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# **Information Theoretic Approach to ASR**



Recognition is achieved by maximizing the probability of the linguistic string, W, given the acoustic evidence, A, i.e., choose the linguistic sequence  $\hat{W}$  such that

 $P(\hat{\boldsymbol{W}}|\boldsymbol{A}) = \max_{\boldsymbol{W}} P(\boldsymbol{W}|\boldsymbol{A})$ 

# **Information Theoretic Approach to ASR**

• From Bayes rule:

$$P(\boldsymbol{W}|\boldsymbol{A}) = \frac{P(\boldsymbol{A}|\boldsymbol{W})P(\boldsymbol{W})}{P(\boldsymbol{A})}$$

- Hidden Markov modelling (HMM) deals with the quantity  $P(\mathbf{A}|\mathbf{W})$
- Change in notation:

 $\begin{array}{ccc} \boldsymbol{A} & \to & \boldsymbol{O} \\ \boldsymbol{W} & \to & \lambda \\ P(\boldsymbol{A} | \boldsymbol{W}) & \to & P(\boldsymbol{O} | \lambda) \end{array}$ 





- Consider 3 mugs, each with mixtures of *state* stones, 1 and 2
- The fractions for the  $i^{th}$  mug are  $a_{i1}$  and  $a_{i2}$ , and  $a_{i1} + a_{i2} = 1$
- Consider 2 urns, each with mixtures of black and white balls
- The fractions for the  $i^{th}$  urn are  $b_i(B)$  and  $b_i(W)$ ;  $b_i(B) + b_i(W) = 1$
- The parameter vector for this model is:

 $\lambda = \{a_{01}, a_{02}, a_{11}, a_{12}, a_{21}, a_{22}, b_1(B), b_1(W), b_2(B), b_2(W)\}$ 

### HMM: An Example (cont'd)



 Observation Sequence:
  $O = \{B, W, B, W, W, B\}$  

 State Sequence:
  $Q = \{1, 1, 2, 1, 2, 1\}$ 

**Goal:** Given the model  $\lambda$  and the observation sequence O, how can the underlying state sequence Q be determined?

# **Elements of a Discrete Hidden Markov Model**

- *N*: number of states in the model
  - states,  $s = \{s_1, s_2, \dots, s_N\}$
  - state at time  $t, q_t \in \mathbf{s}$
- *M*: number of observation symbols (i.e., discrete observations)
  - observation symbols,  $\mathbf{v} = \{v_1, v_2, \dots, v_M\}$
  - observation at time  $t, o_t \in \mathbf{v}$
- $A = \{a_{ij}\}$ : state transition probability distribution -  $a_{ij} = P(q_{t+1} = s_j | q_t = s_i), 1 \le i, j \le N$
- $B = \{b_j(k)\}$ : observation symbol probability distribution in state  $j b_j(k) = P(v_k \text{ at } t | q_t = s_j), 1 \le j \le N, 1 \le k \le M$
- $\pi = \{\pi_i\}$ : initial state distribution

 $-\pi_i = P(q_1 = s_i), \ 1 \le i \le N$ 

Notationally, an HMM is typically written as:  $\lambda = \{A, B, \pi\}$ 

# HMM: An Example (cont'd)

For our simple example:

$$\boldsymbol{\pi} = \{a_{01}, a_{02}\}, \quad \boldsymbol{A} = \begin{bmatrix} a_{11} & a_{12} \\ a_{21} & a_{22} \end{bmatrix}, \text{ and } \quad \boldsymbol{B} = \begin{bmatrix} b_1(B) & b_1(W) \\ b_2(B) & b_2(W) \end{bmatrix}$$

#### **State Diagram**

2-state

3-state





# **Generation of HMM Observations**

- 1. Choose an initial state,  $q_1 = s_i$ , based on the initial state distribution,  $\pi$
- 2. For t = 1 to *T*:
  - Choose o<sub>t</sub> = v<sub>k</sub> according to the symbol probability distribution in state s<sub>i</sub>, b<sub>i</sub>(k)
  - Transition to a new state  $q_{t+1} = s_j$  according to the state transition probability distribution for state  $s_i$ ,  $a_{ij}$
- 3. Increment *t* by 1, return to step 2 if  $t \le T$ ; else, terminate



# **Representing State Diagram by Trellis**





The dashed line represents a *null* transition, where no observation symbol is generated

# Three Basic HMM Problems

1. **Scoring:** Given an observation sequence  $O = \{o_1, o_2, ..., o_T\}$  and a model  $\lambda = \{A, B, \pi\}$ , how do we compute  $P(O \mid \lambda)$ , the probability of the observation sequence?

