EE E6820: Speech & Audio Processing & Recognition

## Lecture 10: ASR: Sequence Recognition



- 2 Statistical sequence recognition
  - Acoustic modeling
  - The Hidden Markov Model (HMM)

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# Signal template matching

Framewise comparison

of unknown word and stored templates:



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# **Dynamic Time Warp (DTW)**

- Find lowest-cost constrained path:
  - matrix d(i,j) of distances between input frame  $f_i$  and reference frame  $r_j$
  - allowable predecessors & transition costs  $T_{\chi y}$



Input frames f<sub>i</sub>

- Best path via traceback from final state
  - store predecessors for each (*i*,*j*)



### **DTW-based recognition**

- Reference templates for each possible word
- For isolated words:
  - mark endpoints of input word
  - calculate scores through each template (+prune)
  - choose best

#### • For continuous speech

one matrix of template slices;
 special-case constraints at word ends



## **DTW-based recognition (2)**

- + Successfully handles timing variation
- + Able to recognize speech at reasonable cost
- Distance metric?
  - pseudo-Euclidean space?
- Warp penalties?
- How to choose templates?
  - several templates per word?
  - choose 'most representative'?
  - align and average?
- → need a rigorous foundation...



# Outline



- Statistical sequence recognition
  - state-based modeling
- Acoustic modeling
- The Hidden Markov Model (HMM)



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# 2 Statistical sequence recognition

- DTW limited because it's hard to optimize
  - interpretation of distance, transition costs?
- Need a theoretical foundation: Probability
- Formulate recognition as MAP choice among models:

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

- *X* = observed features
- $M_j$  = word-sequence models
- $\Theta$  = all current parameters



# **Statistical formulation (2)**

• Can rearrange via Bayes' rule (& drop p(X)):

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

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

likelihood of observations under model

- $p(M_j)$  = prior probability of model
- $\Theta_A$  = acoustics-related model parameters
- $\Theta_L$  = language-related model parameters

### • Questions:

- what form of model to use for  $p(X|M_{j}, \Theta_{A})$ ?
- how to find  $\Theta_A$  (training)?
- how to solve for  $M_j$  (decoding)?



### **State-based modeling**

- Assume discrete-state model for the speech:
  - observations are divided up into time frames
  - model  $\rightarrow$  states  $\rightarrow$  observations:



• Probability of observations given model is:

$$p(X|M_j) = \sum_{\text{all } Q_k} p(X_1^N | Q_k, M_j) \cdot p(Q_k | M_j)$$

- sum over all possible state sequences  $Q_k$
- How do observations depend on states? How do state sequences depend on model?



### The speech recognition chain

• After classification, still have problem of classifying the sequences of frames:



#### • Questions

- what to use for the acoustic classifier?
- how to represent 'model' sequences?
- how to score matches?



# Outline

- **1** Signal template matching
  - Statistical sequence recognition
  - Acoustic modeling
    - defining targets
    - neural networks & Gaussian models
  - The Hidden Markov Model (HMM)



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### **Acoustic Modeling**

• Goal: Convert features into probabilities of particular labels:

i.e find  $p(q_n^i | X_n)$  over some state set  $\{q^i\}$ 

- conventional statistical classification problem
- Classifier construction is data-driven
  - assume we can get examples of known good Xs for each of the  $q^i$ s
  - calculate model parameters by standard training scheme
- Various classifiers can be used
  - GMMs model distribution under each state
  - Neural Nets directly estimate posteriors
- Different classifiers have different properties
  - features, labels limit ultimate performance



# **Defining classifier targets**

- Choice of  $\{q^i\}$  can make a big difference
  - must support recognition task
  - must be a practical classification task
- Hand-labeling is one source...
  - 'experts' mark spectrogram boundaries
- ...Forced alignment is another
  - 'best guess' with existing classifiers, given words
- Result is targets for each training frame:





