Using open source speech recognition software without an American accent.

#### The field

### Julius: C, strange BSDish license, Japanese

#### HTK: (HMM (Hidden Markov Model) Toolkit) C, non-commercial license

Kaldi: C++ library, Apache 2.0 (F.U. HTK)

CMU Sphinx Family: C, Java, BSD-ish

Smaller possibly dormant projects

Shout: C++, GPL

RWTH ASR: C++, non-commercial, German

latros: C, GPL3, Spanish

Carnegie Mellon University Sphinx

Sphinx **C** 1980s

Sphinx 2 **C** 1990s -----

Sphinx 3 **C** 2000s

Sphinx 4 Java

PocketSphinx C

Sphinx 3, 4, and PocketSphinx share tools.

## Sphinx development model

# PhD driven: 1. Have a moderately good idea 2. Write code that "proves" your idea 3. Submit thesis 4. \$\$\$\*

\* \$\$\$ refers to a job at Nuance, Apple, Microsoft, Google, IBM,...

# The "Coral Reef" development model.

What Americans do:

sudo aptitude install \
 pocketsphinx-hmm-wsj1 \
 pocketsphinx-lm-wsj \
 python-pocketsphinx \
 # or gstreamer0.10-pocketsphinx

from pocketsphinx import Decoder

HMM = "/usr/share/pocketsphinx/model/hmm/wsj1/"
LM = "/usr/share/pocketsphinx/model/"\
 "lm/wsj/wlist50.3e-7.vp.tg.lm.DMP"
DICT = "/usr/share/pocketsphinx/model/"\
 "lm/wsj/wlist50.dic"

decoder = Decoder(hmm=HMM, lm=LM, dict=DICT)

fh = open("speech.wav")
fh.seek(44) # skip the WAV header
decoder.decode\_raw(fh)
print decoder.get\_hyp() # short for hypothesis

le-phone words

- NFO: ngram\_search\_fwdtree.c(195): Cr NFO: ngram\_search\_fwdtree.c(203): 0
- le-phone words
- WFO: ngram\_search\_fwdtree.c(325): ma
  WFO: ngram\_search\_fwdtree.c(334): 44
  % single-phone words
- VFO: ngram search fwdflat.c(95): fwd

### A good reference is David Huggins-Daines' live-coding PyCon2010 lightning talk.

#### 1 [11]: fh.seek(44)

Everything goes wrong and he still FO: cm finishes in under 4 minutes. -0.37 0.09 0.63 0.05 0.13 0.07 0.11 0.10 WFO: ngram\_search.c(368): Resized backpointer table to 10000 entries WFO: ngram\_search.c(376): Resized score stack to 200000 entries WFO: ngram\_search.c(368): Resized backpointer table to 20000 entries WFO: ngram\_search.c(376): Resized score stack to 400000 entries WFO: ngram\_search.c(376): Resized score stack to 800000 entries WFO: ngram\_search.c(368): Resized score stack to 800000 entries WFO: ngram\_search.c(368): Resized backpointer table to 40000 entries WFO: ngram\_search.c(368): Resized backpointer table to 40000 entries The configuration lines:

HMM = ".../hmm/wsj1/"
LM = ".../lm/wlist50.3e-7.vp.tg.lm.DMP"
DICT = ".../lm/wsj/wlist50.dic"

HMM Acoustic model (Hidden Markov Model)
LM Language model
DICT Pronunciation dictionary

# The *Acoustic model* matches sounds to phoneme probabilities.

# It looks at the flow, not each sound in isolation.



Model

sequences

The *Language model* rates the probability of a sequence of words.

**X** "Their is moor then won weight hoodoo it" **Y** "There is more than one way to do it"

The acoustic and language models talk to each other through the *dictionary*.

The dictionary maps phoneme sequences to words.

