WRD SX307.WRD SX37.WRD SX397.WRD
```

```
narae@T480s MINGW64 ~/Desktop/speech/TRAIN-DR1-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.
```

Utterance tier

```
narae@T480s MINGW64 ~/Desktop/speech/TRAIN-DR1-FCJF0
$ head SA1.PHN
0 3050 h#
3050 4559 sh
4559 5723 ix
5723 6642 hv
6642 8772 eh
8772 9190 dc1
9190 10337 jh
10337 11517 ih
11517 12500 dc1
12500 12640 d
```

Phone tier

```
narae@T480s MINGW64 ~/Desktop/speech/TRAIN-DR1-FCJF0
$ head SA1.WRD
3050 5723 she
5723 10337 had
9190 11517 your
11517 16334 dark
16334 21199 suit
21199 22560 in
22560 28064 greasy
28064 33360 wash
33754 37556 water
37556 40313 all
```

Word tier

TIMIT data in command-line

- ▶ Use **cat**, **less**, **grep**!

Essentially "Quick brown fox..." sentences for English speech sounds

TIMIT data in command-line

- ▶ Use `cat`, `less`, `grep`!

```
Jane Eyre@T480s MINGW64 ~/Documents/Data_Science/Licensed-  
s Speech Corpus/timit/TIMIT/TRAIN/DR1/FCJF0 (main)  
$ grep dh *PHN  
SA2.PHN:29000 29490 dh  
SI648.PHN:27613 28841 dh  
SI648.PHN:46640 46990 dh  
SX127.PHN:2231 2834 dh  
SX217.PHN:13785 14590 dh  
SX307.PHN:1960 2170 dh
```

Which files have /ð/ sound?

```
Jane Eyre@T480s MINGW64 ~/Documents/Data_Science/Licensed-  
s Speech Corpus/timit/TIMIT/TRAIN/DR1/FCJF0 (main)  
$ cat SA2.TXT  
0 34509 Don't ask me to carry an oily rag like that.
```

```
Jane Eyre@T480s MINGW64 ~/Documents/Data_Science/Licensed-  
s Speech Corpus/timit/TIMIT/TRAIN/DR1/FCJF0 (main)  
$ grep ae *PHN  
SA2.PHN:4600 6864 ae  
SA2.PHN:22266 24898 ae  
SA2.PHN:29490 32279 ae  
SI1027.PHN:41210 43040 ae  
SI648.PHN:12040 13800 ae  
SX127.PHN:10160 11640 ae
```

TextGrid

- ▶ Praat was able to parse TIMIT's PHN file format (phone tier)
 - ▶ Saving it out to a proper **TextGrid** file →
 - ▶ However, Praat couldn't handle:
 - ◆ SA1.TXT (utterance tier)
 - ◆ SA1.WRD (word tier)
- ← How to get them into TextGrid?

There's a python library (or two) for that!

praat-textgrids 1.3.1

`pip install praat-textgrids` 

 **[ˈpɑː.səl,mɑʊθ]**

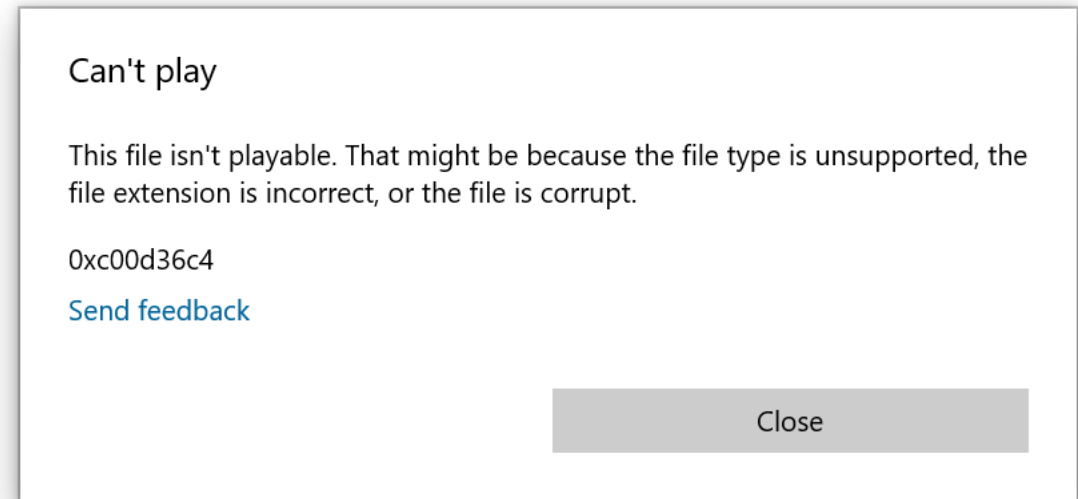
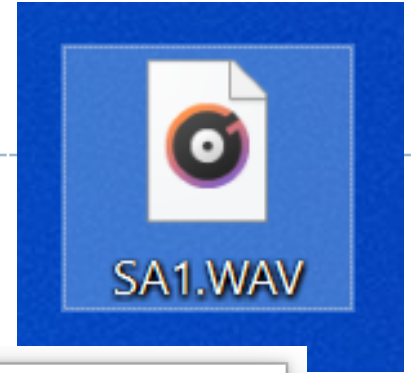
Parselmouth – Praat in Python, the Pythonic way

