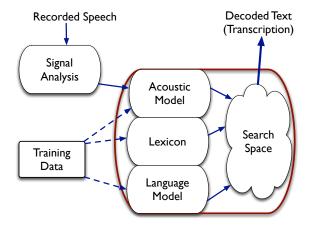
# Words: Pronunciations and Language Models

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## HMM Speech Recognition



- Words and their pronunciations provide the link between sub-word HMMs and language models
- Written by human experts
- Typically based on phones
- Constructing a dictionary involves
  - Selection of the words in the dictionary—want to ensure high coverage of words in test data
  - ② Representation of the pronunciation(s) of each word
- Explicit modelling of pronunciation variation

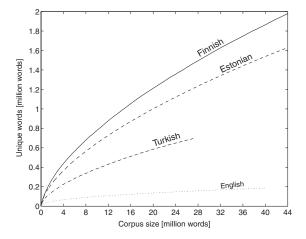
# Out-of-vocabulary (OOV) rate

- OOV rate: percent of word tokens in test data that are not contained in the ASR system dictionary
- Training vocabulary requires pronunciations for *all* words in training data (since training requires an HMM to be constructed for each training utterance)
- Select the recognition vocabulary to minimize the OOV rate (by testing on development data)
- Recognition vocabulary may be different to training vocabulary
- Empirical result: each OOV word results in 1.5–2 extra errors (>1 due to the loss of contextual information)

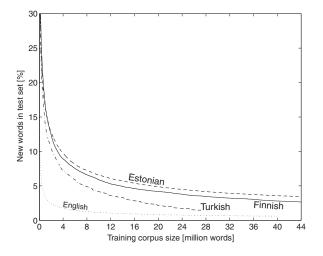
# Multilingual aspects

- Many languages are morphologically richer than English: this has a major effect of vocabulary construction and language modelling
- Compounding (eg German): decompose compund words into constituent parts, and carry out pronunciation and language modelling on the decomposed parts
- Highly inflected languages (eg Arabic, Slavic languages): specific components for modelling inflection (eg factored language models)
- Inflecting and compounding languages (eg Finnish)
- All approaches aim to reduce ASR errors by reducing the OOV rate through modelling at the morph level; also addresses data sparsity

## Vocabulary size for different languages



M. Creutz et al, "Morph-based speech recognition and modeling OOV words across languages", ACM Trans Speech and Language Processing, 5(1), art. 3. http://doi.acm.org/10.1145/1322391.1322394



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# Single and multiple pronunciations

- Words may have multiple pronunciations:
  - Accent, dialect: tomato, zebra global changes to dictionary based on consistent pronunciation variations
  - Phonological phenomena: handbag/ h ae m b ae g I can't stay / [ah k ae n s t ay]
  - Part of speech: project, excuse
- This seems to imply many pronunciations per word, including:
  - Global transform based on speaker characteristics
  - Context-dependent pronunciation models, encoding of phonological phenomena
- **BUT** state-of-the-art large vocabulary systems average about 1.1 pronunciations per word: most words have a single pronunciation

# Consistency vs Fidelity

- Empirical finding: adding pronunciation variants can result in reduced accuracy
- Adding pronunciations gives more "flexibility" to word models and increases the number of potential ambiguities—more possible state sequences to match the observed acoustics
- Speech recognition uses a consistent rather than a faithful representation of pronunciations
- A consistent representation requires only that the same word has the same phonemic representation (possibly with alternates): the training data need only be transcribed at the word level
- A faithful phonemic representation requires a detailed phonetic transcription of the training speech (much too expensive for large training data sets)

- State-of-the-art systems absorb variations in pronunciation in the acoustic models
- Context-dependent acoustic models may be though of as giving broad class representation of word context
- Cross-word context dependent models can implicitly represent cross-word phonological phenomena
- Hain (2002): a carefully constructed single pronunciation dictionary (using most common alignments) can result in a more accurate system than a multiple pronunciation dictionary

## Mathematical framework

HMM Framework for speech recognition. Let W be the universe of possible utterances, and X be the observed acoustics, then we want to find:

$$W^* = \arg \max_{W} P(W \mid X)$$
  
=  $\arg \max_{W} \frac{P(X \mid W)P(W)}{P(X)}$   
=  $\arg \max_{W} P(X \mid W)P(W)$ 

