EE E6820: Speech & Audio Processing & Recognition

Lecture 11: ASR: Training & Systems



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- Training HMMs
- Language modeling



Discrimination & adaptation

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HMM review

- HMM M_j is specified by:
- states q^i \odot k a t \odot
- transition probabilities a_{ij}

 $p(q_n^j | q_{n-1}^i) \equiv a_{ij}$ - emission distributions $b_i(x)$ $p(x | q^i) \equiv b_i(x)$

+ (initial state probabilities $p(q_1^i) \equiv \pi_i$)

• See e6820/papers/Rabiner89-hmm.pdf



HMM summary (1)

- HMMs are a generative model: recognition is inference of $p(M_j|X)$
- During generation, behavior of model depends only on current state *q_n*:
 - transition probabilities $p(q_{n+1} | q_n) = a_{ij}$
 - observation distributions $p(x_n | q_n) = b_i(x)$

• Given states
$$Q = \{q_1, q_2, ..., q_N\}$$

+ observations $X = X_1^N = \{x_1, x_1, ..., x_N\}$

Markov assumption makes

$$p(X, Q|M) = \prod_{n} p(x_{n}|q_{n}) p(q_{n}|q_{n-1})$$

• Given observed emissions *X*, can calculate: $p(X|M_j) = \sum_{\text{all } Q} p(X|Q, M) p(Q|M)$

HMM summary (2)

• Calculate p(X|M) via forward recursion:

$$p(X_1^n, q_n^j) = \alpha_n(j) = \left[\sum_{i=1}^S \alpha_{n-1}(i)a_{ij}\right] \cdot b_j(x_n)$$

• Viterbi (best path) approximation

$$\alpha_n^*(j) = \left[\max_i \left\{\alpha_{n-1}^*(i)a_{ij}\right\}\right] \cdot b_j(x_n)$$

- then backtrace...

$$Q^* = \underset{Q}{\operatorname{argmax}} p(X, Q|M)$$

• Pictorially:



Outline



Hidden Markov Model review



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Training HMMs

- Viterbi training
- EM for HMM parameters
- Forward-backward (Baum-Welch)
- Language modeling
 - **Discrimination & adaptation**





Training HMMs

- Probabilistic foundation allows us to train HMMs to 'fit' training data
 - i.e. estimate a_{ij} , $b_i(x)$ given data
 - better than DTW...
- Algorithms to improve $p(M \mid X)$ are key to success of HMMs
 - maximum-likelihood of models...
- State alignments *Q* of training examples are generally unknown
 - else estimating parameters would be easy
- → Viterbi training
 - choose 'best' labels (heuristic)

\rightarrow EM training

- 'fuzzy labels' (guaranteed local convergence)



Overall training procedure



Viterbi training

- "Fit models to data"
 - = Viterbi best-path alignment



• "Re-estimate model parameters":

pdf e.g. 1D Gauss:
$$\mu_i = \frac{\sum_{n \in q^i} x_n}{\#(q_n^i)}$$

 $\#(q_n^i) \rightarrow q^j)$

count transitions:
$$a_{ij} = \frac{\#(q_{n-1} \rightarrow q_n)}{\#(q_n^i)}$$

- And repeat...
- But: converges only if good initialization



EM for HMMs

- Expectation-Maximization (EM): optimizes models with unknown parameters
 - finds locally-optimal parameters Θ to maximize data likelihood $p(x_{train}|\Theta)$
 - makes sense for decision rules like $p(x|M_j) \cdot p(M_j)$
- Principle:

Adjust Θ to maximize expected log likelihood of known *x* & unknown *u*:

$$E[\log p(x, u|\Theta)] = \sum_{u} p(u|x, \Theta_{old}) \log[p(x|u, \Theta)p(u|\Theta)]$$

- for GMMs, unknowns = mix assignments *k*
- for HMMs, unknowns = hidden state q_n (take Θ to include M_j)
- Interpretation: "fuzzy" values for unknowns