```
File type = "ooTextFile"
Object class = "TextGrid"

xmin = 0
xmax = 2.92
tiers? <exists>
size = 2
item []:
  item [1]:
    class = "IntervalTier"
    name = "phn"
    xmin = 0
    xmax = 2.92
    intervals: size = 37
    intervals [1]:
      xmin = 0
      xmax = 0.19062500000000002
      text = "h#"
    intervals [2]:
      xmin = 0.19062500000000002
      xmax = 0.2849375
      text = "sh"
    intervals [3]:
      xmin = 0.2849375
      xmax = 0.3576875
      text = "ix"
    intervals [4]:
      xmin = 0.3576875
      xmax = 0.415125
      text = "hv"
    intervals [5]:
      xmin = 0.415125
      xmax = 0.54825
      text = "eh"
    intervals [6]:
```

.WAV format?

- ▶ Also, even though PRAAT was able to open the .WAV files, Windows 10 cannot...
- ▶ These files are not really .WAV...
 - ◆ **SPHERE format**, normally with **.SPH** extension.
- ▶ How to convert to WAV?



Solution 1:

Praat script

► Write a praat script

- ◆ ([Or, grab someone else's...](#))

```
# prep_audio_mfa.praat
# Written by E. Chodroff
# Oct 23 2018
# extract left channel and resample to 16 kHz for all wav files in a directory

### CHANGE ME!
# don't forget the slash at the end of the path
dir$ = "/Users/Eleanor/Desktop/align_input/"
###

Create Strings as file list: "files", dir$ + "*.wav"
nFiles = Get number of strings

for i from 1 to nFiles
    # read in WAV file
    selectObject: "Strings files"
    filename$ = Get string: i
    Read from file: dir$ + filename$

    # extract left channel
    Extract one channel: 1

    # resample to 16kHz with 50 point precision (default)
    Resample: 16000, 50

    # save WAV file
    Save as WAV file: dir$ + filename$

    # clean up
    select all
    minusObject: "Strings files"
    Remove
endfor
```


Solution 2:

SoX + bash shell

```
sox <input-file> -b 16 -t wav <output-file>
```

Declared as `x`,
subsequent
references as `$x`

```
for x in *.WAV  
do  
sox $x -b 16 0t wav true_wav/$x  
echo $x finished  
done
```

```
narae@T480s MINGW64 ~/Desktop/FCJF0  
$ alias sox="/d/util/sox-14.4.2/sox.exe"  
  
narae@T480s MINGW64 ~/Desktop/FCJF0  
$ ls  
SA1.PHN  SA2.WAV      SI1657.PHN  SI648.WAV  SX217.PHN  SX307.WAV  SX397.PHN  
SA1.TXT  SA2.WRD      SI1657.TXT  SI648.WRD  SX217.TXT  SX307.WRD  SX397.TXT  
SA1.WAV  SI1027.PHN  SI1657.WAV  SX127.PHN  SX217.WAV  SX37.PHN   SX397.WAV  
SA1.WRD  SI1027.TXT  SI1657.WRD  SX127.TXT  SX217.WRD  SX37.TXT   SX397.WRD  
SA2.PHN  SI1027.WAV  SI648.PHN   SX127.WAV  SX307.PHN  SX37.WAV   true_wav/  
SA2.TXT  SI1027.WRD  SI648.TXT   SX127.WRD  SX307.TXT  SX37.WRD  
  
narae@T480s MINGW64 ~/Desktop/FCJF0  
$ sox SA1.WAV -b 16 -t wav true_wav/SA1.wav  
  
narae@T480s MINGW64 ~/Desktop/FCJF0  
$ ls true_wav/  
SA1.wav
```

converting a single file