Words are composed of a sequence of HMM states Q:

$$egin{aligned} &\mathcal{W}^* = rg\max_W P(X \mid Q, W) P(Q, W) \ &\simeq rg\max_W \sum_Q P(X \mid Q) P(Q \mid W) P(W) \ &\simeq rg\max_W \max_Q P(X \mid Q) P(Q \mid W) P(W) \end{aligned}$$

#### • Acoustic model $P(X \mid Q)$

Probability of the acoustics given the phone states: context-dependent HMMs using state clustering, phonetic decision trees, etc.

# Pronunciation model P(Q | W) Probability of the phone states given the words; may be as simple a dictionary of pronunciations, or a more complex model

#### • Language model P(W)Probability of a sequence of words. Typically an *n*-gram

- Basic idea The language model is the prior probability of the word sequence *P*(*W*)
- Use a language model to disambiguate between similar acoustics when combining linguistic and acoustic evidence never mind the nudist play / never mind the new display
- Use hand constructed networks in limited domains
- Statistical language models: cover "ungrammatical" utterances, computationally efficient, trainable from huge amounts of data, can assign a probability to a sentence fragment as well as a whole sentence

## Statistical language models

- For use in speech recognition a language model must be: statistical, have wide coverage, and be compatible with left-to-right search algorithms
- Only a few grammar-based models have met this requirement (eg Chelba and Jelinek, 2000), and do not yet scale as well as simple statistical models
- Until very recently **n-grams** were the state-of-the-art language model for ASR
  - Unsophisticated, linguistically implausible
  - Short, finite context
  - Model solely at the shallow word level
  - But: wide coverage, able to deal with "ungrammatical" strings, statistical and scaleable
- Probability of a word depends only on the identity of that word and of the preceding n-1 words. These short sequences of n words are called n-grams.

# Bigram language model

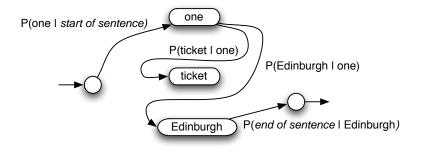
• Word sequence  $\mathbf{W} = w_1, w_2, \dots, w_M$ 

$$P(\mathbf{W}) = P(w_1)P(w_2 \mid w_1)P(w_3 \mid w_1, w_2)$$
  
...  $P(w_M \mid w_1, w_2, ..., w_{M-1})$ 

- Bigram approximation—consider only one word of context:
  P(W) ≃ P(w<sub>1</sub>)P(w<sub>2</sub> | w<sub>1</sub>)P(w<sub>3</sub> | w<sub>2</sub>)...P(w<sub>M</sub> | w<sub>M-1</sub>)
- Parameters of a bigram are the conditional probabilities  $P(w_i \mid w_j)$
- Maximum likelihood estimates by counting:

$$P(w_i|w_j) \sim rac{c(w_j,w_i)}{c(w_j)}$$

where  $c(w_j, w_i)$  is the number of observations of  $w_j$  followed by  $w_i$ , and  $c(w_j)$  is the number of observations of  $w_j$ (irrespective of what follows)



- n-grams can be represented as probabilistic finite state networks
- only some arcs (and nodes) are shown for clarity: in a full model there is an arc from every word to every word
- note the special start and end sentence probabilities

## The zero probability problem

- Maximum likelihood estimation is based on counts of words in the training data
- If a n-gram is not observed, it will have a count of 0—and the maximum likelihood probability estimate will be 0
- The zero probability problem: just because something does not occur in the training data does not mean that it will not occur
- As n grows larger, so the data grow sparser, and the more zero counts there will be
- Solution: smooth the probability estimates so that unobserved events do not have a zero probability
- Since probabilities sum to 1, this means that some probability is redistributed from observed to unobserved n-grams