What EM does

 Maximize data likelihood by repeatedly estimating unknowns and re-maximizing expected log likelihood:





EM for HMMs (2)

• Expected log likelihood for HMM:

$$\begin{split} &\sum_{\text{all } Q_k} p(Q_k | X, \Theta_{\text{old}}) \log[p(X | Q_k, \Theta) p(Q_k | \Theta)] \\ &= \sum_{\text{all } Q_k} p(Q_k | X, \Theta_{\text{old}}) \log\left[\prod_n p(x_n | q_n) \cdot p(q_n | q_{n-1})\right] \\ &= \sum_{n=1}^N \sum_{i=1}^S p(q_n^i | X, \Theta_{\text{old}}) \log p(x_n | q_n^i, \Theta) \\ &+ \sum_{i=1}^S p(q_1^i | X, \Theta_{\text{old}}) \log p(q_1^i | \Theta) \\ &+ \sum_{n=2}^N \sum_{i=1}^S \sum_{j=1}^S p(q_{n-1}^i, q_n^j | X, \Theta_{\text{old}}) \log p(q_n^j | q_{n-1}^i, \Theta) \end{split}$$

- closed-form maximization by differentiation etc.





EM update equations

• For acoustic model (e.g. 1-D Gauss):

$$\mu_{i} = \frac{\sum_{n} p(q_{n}^{i} | X, \Theta_{\text{old}}) \cdot x_{n}}{\sum_{n} p(q_{n}^{i} | X, \Theta_{\text{old}})}$$

• For transition probabilities:

$$p(q_n^j | q_{n-1}^i) = a_{ij}^{\text{new}} = \frac{\sum_n p(q_{n-1}^i, q_n^j | X, \Theta_{\text{old}})}{\sum_n p(q_{n-1}^i | X, \Theta_{\text{old}})}$$

- Fuzzy versions of Viterbi training
 - reduce to Viterbi if p(q|X) = 1/0
- Require 'state occupancy probabilities',

$$p(q_n^i | X_1^N, \Theta_{\text{old}})$$



The forward-backward algorithm

- We need $p(q_n^i | X_1^N)$ for EM updates (Θ implied)
- Forward algorithm gives $\alpha_n(i) = p(q_n^i, X_1^n)^{\bullet}$

- excludes influence of remaining data X_{n+1}^N

Hence, define
$$\beta_n(i) = p(X_{n+1}^N | q_n^i, X_1^n)$$

so that $\alpha_n(i) \cdot \beta_n(i) = p(q_n^i, X_1^n)$
then $p(q_n^i | X_1^N) = \frac{\alpha_n(i) \cdot \beta_n(i)}{\sum_j \alpha_n(j) \cdot \beta_n(j)}$

• **Recursive** definition for β :

$$\beta_n(i) = \sum_j \beta_{n+1}(j) a_{ij} b_j(x_{n+1})$$

- recurses backwards from final state *N*



Estimating a_{ij} from α & β

• From EM equations:

$$p(q_n^j | q_{n-1}^i) = a_{ij}^{\text{new}} = \frac{\sum_n p(q_{n-1}^i, q_n^j | X, \Theta_{\text{old}})}{\sum_n p(q_{n-1}^i | X, \Theta_{\text{old}})}$$

- prob. of transition normalized by prob. in first

• Obtain from
$$p(q_{n-1}^{i}, q_{n}^{j}, X|\Theta_{old})$$

$$= p(X_{n+1}^{N} | q_{n}^{j}) p(x_{n} | q_{n}^{j}) p(q_{n}^{j} | q_{n-1}^{i}) p(q_{n-1}^{i}, X_{1}^{n-1})$$

$$= \beta_{n}(j) \cdot b_{j}(x_{n}) \cdot a_{ij} \cdot \alpha_{n-1}(i)$$

$$\stackrel{b_{j}(x_{n})}{\longrightarrow} \stackrel{b_{j}(x_{n})}{\longrightarrow} \stackrel{\phi}{\longrightarrow} \stackrel{\phi$$