PERKY P ER K IY PERL P ER L PERLA P ER L AH WIND WIND(2) W AY N D WIH N D

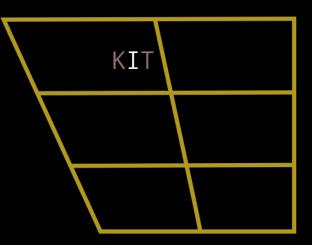
Extracts from the 130,000 words, 39 phoneme CMU dictionary of General American English. The dictionary is a blunt instrument.

The sounds of words adapt to the context.

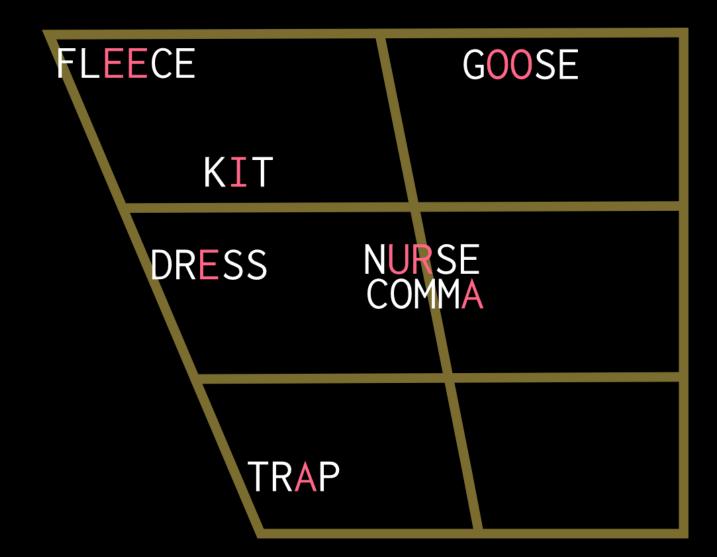
# Vowel charts

# The symbol position shows the tounge position

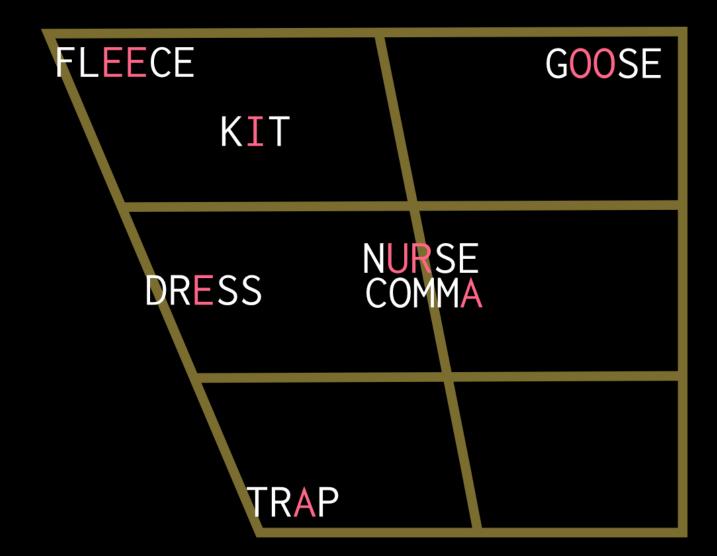
KIT represents a *lexical set* A bit like IPA's phonemic /I/ (but *not* IPA phonetic [I]) *fish* and *chips* use the KIT vowel



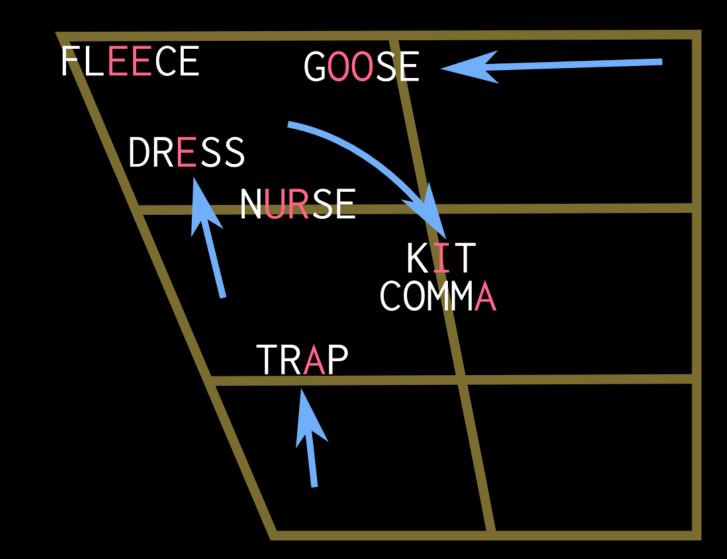
# A selection of General American vowels



