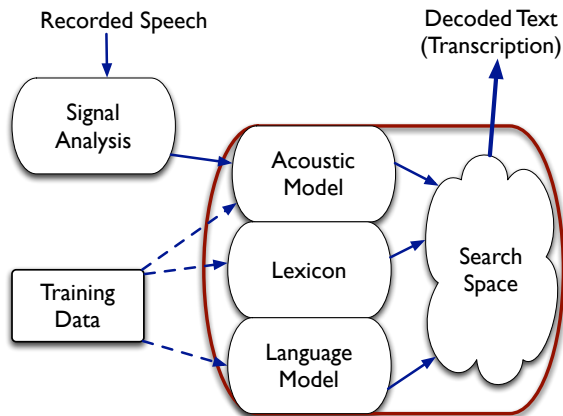


# Words: Pronunciations and Language Models

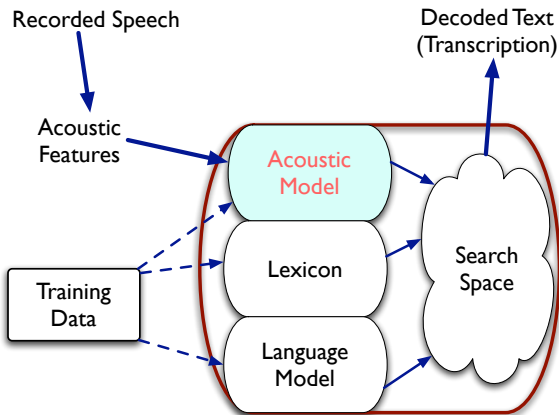
Steve Renals

Automatic Speech Recognition— ASR Lecture 8  
10 February 2014

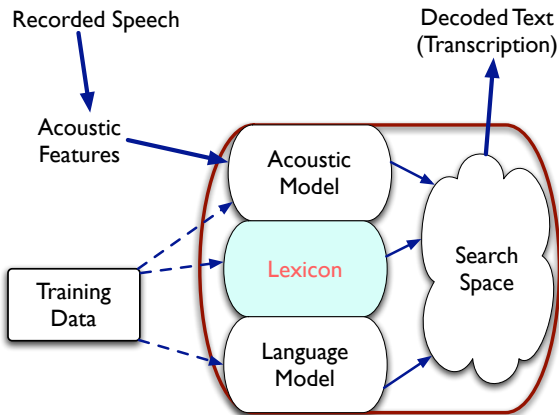
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- Explicit modelling of pronunciation variation

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- 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

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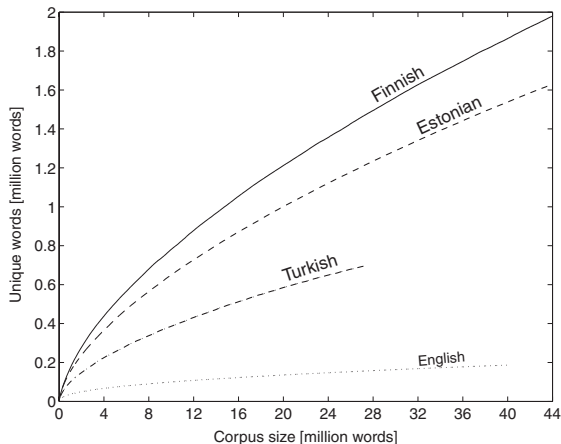
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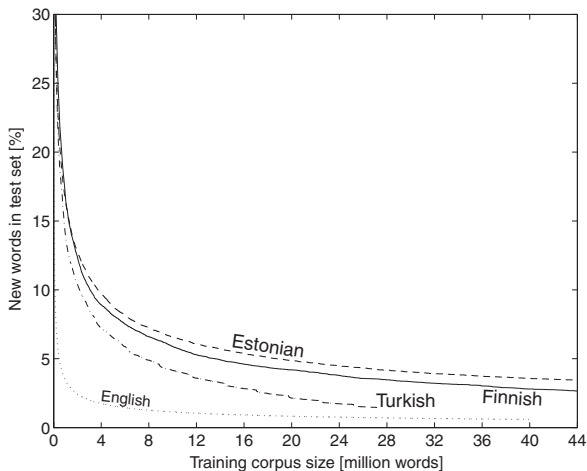
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- **Inflecting and compounding languages** (eg Finnish)
- All approaches aim to reduce ASR errors by reducing the OOV rate through modelling at the morph level; also addresses data sparsity

# 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:
  - ① Accent, dialect: *tomato*, *zebra*  
global changes to dictionary based on consistent pronunciation variations
  - ② Phonological phenomena: *handbag* / h æ m b æ g  
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- This seems to imply many pronunciations per word, including:
  - ① Global transform based on speaker characteristics
  - ② 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

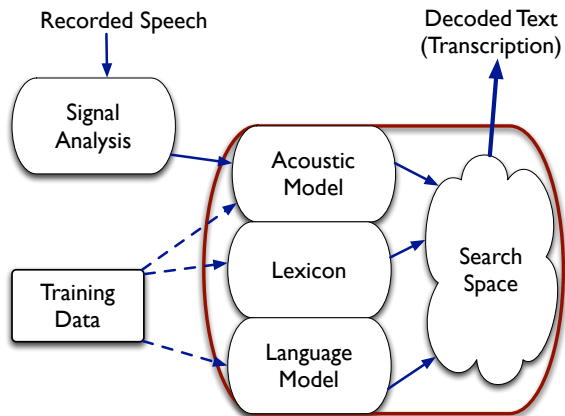
# Consistency vs Fidelity

- **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 thought 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