==> The Forward-Backward Algorithm

- 2. Matching: Given an observation sequence O = {o<sub>1</sub>, o<sub>2</sub>, ..., o<sub>T</sub>}, how do we choose a state sequence Q = {q<sub>1</sub>, q<sub>2</sub>, ..., q<sub>T</sub>} which is optimum in some sense?
   => The Viterbi Algorithm
- 3. **Training:** How do we adjust the model parameters  $\lambda = \{A, B, \pi\}$  to maximize  $P(\mathbf{O} \mid \lambda)$ ?

==> The Baum-Welch Re-estimation Procedures



 $P(\boldsymbol{O}|\lambda) = \sum_{all \boldsymbol{Q}} P(\boldsymbol{O}, \boldsymbol{Q}|\lambda)$ 

 $P(\boldsymbol{O}, \boldsymbol{Q} | \lambda) = P(\boldsymbol{O} | \boldsymbol{Q}, \lambda) P(\boldsymbol{Q} | \lambda)$ 

• Consider the *fixed* state sequence:  $Q = q_1 q_2 \dots q_T$ 

 $P(O|Q,\lambda) = b_{q_1}(o_1)b_{q_2}(o_2)\dots b_{q_T}(o_T)$ 

$$P(\boldsymbol{Q}|\lambda) = \pi_{q_1}a_{q_1q_2}a_{q_2q_3}\dots a_{q_{T-1}q_T}$$

Therefore:

$$P(\mathbf{O}|\lambda) = \sum_{q_1, q_2, \dots, q_T} \pi_{q_1} b_{q_1}(o_1) a_{q_1 q_2} b_{q_2}(o_2) \dots a_{q_{T-1} q_T} b_{q_T}(o_T)$$

• Calculation required  $\approx 2T \cdot N^T$  (there are  $N^T$  such sequences) For  $N = 5, T = 100 \Rightarrow 2 \cdot 100 \cdot 5^{100} \approx 10^{72}$  computations!

# The Forward Algorithm

• Let us define the forward variable,  $\alpha_t(i)$ , as the probability of the partial observation sequence up to time *t* and state  $s_i$  at time *t*, given the model, i.e.

$$\alpha_t(i) = P(o_1 o_2 \dots o_t, q_t = s_i | \lambda)$$

• It can easily be shown that:

$$\alpha_1(i) = \pi_i b_i(o_1), \qquad 1 \le i \le N$$

$$P(\boldsymbol{O}|\lambda) = \sum_{i=1}^{N} \alpha_{T}(i)$$

• By induction:

$$\alpha_{t+1}(j) = \left[\sum_{i=1}^{N} \alpha_t(i) a_{ij}\right] b_j(o_{t+1}), \qquad \begin{array}{l} 1 \le t \le T-1 \\ 1 \le j \le N \end{array}$$

• Calculation is on the order of  $N^2T$ . For  $N = 5, T = 100 \Rightarrow 100 \cdot 5^2$  computations, instead of  $10^{72}$ 

### **Forward Algorithm Illustration**



# The Backward Algorithm

• Similarly, let us define the backward variable,  $\beta_t(i)$ , as the probability of the partial observation sequence from time t + 1 to the end, given state  $s_i$  at time t and the model, i.e.

$$\beta_t(i) = P(o_{t+1}o_{t+2}\dots o_T | q_t = s_i, \lambda)$$

• It can easily be shown that:

$$\beta_T(i) = 1, \qquad 1 \le i \le N$$

and:

$$P(\boldsymbol{O}|\lambda) = \sum_{i=1}^{N} \pi_i b_i(o_1) \beta_1(i)$$

• By induction:

$$\beta_t(i) = \sum_{j=1}^N a_{ij} b_j(o_{t+1}) \beta_{t+1}(j), \qquad \begin{array}{c} t = T - 1, T - 2, \dots, 1 \\ 1 \le i \le N \end{array}$$

### **Backward Procedure Illustration**



### **Finding Optimal State Sequences**

- One criterion chooses states,  $q_t$ , which are *individually* most likely
  - This maximizes the expected number of correct states
- Let us define  $\gamma_t(i)$  as the probability of being in state  $s_i$  at time t, given the observation sequence and the model, i.e.