GMM-HMMs in practice

- GMMs as acoustic models: train by including mixture indices as unknowns
 - just more complicated equations...

$$\mu_{ik} = \frac{\sum_{n} p(m_k | q^i, x_n, \Theta_{\text{old}}) p(q_n^i | X, \Theta_{\text{old}}) \cdot x_n}{\sum_{n} p(m_k | q^i, x_n, \Theta_{\text{old}}) p(q_n^i | X, \Theta_{\text{old}})}$$

• Practical GMMs:

- 9 to 39 feature dimensions
- 2 to 64 Gaussians per mixture depending on number of training examples

• Lots of data \rightarrow can model more classes

- e.g context-independent (CI): $q^i = ae aa ax$...

→context-dependent (CD): $q^i = b$ -ae-b b-ae-k ...



HMM training in practice

- EM only finds local optimum
 → critically dependent on initialization
 - approximate parameters / rough alignment



• Applicable for more than just words...



Training summary

Training data + basic model topologies
 → derive fully-trained models



- alignment all handled implicitly
- What do the states end up meaning?
 - not necessarily what you intended;
 whatever locally maximizes data likelihood
- What if the models or transcriptions are bad?
 - slow convergence, poor discrimination in models
- Other kinds of data, transcriptions
 - less constrained initial models...



Outline

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 - Training HMMs
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Language modeling

- Pronunciation models
- Grammars
- Decoding
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Discrimination & adaptation





Language models

• Recall, MAP recognition criterion:

$$M^{*} = \underset{M_{j}}{\operatorname{argmax}} p(M_{j}|X, \Theta)$$

=
$$\underset{M_{j}}{\operatorname{argmax}} p(X|M_{j}, \Theta_{A}) p(M_{j}|\Theta_{L})$$

- So far, looked at $p(X|M_j, \Theta_A)$
- What about $p(M_j | \Theta_L)$?
 - M_i is a particular word sequence
 - Θ_L are parameters related to the language
- Two components:
 - link state sequences to words $p(Q|w_i)$
 - priors on word sequences $p(w_i | M_j)$



HMM Hierarchy

• HMMs support composition

- can handle time dilation, pronunciation, grammar all within the same framework





Pronunciation models

- Define states within each word $p(Q|w_i)$
- Can have unique states for each word ('whole-word' modeling), or ...
- Sharing (tying) subword units between words to reflect underlying phonology
 - more training examples for each unit
 - generalizes to unseen words
 - (or can do it automatically...)

• Start e.g. from pronouncing dictionary:

ZERO(0.5)	z iy r ow
ZERO(0.5)	z ih r ow
ONE(1.0)	w ah n
TWO(1.0)	tcl t uw

• • •

Learning pronunciations

- 'Phone recognizer' transcribes training data as phones
 - align to 'canonical' pronunciations



- infer modification rules
- predict other pronunciation variants
- e.g. 'd deletion':

 $d \rightarrow \emptyset / 1 [stop] p = 0.9$

• Generate pronunciation variants; use forced alignment to find weights



Grammar

- Account for different likelihoods of different words and word sequences $p(w_i|M_j)$
- 'True' probabilities are very complex for LVCSR
 - need parses, but speech often agrammatic
- → Use n-grams:

$$p(w_n | w_1^L) = p(w_n | w_{n-K}, ..., w_{n-1})$$

- e.g. n-gram models of Shakespeare:
- n=1 To him swallowed confess hear both. Which. Of save on ...
- n=2 What means, sir. I confess she? then all sorts, he is trim, ...
- n=3 Sweet prince, Falstaff shall die. Harry of Monmouth's grave...
- n=4 King Henry. What! I will go seek the traitor Gloucester. ...