### **Forced alignment**

• Best labeling given existing classifier constrained by known word sequence



### **Gaussian Mixture Models vs. Neural Nets**

- GMMs fit distribution of features under states:
  - separate 'likelihood' model for each state  $q^i$

$$p(\mathbf{x}|q^{k}) = \frac{1}{(\sqrt{2\pi})^{d} |\Sigma_{k}|^{1/2}} \cdot \exp\left[-\frac{1}{2}(\mathbf{x} - \boldsymbol{\mu}_{k})^{T} \Sigma_{k}^{-1}(\mathbf{x} - \boldsymbol{\mu}_{k})\right]$$

- match any distribution given enough data
- Neural nets estimate posteriors directly  $p(q^k | \mathbf{x}) = F[\sum_j w_{jk} \cdot F[\sum_j w_{ij} x_i]]$ 
  - parameters set to discriminate classes
- Posteriors & likelihoods related by Bayes' rule:

$$p(q^{k}|\mathbf{x}) = \frac{p(\mathbf{x}|q^{k}) \cdot Pr(q^{k})}{\sum_{j} p(\mathbf{x}|q^{j}) \cdot Pr(q^{j})}$$



# Outline

- **1** Signal template matching
  - Statistical sequence recognition
  - Acoustic classification

#### The Hidden Markov Model (HMM)

- generative Markov models
- hidden Markov models
- model fit likelihood
- HMM examples



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### **Markov models**

- A (first order) Markov model is a finite-state system whose behavior depends only on the current state
- E.g. generative Markov model:



#### SAAAAAABBBBBBBBCCCCBBBBBBCE



### **Hidden Markov models**

- = Markov model where state sequence  $Q = \{q_n\}$  is not directly observable (= 'hidden')
- But, observations *X* do depend on *Q*:
  - $x_n$  is rv that depends on current state: p(x|q)



### (Generative) Markov models (2)

- HMM is specified by:
- states  $q^i$

- (k) (a) (t) (•)
- transition probabilities  $a_{ij}$  $p(q_n^j | q_{n-1}^i) \equiv a_{ij}$

 $\odot$ 

- emission distributions  $b_i(x)$  $p(x|q^i) \equiv b_i(x)^{p(x|q)}$
- + (initial state probabilities  $p(q_1^i) \equiv \pi_i$ )



### Markov models for speech

- Speech models  $M_j$ 
  - usually left-to-right HMMs (sequence constraint)
  - observation & evolution are conditionally independent of rest given (hidden) state  $q_n$





- self-loops for time dilation



### Markov models for sequence recognition

- **Independence** of observations: ullet
  - observation  $x_n$  depends only current state  $q_n$

$$p(X|Q) = p(x_1, x_2, \dots x_N | q_1, q_2, \dots q_N)$$
  
=  $p(x_1|q_1) \cdot p(x_2|q_2) \cdot \dots p(x_N|q_N)$   
=  $\prod_{n=1}^{N} p(x_n|q_n) = \prod_{n=1}^{N} b_{q_n}(x_n)$ 

**Markov transitions:** •

- transition to next state 
$$q_{i+1}$$
 depends only on  $q_i$   
 $p(Q|M) = p(q_1, q_2, ..., q_N|M)$   
 $= p(q_N|q_1...q_{N-1})p(q_{N-1}|q_1...q_{N-2})...p(q_2|q_1)p(q_1)$   
 $= p(q_N|q_{N-1})p(q_{N-1}|q_{N-2})...p(q_2|q_1)p(q_1)$   
 $= p(q_1)\prod_{n=2}^{N} p(q_n|q_{n-1}) = \pi_{q_1}\prod_{n=2}^{N} a_{q_{n-1}q_n}$ 

### **Model-fit calculation**

• From 'state-based modeling':

$$p(X|M_j) = \sum_{\text{all } Q_k} p(X_1^N | Q_k, M_j) \cdot p(Q_k | M_j)$$

• For HMMs:

$$p(X|Q) = \prod_{n=1}^{N} b_{q_n}(x_n)$$
$$p(Q|M) = \pi_{q_1} \cdot \prod_{n=2}^{N} a_{q_{n-1}q_n}$$

- Hence, solve for  $M^*$ :
  - calculate  $p(X|M_j)$  for each available model, scale by prior  $p(M_j) \rightarrow p(M_j|X)$
- Sum over all  $Q_k$ ???