$$\gamma_t(i) = P(q_t = s_i | \boldsymbol{O}, \lambda)$$
  $\sum_{i=1}^N \gamma_t(i) = 1, \quad \forall t$ 

• Then the individually most likely state,  $q_t$ , at time t is:

 $q_t = \underset{1 \le i \le N}{\operatorname{argmax}} \ \gamma_t(i) \qquad 1 \le t \le T$ 

• Note that it can be shown that:

$$\gamma_t(i) = \frac{\alpha_t(i)\beta_t(i)}{P(\boldsymbol{O}|\lambda)}$$

# **Finding Optimal State Sequences**

- The individual optimality criterion has the problem that the optimum state sequence may not obey state transition constraints
- Another optimality criterion is to choose the state sequence which maximizes  $P(Q, O|\lambda)$ ; This can be found by the *Viterbi* algorithm
- Let us define  $\delta_t(i)$  as the highest probability along a single path, at time t, which accounts for the first t observations, i.e.

$$\delta_t(i) = \max_{q_1, q_2, \dots, q_{t-1}} P(q_1 q_2 \dots q_{t-1}, q_t = s_i, o_1 o_2 \dots o_t | \lambda)$$

- By induction:  $\delta_{t+1}(j) = [\max_{i} \delta_t(i)a_{ij}]b_j(o_{t+1})$
- To retrieve the state sequence, we must keep track of the state sequence which gave the best path, at time *t*, to state *s<sub>i</sub>* 
  - We do this in a separate array  $\psi_t(i)$



1. Initialization:

$$\begin{aligned} \delta_1(i) &= \pi_i b_i(o_1), & 1 \leq i \leq N \\ \psi_1(i) &= 0 \end{aligned}$$

2. Recursion:

$$\begin{split} \delta_t(j) &= \max_{1 \leq i \leq N} [\delta_{t-1}(i)a_{ij}] b_j(o_t), & 2 \leq t \leq T \quad 1 \leq j \leq N \\ \psi_t(j) &= \operatorname*{argmax}_{1 \leq i \leq N} [\delta_{t-1}(i)a_{ij}], & 2 \leq t \leq T \quad 1 \leq j \leq N \end{split}$$

3. Termination:

$$P^* = \max_{1 \le i \le N} [\delta_T(i)]$$
  
$$q_T^* = \operatorname*{argmax}_{1 \le i \le N} [\delta_T(i)]$$

4. Path (state-sequence) backtracking:

 $q_t^* = \psi_{t+1}(q_{t+1}^*), \quad t = T-1, T-2, \dots, 1$ 

#### Computation $\approx N^2 T$

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### The Viterbi Algorithm: An Example



 $O = \{a a b b\}$ 



### The Viterbi Algorithm: An Example (cont'd)

	0	а	аа	aab	aabb
<i>S</i> <sub>1</sub>	1.0	<i>s</i> <sub>1</sub> , <i>a</i> .4	<i>s</i> <sub>1</sub> , <i>a</i> .16	<i>s</i> <sub>1</sub> , <i>b</i> .016	<i>s</i> <sub>1</sub> , <i>b</i> .0016
<i>s</i> <sub>2</sub>	<i>s</i> <sub>1</sub> , 0 .2	$s_1, 0$ .08 $s_1, a$ .21 $s_2, a$ .04	<i>s</i> <sub>1</sub> , 0 .032 <i>s</i> <sub>1</sub> , <i>a</i> .084 <i>s</i> <sub>2</sub> , <i>a</i> .042	$s_1, 0$ .0032 $s_1, b$ .0144 $s_2, b$ .0168	$s_1, 0$ .00032 $s_1, b$ .00144 $s_2, b$ .00336
<i>S</i> <sub>3</sub>	<i>s</i> <sub>2</sub> , 0 .02	<i>s</i> <sub>2</sub> , 0 .021 <i>s</i> <sub>2</sub> , <i>a</i> .03	<i>s</i> <sub>2</sub> , 0 .0084 <i>s</i> <sub>2</sub> , <i>a</i> .0315	<i>s</i> <sub>2</sub> , 0 .00168 <i>s</i> <sub>2</sub> , <i>b</i> .0294	<i>s</i> <sub>2</sub> , 0 .000336 <i>s</i> <sub>2</sub> , <i>b</i> .00588