```
narae@T480s MINGW64 ~/Desktop/FCJF0  
$ for x in *WAV  
> do  
> sox $x -b 16 -t wav true_wav/$x  
> echo $x finished  
> done  
SA1.WAV finished  
SA2.WAV finished  
SI1027.WAV finished  
SI1657.WAV finished  
SI648.WAV finished  
SX127.WAV finished  
SX217.WAV finished  
SX307.WAV finished  
SX37.WAV finished  
SX397.WAV finished  
  
narae@T480s MINGW64 ~/Desktop/FCJF0  
$ ls true_wav/  
SA1.wav  SI1027.WAV  SI648.WAV  SX217.WAV  SX37.WAV  
SA2.WAV  SI1657.WAV  SX127.WAV  SX307.WAV  SX397.WAV
```

for loop in bash!

Command-line conversion to mp3: with ffmpeg

► `ffmpeg -i input.wav output.mp3`

```
MINGW64:/c/Users/narae/Desktop/speech/TRAIN-DR1-FCJF0
narae@T480s MINGW64 ~/Desktop/speech/TRAIN-DR1-FCJF0
$ ls
SA1.PHN  SA2.TXT      SI1027.WAV  SI1657.WRD  SX127.PHN  SX217.TXT  SX307.WAV  SX37.WRD
SA1.TXT  SA2.WAV      SI1027.WRD  SI648.PHN   SX127.TXT  SX217.WAV  SX307.WRD  SX397.PHN
SA1.WAV  SA2.WRD      SI1657.PHN  SI648.TXT   SX127.WAV  SX217.WRD  SX37.PHN   SX397.TXT
SA1.WRD  SI1027.PHN  SI1657.TXT  SI648.WAV   SX127.WRD  SX307.PHN  SX37.TXT   SX397.WAV
SA2.PHN  SI1027.TXT  SI1657.WAV  SI648.WRD   SX217.PHN  SX307.TXT  SX37.WAV   SX397.WRD

narae@T480s MINGW64 ~/Desktop/speech/TRAIN-DR1-FCJF0
$ alias ffmpeg='/d/util/ffmpeg-4.3.2-2021-02-02-essentials_build/bin/ffmpeg.exe'

narae@T480s MINGW64 ~/Desktop/speech/TRAIN-DR1-FCJF0
$ ffmpeg -i SA1.WAV SA1.mp3
ffmpeg version 4.3.2-2021-02-02-essentials_build-www.gyan.dev Copyright (c) 2000-2021 the FFmpeg
g developers
built with gcc 10.2.0 (Rev6. Built by MSYS2 project)
```

But, can we do this
with *every* wav file...?

For loop in bash

```
for x in *.WAV
do
fname=`basename $x .WAV`
newname=$fname.mp3
echo $newname
ffmpeg -i $x converted/$newname
done
```

x is newly created as a variable

subsequent mention of \$x requires \$

We want new file names with .mp3 extension

Loop content begins with do and ends with done

General-purpose audio/video manipulation software

▶ Audacity

- ◆ Open-source audio software



▶ SoX

- ◆ Sound eXchange; audio format conversion tool

Powerful
command-line tools!!

▶ FFmpeg

- ◆ For recording and converting audio/video data

https://musicinformationretrieval.com/sox_and_ffmpeg.html

Popular speech data analysis tools for linguists (1)

- ▶ [Praat](#) (Boersma & Weenink, 2021)
- ▶ [Klatt formant synthesizer](#) (Klatt 1975, 1984)
- ▶ Forced aligners
 - ◆ [Penn Phonetics Lab Forced Aligner](#) (Yuan & Liberman 2009) → legacy, became FAVE-align
 - ◆ [FAVE-align](#) (Rosenfelder et al. 2011)
 - ◆ [Montreal Forced Aligner](#) (McAuliffe et al. 2017) ← we'll take a look
 - ◆ [EasyAlign](#) (Goldman 2011 -- Windows only)
- ▶ [ELAN](#) multimodal annotator (Wittenberg et al. 2006)
 - ◆ Audio as well as video!

Popular speech data analysis tools for linguists (2)

Some tools are online:

- ▶ [NORM](#): the Vowel Normalization and Plotting Suite
- ▶ [DARLA](#): Dartmouth Linguistic Automation

← You upload an audio file and a transcript file, the site will process them and email you the results, etc!

Wrapping up

- ▶ Next class:
 - ◆ Forced alignment overview
 - ◆ Quick survey: speech data processing in Python
 - ◆ Primer on ASR
- ▶ 3rd progress report due on Tuesday
- ▶ Also coming up: project presentations