# HMM Speech Recognition



# 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:

$$\begin{aligned}W^* &= \arg \max_W P(W | X) \\ &= \arg \max_W \frac{P(X | W)P(W)}{P(X)} \\ &= \arg \max_W P(X | W)P(W)\end{aligned}$$

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Words are composed of a sequence of HMM states  $Q$ :

$$\begin{aligned}W^* &= \arg \max_W P(X | Q, W)P(Q, W) \\ &\simeq \arg \max_W \sum_Q P(X | Q)P(Q | W)P(W) \\ &\simeq \arg \max_W \max_Q P(X | Q)P(Q | W)P(W)\end{aligned}$$

# Three levels of model

- **Acoustic model**  $P(X | Q)$   
Probability of the acoustics given the phone states:  
context-dependent HMMs using state clustering, phonetic  
decision trees, etc.

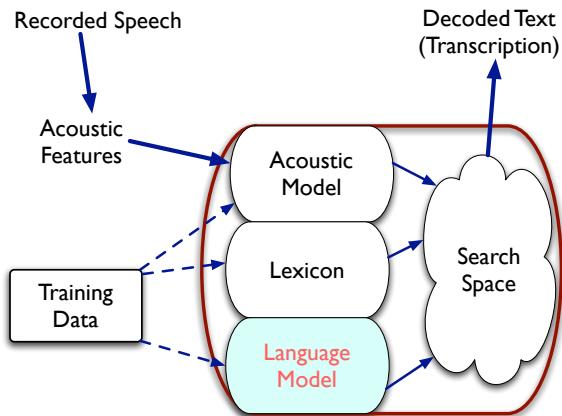
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- **Language model**  $P(W)$   
Probability of a sequence of words. Typically an  $n$ -gram

# HMM Speech Recognition





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*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

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  - 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  $\mathbf{W} = w_1, w_2, \dots, w_M$

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

- Bigram approximation—consider only one word of context:

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- Parameters of a bigram are the conditional probabilities

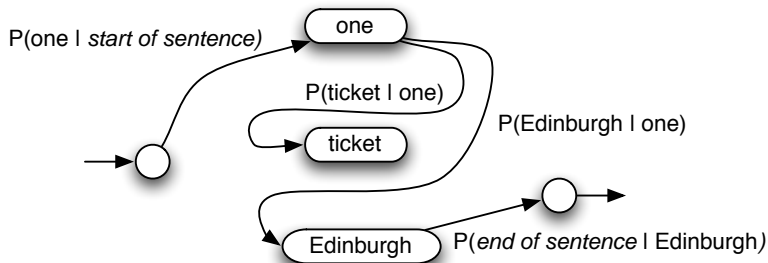
$$P(w_i | w_j)$$

- Maximum likelihood estimates by counting:

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

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)

# Bigram network



- 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



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- 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

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- 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
- Use successively less context: trigram  $\rightarrow$  bigram  $\rightarrow$  unigram
- Back-off models redistribute the probability “freed” by discounting the n-gram counts



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- For a bigram

$$P(w_i | w_j) = \frac{c(w_j, w_i) - D}{c(w_j)} \quad \text{if } c(w_j, w_i) > c$$
$$= P(w_i) b_{w_j} \quad \text{otherwise}$$

$c$  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

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

With  $\sum_i \lambda_i = 1$

- Interpolation with coefficients conditioned on the context

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

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

- 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  $N$

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

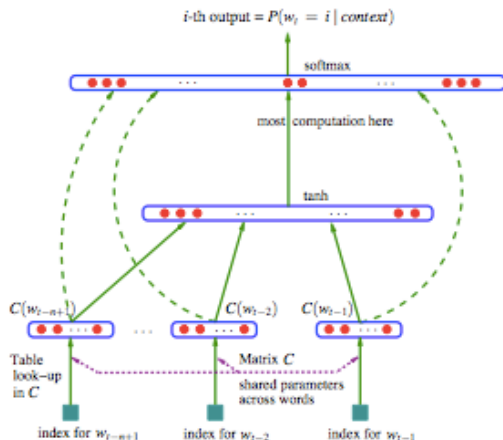
- Perplexity of a bigram LM

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

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

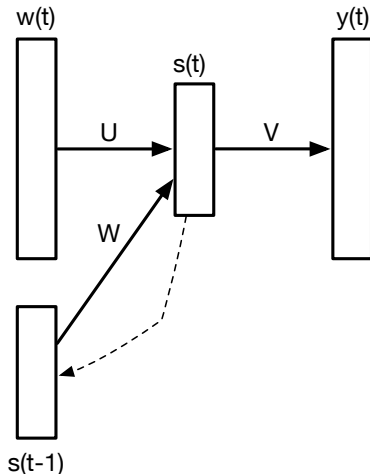
- Work in log probabilities
- The ARPA language model format is commonly used to store n-gram language models (unless they are very big)
- Many toolkits: SRILM, IRSTLM, KenLM, Cambridge-CMU toolkit, ...
- Some research issues:
  - Advanced smoothing
  - Adaptation to new domains
  - Incorporating topic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

- Jurafsky and Martin, chapter 4
- Fosler-Lussier (2003) - pronunciation modelling tutorial
- Hain (2002) - implicit pronunciation modelling by context-dependent acoustic models
- Bisani and Ney (2008) - Sequitur G2P (<http://www-i6.informatik.rwth-aachen.de/web/Software/g2p.html>)
- Gotoh and Renals (2003) - language modelling tutorial
- Good coverage of n-grams in Manning and Schütze (1999)
- 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)