- What is the probability of an unseen n-gram?
- Add-one smoothing: add one to all counts and renormalize.
  - "Discounts" non-zero counts and redistributes to zero counts
  - Since most n-grams are unseen (for large n more types than tokens!) this gives too much probability to unseen n-grams (discussed in Manning and Schütze)
- Absolute discounting: subtract a constant from the observed (non-zero count) n-grams, and redistribute this subtracted probability over the unseen n-grams (zero counts)
- Kneser-Ney smoothing: family of smoothing methods based on absolute discounting that are at the state of the art (Goodman, 2001)

# Backing off

- How is the probability distributed over unseen events?
- Basic idea: estimate the probability of an unseen n-gram using the (n-1)-gram estimate
- $\bullet~$  Use successively less context: trigram  $\rightarrow~$  bigram  $\rightarrow~$  unigram
- Back-off models redistribute the probability "freed" by discounting the n-gram counts
- For a bigram

$$egin{aligned} & P(w_i \mid w_j) = rac{c(w_j, w_i) - D}{c(w_j)} & ext{if } c(w_j, w_i) > c \ & = P(w_i) b_{w_j} & ext{otherwise} \end{aligned}$$

c is the count threshold, and D is the discount.  $b_{w_j}$  is the backoff weight required for normalization

## Interpolation

- Basic idea: Mix the probability estimates from all the estimators: estimate the trigram probability by mixing together trigram, bigram, unigram estimates
- Simple interpolation

$$\hat{P}(w_n \mid w_{n-2}, w_{n-1}) = \\\lambda_3 P(w_n \mid w_{n-2}, w_{n-1}) + \lambda_2 P(w_n \mid w_{n-1}) + \lambda_1 P(w_n)$$

With  $\sum_i \lambda_i = 1$ 

Interpolation with coefficients conditioned on the context

$$\hat{P}(w_{n} \mid w_{n-2}, w_{n-1}) = \\\lambda_{3}(w_{n-2}, w_{n-1})P(w_{n} \mid w_{n-2}, w_{n-1}) + \\\lambda_{2}(w_{n-2}, w_{n-1})P(w_{n} \mid w_{n-1}) + \lambda_{1}(w_{n-2}, w_{n-1})P(w_{n})$$

• Set  $\lambda$  values to maximise the likelihood of the interpolated language model generating a *held-out* corpus (possible to use EM to do this)

# Perplexity

- Measure the quality of a language model by how well it predicts a test set *W* (i.e. estimated probability of word sequence)
- Perplexity (*PP*(*W*)) inverse probability of the test set *W*, normalized by the number of words *N*

$$PP(W) = P(W)^{\frac{-1}{N}} = P(w_1w_2...w_N)^{\frac{-1}{N}}$$

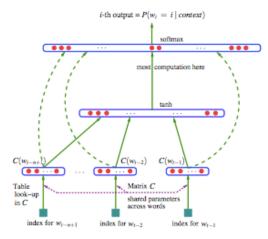
Perplexity of a bigram LM

 $PP(W) = (P(w_1)P(w_2|w_1)P(w_3|w_2)\dots P(w_N|w_{N-1}))^{\frac{-1}{N}}$ 

- Example perplexities for different n-gram LMs trained on Wall St Journal (38M words)
  - Unigram 962
  - Bigram 170
  - Trigram 109

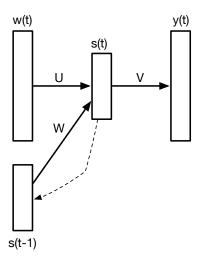
- Work in log probabilities
- The ARPA language model format is commonly used to store n-gram language models (unless they are very big)
- Many toolkits: SRILM, IRSTLM, KenLM, Cambridge-CMU toolkit, ...
- Some research issues:
  - Advanced smoothing
  - Adaptation to new domains
  - Incorporating topic information
  - Long-distance dependencies
  - Distributed representations

## Neural Probabilistic Language Model



Bengio 2003

## Recurrent Neural Network Language Model



Mikolov et al (2010,2011) - state of the art performance

## References

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- Jelinek (1991) "Up from trigrams!"
- Chelba and Jelinek (2000) example of a probabilistic grammar-based language model
- Goodman (2001) state-of-the-art smoothing for n-grams (Modified Kneser-Ney smoothing)
- Bengio (2003) Neural probabilistic language model
- Mikolov et al (2011) strategies for training large scale neural network language models (RNNs)