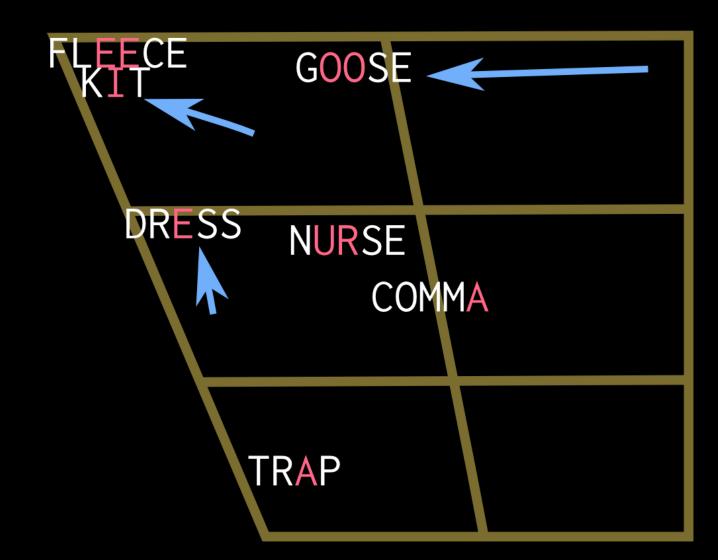
# The same vowels in RP (high class British)



# New Zealand English



# Australian English



General American English

*Father* and *bother* rhyme (but *farther* doesn't).

*Caught* sounds like *cot*, not *court*.

*Marry* and *merry* are indistinguishable.

Pants rhymes with Glance, Hurry with Furry.

*Panda* and *pander* sound different (American English is rhotic).

New Zealand English

*bear* and *beer* are <del>becoming</del> indistinguishable

The vowel in *fish* is the vowel in *the*.

Otherwise it makes similar distinctions to RP, but not sounds.

#### How do we get a New Zealand English model?

For the acoustic model there are two methods:

1. Adapt an existing one

2. Create a new one

Existing language models *might* be OK.

Adapting an acoustic model

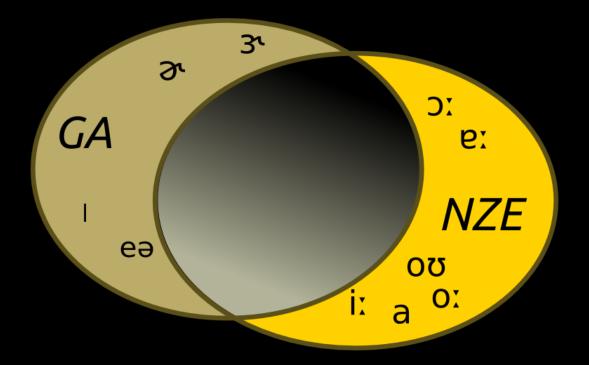
Read http://cmusphinx.sf.net/wiki/tutorialadapt

Requires a "small" corpus of transcribed speech.

The *phoneme set* is fixed.

The CMU dictionary uses 39 phonemes.

## New Zealand English uses about 44 ... but not all of CMU's 39.



## An adaptation using the CMU dictionary

can learn mergers (at a cost)
can't make new distinctions

Effectively down to 35 or so phonemes

# Useless duplication:BEARBEHR#BEHBEERBIHR#BEH

### wrongly conjoined: FATHER F AA DH ER BOTHER B AA DH ER # B <u>OH</u> DH <u>AX</u>

Just wrong: TUNE T UW N # T Y UW N Squeezing NZE into the CMU phoneset A new dictionary could remap vowels: F AA DH IH B AO DH IH # ?? AO is the PAW vowel # IH is the KIT vowel.