• Big win in recognizer WER

- raw recognition results often highly ambiguous
- grammar guides to 'reasonable' solutions



Smoothing LVCSR grammars

- n-grams (n=3 or 4) are estimated from large text corpora
 - 100M+ words
 - but: not like spoken language
- 100,000 word vocabulary $\rightarrow 10^{15}$ trigrams!
 - never see enough examples
 - unobserved trigrams should NOT have Pr=0!
- Backoff to bigrams, unigrams
 - $p(w_n)$ as an approx to $p(w_n | w_{n-1})$ etc.
 - interpolate 1-gram, 2-gram, 3-gram with learned weights?
- Lots of ideas e.g. category grammars
 - e.g. $p(PLACE | "went", "to") \cdot p(w_n | PLACE)$
 - how to define categories?
 - how to tag words in training corpus?



Decoding

- How to find the MAP word sequence?
- States, prons, words define one big HMM
 - with 100,000+ individual states for LVCSR!
- → Exploit hierarchic structure
 - phone states independent of word
 - next word (semi) independent of word history





Decoder pruning

- Searching 'all possible word sequences'?
 - need to restrict search to most promising ones:
 beam search
 - sort by estimates of total probability
 - = Pr(so far) + lower bound estimate of remains
 - trade search errors for speed

• Start-synchronous algorithm:

- extract top hypothesis from queue:

 $[P_n, \{w_1, ..., w_k\}, n]$ pr. so far words next time frame

find plausible words {wⁱ} starting at time n
 → new hypotheses:

$$[P_n \cdot p(X_n^{n+N-1} | w^i) \cdot p(w^i | w_k...), \{w_1, ..., w_k, w^i\}, n+N]$$

- discard if too unlikely, or queue is too long
- else re-insert into queue and repeat

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Discrimination & adaptation

- Discriminant models
- Neural net acoustic models
- Model adaptation



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Discriminant models

- EM training of HMMs is maximum likelihood
 - i.e. choose single Θ to max $p(X_{trn} | \Theta)$
 - Bayesian approach: actually $p(\Theta | X_{trn})$
- **Decision rule is** max $p(X | M) \cdot p(M)$
 - training will increase $p(X | M_{correct})$
 - may also increase $p(X | M_{wrong})$...as much?
- **Discriminant** training tries directly to increase discrimination between right & wrong models

- e.g. Maximum Mutual Information (MMI)

$$I(M_{j}, X | \Theta) = \log \frac{p(M_{j}, X | \Theta)}{p(M_{j} | \Theta) p(X | \Theta)}$$

$$= \log \frac{p(X | M_{j}, \Theta)}{\sum p(X | M_{k}, \Theta) p(M_{k} | \Theta)}$$
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Neural Network Acoustic Models

- Single model generates posteriors directly for all classes at once = frame-discriminant
- Use regular HMM decoder for recognition

- set
$$b_i(x_n) = p(x_n | q^i) \propto p(q^i | x_n) / p(q^i)$$

- Nets are less sensitive to input representation
 - skewed feature distributions
 - correlated features
- Can use temporal context window to let net 'see' feature dynamics:



Neural nets: Practicalities

- Typical net sizes:
 - input layer: 9 frames x 9-40 features ~ 300 units
 - hidden layer: 100-8000 units, dep. train set size
 - output layer: 30-60 context-independent phones
- Hard to make context dependent
 - problems training many classes that are similar?





Model adaptation

- Practical systems often suffer from mismatch
 - test conditions are not like training data: accent, microphone, background noise ...
- Desirable to continue tuning during recognition
 = adaptation
 - but: no 'ground truth' labels or transcription
- Assume that recognizer output is correct; Estimate a few parameters from those labels
 - e.g. Maximum Likelihood Linear Regression (MLLR)



Recap: Recognizer Structure



• Now we have it all!



Summary

• Hidden Markov Models

- state transitions and emission likelihoods in one
- best path (Viterbi) performs recognition

• HMMs can be trained

- Viterbi training makes intuitive sense
- EM training is guaranteed to converge
- acoustic models (e.g. GMMs) train at same time
- Language modeling captures higher structure
 - pronunciation, word sequences
 - fits directly into HMM state structure
 - need to 'prune' search space in decoding
- Further improvements...
 - discriminant training moves models 'apart'
 - adaptation adjusts models in new situations