### Matching Using Forward-Backward Algorithm

	0	а	аа	aab	aabb
<i>S</i> <sub>1</sub>	1.0	<i>s</i> <sub>1</sub> , <i>a</i> .4	<i>s</i> <sub>1</sub> , <i>a</i> .16	<i>s</i> <sub>1</sub> , <i>b</i> .016	<i>s</i> <sub>1</sub> , <i>b</i> .0016
<i>s</i> <sub>2</sub>	<i>s</i> <sub>1</sub> , 0.2	$s_1, 0$ .08 $s_1, a$ .21 $s_2, a$ .04	<i>s</i> <sub>1</sub> , 0 .032 <i>s</i> <sub>1</sub> , <i>a</i> .084 <i>s</i> <sub>2</sub> , <i>a</i> .066	$s_1, 0$ .0032 $s_1, b$ .0144 $s_2, b$ .0364	$s_1, 0$ .00032 $s_1, b$ .00144 $s_2, b$ .0108
<i>S</i> <sub>3</sub>	<i>s</i> <sub>2</sub> , 0.02	<i>s</i> <sub>2</sub> , 0 .033 <i>s</i> <sub>2</sub> , <i>a</i> .03	<i>s</i> <sub>2</sub> , 0 .0182 <i>s</i> <sub>2</sub> , <i>a</i> .0495	<i>s</i> <sub>2</sub> , 0 .0054 <i>s</i> <sub>2</sub> , <i>b</i> .0637	<i>s</i> <sub>2</sub> , 0 .001256 <i>s</i> <sub>2</sub> , <i>b</i> .0189



# **Baum-Welch Re-estimation**

- Baum-Welch re-estimation uses EM to determine ML parameters
- Define  $\xi_t(i, j)$  as the probability of being in state  $s_i$  at time t and state  $s_j$  at time t + 1, given the model and observation sequence

$$\xi_t(i,j) = P(q_t = s_i, q_{t+1} = s_j | \boldsymbol{O}, \lambda)$$

- Then:  $\xi_t(i,j) = \frac{\alpha_t(i)a_{ij}b_j(o_{t+1})\beta_{t+1}(j)}{P(\boldsymbol{O}|\lambda)}$   $\gamma_t(i) = \sum_{j=1}^N \xi_t(i,j)$
- Summing  $\gamma_t(i)$  and  $\xi_t(i, j)$ , we get:

 $\sum_{t=1}^{T-1} \gamma_t(i) = \text{ expected number of transitions from } s_i$  $\sum_{t=1}^{T-1} \xi_t(i,j) = \text{ expected number of transitions from } s_i \text{ to } s_j$ 

### **Baum-Welch Re-estimation Procedures**



### **Baum-Welch Re-estimation Formulas**

 $\bar{\pi}$  = expected number of times in state  $s_i$  at t = 1=  $\gamma_1(i)$ 

 $\bar{a}_{ij} = \frac{\text{expected number of transitions from state } s_i \text{ to } s_j}{\text{expected number of transitions from state } s_i}$  $= \frac{\sum_{t=1}^{T-1} \xi_t(i,j)}{\sum_{t=1}^{T-1} \gamma_t(i)}$ 

 $\bar{b}_{j}(k) = \frac{\text{expected number of times in state } s_{j} \text{ with symbol } v_{k}}{\text{expected number of times in state } s_{j}}$  $= \frac{\sum_{t=1}^{T} \gamma_{t}(j)}{\sum_{t=1}^{T} \gamma_{t}(j)}$ 

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### **Baum-Welch Re-estimation Formulas**

- If  $\lambda = (\mathbf{A}, \mathbf{B}, \pi)$  is the initial model, and  $\overline{\lambda} = (\mathbf{A}, \mathbf{B}, \mathbf{\pi})$  is the re-estimated model. Then it can be proved that either:
  - 1. The initial model,  $\lambda$ , defines a critical point of the likelihood function, in which case  $\bar{\lambda} = \lambda$ , or
  - 2. Model  $\overline{\lambda}$  is more likely than  $\lambda$  in the sense that  $P(\boldsymbol{O}|\overline{\lambda}) > P(\boldsymbol{O}|\lambda)$ , i.e., we have found a new model  $\overline{\lambda}$  from which the observation sequence is more likely to have been produced.
- Thus we can improve the probability of  $\boldsymbol{O}$  being observed from the model if we iteratively use  $\bar{\lambda}$  in place of  $\lambda$  and repeat the re-estimation until some limiting point is reached. The resulting model is called the maximum likelihood HMM.