There is no good way to do it.

# NZE pronouncing dictionaries

# There are none.

## British English pronouncing dictionaries

- · CELEX proprietary
- COALD, beep, Oxford 710 non-commercial
- UNISYN non-commercial, multi-accent

Generating a dictionary via grapheme-to-phoneme systems

Espeak, Festival, FreeTTS read English

Espeak specialises in British English

Espeak speaks IPA

```
$ espeak -q --ipa 'father bother'
f'aːðə b'bðə
```

Espeak is rule based, so it is never stuck for an answer:

\$ espeak -q --ipa 'Wellington Perlmongers'
w'ɛlɪŋtən p'ɜːlmʌŋgəz
("p'ɜːlmʌŋəz" might be better)

#### Transcribed speech

# Sphinx likes to be trained with short (5 to 30 second) snippets of speech.

#### There is a control file a bit like this:

COLENSO SAID THAT HE COULDN'T ++UM++ PERSUADE (DGI038-auto-series-2-033) OUR HEARTS AND OUR BACKS TO HOIST ANCHOR (LunaTick-20080319-ill-mfc-a0411) SAVINGS IN THAT YEAR WERE NAUGHT POINT TWO (questions-20120802-wav-0007)

#### and audio file with matching names.

Accurate transcribing is slow

Slower than recording.

Cutting up long transcriptions is slower than recording but faster than transcribing.

# Sources of transcribed speech

- Voxforge
- Wellington Corpus
- · Canterbury Corpus
- Hansard
- TV and radio

http://voxforge.org aims to create GPL speech models.

It has about an hour of NZE from about 12 volunteers reading prompts Wellington Corpus of Spoken English

A million spoken words (half natural conversation)

Recorded and transcribed in the 1990s by VUW linguistics students

Imprecise timing information

On-site access only.

Canterbury Corpus

About 150 hours of natural conversation

Transcribed in 2000s by Canterbury students with software help

Good timing information

On-site access only.

Both Wellington and Canterbury linguistics people are keen to help

Canterbury have resources, namely Robert Fromont

VUW are skint.

#### Hansard OK, but prone to acronym expansion, missed interjections, and correction.

MP: "to the minister: which assets identified by CERA..." Hansard: "to the Minister for Canterbury Earthquake Recovery: Which of the assets identified by the Canterbury Earthquake Recovery Authority"

MP: "twelve hundred" Hansard: "1200" espeak: "one thousand two hundred"

## Radio The news on the RNZ website is *almost* the same as the news on the radio.

#### TV Scripts and teletext subtitles.

#### Adaptation recap

- Requires a bit of speech
- Forces NZE into a foreign phoneme set
- Is quick enough to play around with

# Creating an acoustic model

A general model requires at least 50 hours of speech

The dictionary and phoneme set can be tailored to suit

http://cmusphinx.sf.net/wiki/tutorialam

#### The instructions are long and complicated

# but that is sort of moot

## because I don't have 50 hours of speech

#### It seems the training software is:

Good at uncovering transcription errors
Less good at providing error messages

Tidying up the data is a huge job

Progress so far

I am focusing on adapting existing models

Related ephemera at http://github.com/douglasbagnall/nze-vox (Makefile, python and shell scripts, wiki pages).

My aim is automated testing of dictionaries and other parameters

# Robert Fromont is working on a model from scratch, using:

- the Canterbury Corpus
- $\cdot$  the non-free CELEX dictionary

The problems seem to be with the data.

It *should* eventually be simple to replace CELEX with a free dictionary.

Why I am interested

I work as an artist.

I am making an artwork that eavesdrops and produces video relevant to what people are talking about.

I want it to work.