 $\Sigma = p(X | M) = 0.4020$ (length 3 paths only)

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 $\Sigma = 0.1109$ 



### The 'forward recursion'

- Dynamic-programming-like technique to calculate sum over all  $Q_k$
- Define  $\alpha_n(i)$  as the probability of getting to state  $q^i$  at time step *n* (by any path):  $\alpha_n(i) = p(x_1, x_2, \dots x_n, q_n = q^i) \equiv p(X_1^n, q_n^i)$
- Then  $\alpha_{n+1}(j)$  can be calculated recursively:





### **Forward recursion (2)**

• Initialize 
$$\alpha_1(i) = \pi_i \cdot b_i(x_1)$$

• Then total probability 
$$p(X_1^N | M) = \sum_{i=1}^{S} \alpha_N(i)$$

→ Practical way to solve for  $p(X | M_j)$ and hence perform recognition





# **Optimal path**

- May be interested in actual  $q_n$  assignments
  - which state was 'active' at each time frame
  - e.g. phone labelling (for training?)
- Total probability is over all paths...
- ... but can also solve for single best path
   "Viterbi" state sequence
- Probability along best path to state  $q_{n+1}^{j}$ :  $\alpha_{n+1}^{*}(j) = \left[\max_{i} \left\{\alpha_{n}^{*}(i)a_{ij}\right\}\right] \cdot b_{j}(x_{n+1})$ 
  - backtrack from final state to get best path
  - final probability is product only (no sum)
     → log-domain calculation is just summation
- Total probability often dominated by best path:

$$p(X, Q^* \middle| M) \approx p(X | M)$$



# Interpreting the Viterbi path

- Viterbi path assigns each  $x_n$  to a state  $q^i$ 
  - performing classification based on  $b_i(x)$
  - ... at the same time as applying transition constraints  $a_{ii}$



#### • Can be used for segmentation

- train an HMM with 'garbage' and 'target' states
- decode on new data to find 'targets', boundaries

#### • Can use for (heuristic) training

- e.g. train classifiers based on labels...



### **Recognition with HMMs**

- Isolated word
  - choose best  $p(M|X) \propto p(X|M)p(M)$



- Continuous speech
  - Viterbi decoding of one large HMM gives words



### HMM examples: Different state sequences



#### **Model matching: Emission probabilities Observation** А sequence Κ<sup>2</sup> $x_n$ 0 0 14 16 í٨ 2 6 8 10 12 18 Δ time n / steps Model $M_1$ $\log p(X \mid M) = -32.1$ state alignment $\log p(\bar{X}, Q^* | M) = -33.5$ log trans.prob $\log p(Q^* | M) = -7.5$ log obs.l'hood $\log p(X \mid Q^*, M) = -26.0_{-10}^{-5}$ Model $M_2$ $\log p(X | M) = -47.0$ state alignment $\log p(\bar{X}, Q^* | M) = -47.5$ log trans.prob $\log p(Q^* | M) = -8.3$ log obs.l'hood $\log p(X \mid Q^*, M) = -39.2$ 24

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## Validity of HMM assumptions

- Key assumption is conditional independence: Given  $q^i$ , future evolution & obs. distribution are independent of previous events
  - duration behavior: self-loops imply exponential distribution



- independence of successive  $x_n$ s





### **Recap: Recognizer Structure**



- We now know how to execute each stage
- .. but how to train HMMs?
- .. where to get word/language models?



## Summary

- Speech is modeled as a sequence of features
  - need temporal aspect to recognition
  - best time-alignment of templates = DTW

#### • Hidden Markov models are rigorous solution

- self-loops allow temporal dilation
- exact, efficient likelihood calculations

Parting thought: How to set the HMM parameters? (training)