# **Multiple Observation Sequences**

- Speech recognition typically uses left-to-right HMMs. These HMMs can not be trained using a single observation sequence, because only a small number of observations are available to train each state. To obtain reliable estimates of model parameters, one must use multiple observation sequences. In this case, the re-estimation procedure needs to be modified.
- Let us denote the set of *K* observation sequences as

 $O = \{O^{(1)}, O^{(2)}, \dots, O^{(K)}\}$ 

where  $O^{(k)} = \{o_1^{(k)}, o_2^{(k)}, ..., o_{T_k}^{(k)}\}$  is the *k*-th observation sequence.

 Assume that the observations sequences are mutually independent, we want to estimate the parameters so as to maximize

$$P(\boldsymbol{O} \mid \lambda) = \prod_{k=1}^{K} P(\boldsymbol{O}^{(k)} \mid \lambda) = \prod_{k=1}^{K} P_k$$

# Multiple Observation Sequences (cont'd)

• Since the re-estimation formulas are based on frequency of occurrence of various events, we can modify them by adding up the individual frequencies of occurrence for each sequence

$$\bar{a}_{ij} = \frac{\sum_{k=1}^{K} \sum_{t=1}^{T_k-1} \xi_t^k(i,j)}{\sum_{k=1}^{K} \sum_{t=1}^{T_k-1} \gamma_t^k(i)} = \frac{\sum_{k=1}^{K} \frac{1}{P_k} \sum_{t=1}^{T_k-1} \alpha_t^k(i) a_{ij} b_j(o_{t+1}^{(k)}) \beta_{t+1}^k(j)}{\sum_{k=1}^{K} \frac{1}{P_k} \sum_{t=1}^{T_k-1} \alpha_t^k(i) \beta_t^k(i)}$$

$$\bar{b}_j(\ell) = \frac{\sum_{k=1}^{K} \sum_{t=1}^{T_k} \gamma_t^k(j)}{\sum_{k=1}^{K} \sum_{t=1}^{T_k} \gamma_t^k(j)} = \frac{\sum_{k=1}^{K} \frac{1}{P_k} \sum_{t=1}^{T_k} \alpha_t^k(i) \beta_t^k(i)}{\sum_{k=1}^{K} \frac{1}{P_k} \sum_{t=1}^{T_k} \alpha_t^k(i) \beta_t^k(i)}$$



• Word-based HMMs are appropriate for small vocabulary speech recognition. For large vocabulary ASR, sub-word-based (e.g., phone-based) models are more appropriate.



SUB-WORD UNIT



# Phone-based HMMs (cont'd)

• The phone models can have many states, and words are made up from a concatenation of phone models.



### **Continuous Density Hidden Markov Models**

- A *continuous density* HMM replaces the discrete observation probabilities,  $b_j(k)$ , by a continuous PDF  $b_j(\mathbf{x})$
- A common practice is to represent  $b_j(\mathbf{x})$  as a mixture of Gaussians:

$$b_j(\boldsymbol{x}) = \sum_{k=1}^M c_{jk} N[\boldsymbol{x}, \mu_{jk}, \boldsymbol{\Sigma}_{jk}] \qquad 1 \le j \le N$$

where  $c_{jk}$  is the mixture weight  $c_{jk} \ge 0$   $(1 \le j \le N, 1 \le k \le M, \text{ and } \sum_{k=1}^{M} c_{jk} = 1, 1 \le j \le N),$  N is the normal density, and  $\mu_{jk}$  and  $\Sigma_{jk}$  are the mean vector and covariance matrix

associated with state *j* and mixture *k*.

# Acoustic Modelling Variations

- Semi-continuous HMMs first compute a VQ codebook of size M
  - The VQ codebook is then modelled as a family of Gaussian PDFs
  - Each codeword is represented by a Gaussian PDF, and may be used together with others to model the acoustic vectors
  - From the CD-HMM viewpoint, this is equivalent to using the same set of *M* mixtures to model all the states
  - It is therefore often referred to as a *Tied Mixture* HMM
- All three methods have been used in many speech recognition tasks, with varying outcomes
- For large-vocabulary, continuous speech recognition with sufficient amount (i.e., tens of hours) of training data, CD-HMM systems currently yield the best performance, but with considerable increase in computation



- Scaling: to prevent underflow
- Segmental K-means Training: to train observation probabilities by first performing Viterbi alignment
- Initial estimates of  $\lambda$ : to provide robust models
- Pruning: to reduce search computation



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