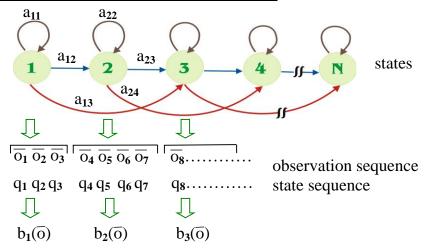
2.0 Fundamentals of Speech Recognition

References for 2.0

1.3, 3.3, 3.4, 4.2, 4.3, 6.4, 7.2, 7.3, of Bechetti

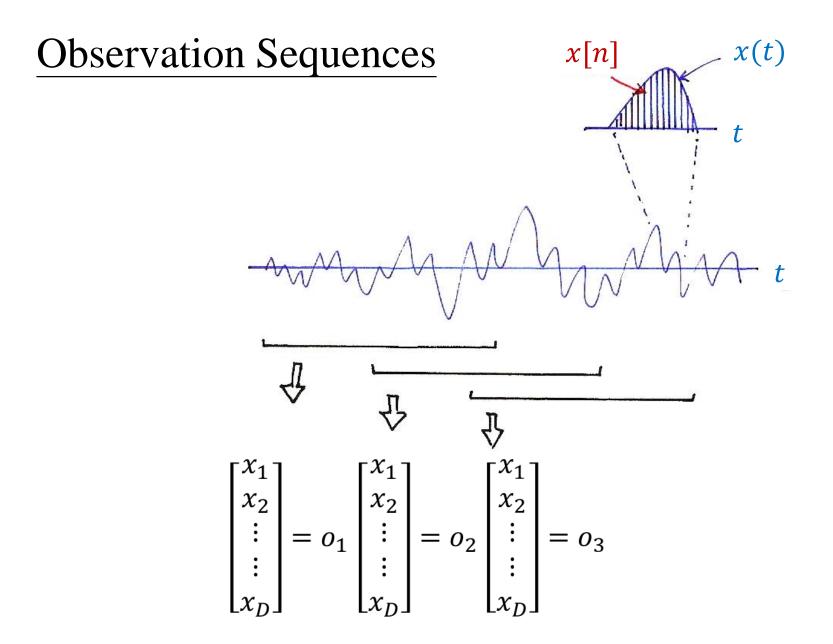
2.0 Fundamentals of Speech Recognition

Hidden Markov Models (HMM)

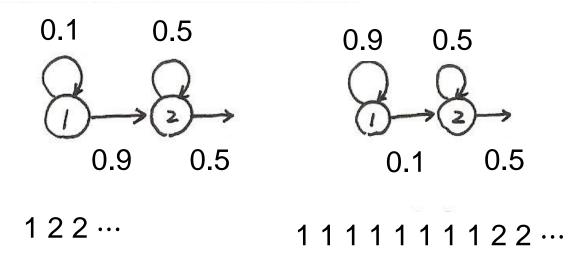


• Formulation

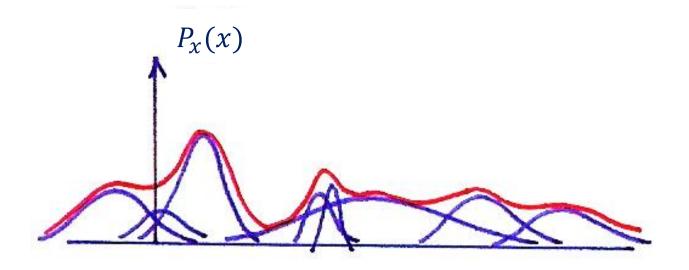
$$\begin{split} \overline{o}_t &= [x_1, \, x_2, \, \dots x_D]^T \quad \text{feature vectors for a frame at time t} \\ q_t &\in \left\{1, 2, 3 \dots N \right. \right\} \qquad \text{state number for feature vector } o_t \\ A &= [a_{ij}] \;, \qquad a_{ij} = Prob[\; q_t = j \; | \; q_{t-1} = i \;] \\ \qquad \qquad \text{state transition probability} \\ B &= [b_j(\overline{o}), \, j \underset{M}{=} 1, 2, \dots N] \quad \text{observation (emission) probability} \\ b_j(\overline{o}) &= \sum_{k=1}^{L} c_{jk} b_{jk}(\overline{o}) \quad \text{Gaussian Mixture Model (GMM)} \\ b_{jk}(\overline{o}) &: \text{multi-variate Gaussian distribution} \\ \qquad \qquad \text{for the k-th mixture (Gaussian) of the j-th state} \\ M &: \text{total number of mixtures} \\ \sum_{k=1}^{M} c_{jk} &= 1 \\ \pi &= \left[\; \pi_1, \, \pi_2, \, \dots \pi_N \right] \quad \text{initial probabilities} \\ \qquad \qquad \pi_i &= Prob[q_1 = i] \\ \text{HMM} : (\; A \;, \; B, \; \pi \;) &= \lambda \end{split}$$



State Transition Probabilities



1-dim Gaussian Mixtures



• Gaussian Random Variable X

$$f_{\mathbf{X}}(x) = \frac{1}{(2\pi\sigma^2)^{1/2}} e^{-(x-\mathbf{m})^2/2\sigma^2}$$

• Multivariate Gaussian Distribution for n Random Variables

$$\overline{X} = [X_1, X_2, \dots, X_n]^t$$

$$f_{\overline{X}}(\overline{x}) = \frac{1}{(2\pi)^{n/2} \Delta^{1/2}} e^{-\frac{1}{2}[(x-\mu)^t \sum^{-1}(x-\mu)]}$$

$$\overline{\mu} = [\mu_{X_1}, \mu_{X_2}, \dots, \mu_{X_n}]^t$$

$$\sum = [\sigma_{ij}], \text{ covariance matrix}$$

$$\sigma_{ij} = E[(X_i - \mu_{X_i})(X_j - \mu_{X_j})]$$

$$\Delta : \text{ determinant of } \sum$$

$$\sigma_{ij} = E[(x_i - \mu_{x_i})(x_j - \mu_{x_j})]$$

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Multivariate Gaussian Distribution

$$(\overline{\chi} - \overline{\mu})^{\dagger} \sum^{-1} (\overline{\chi} - \overline{\mu}) = \begin{pmatrix} \begin{pmatrix} \chi_{1} \\ \chi_{2} \\ \vdots \\ \chi_{n} \end{pmatrix} - \begin{pmatrix} \chi_{1} \\ M_{2} \\ \vdots \\ M_{n} \end{pmatrix}^{\dagger} \sum^{-1} (\overline{\chi} - \overline{\mu})$$

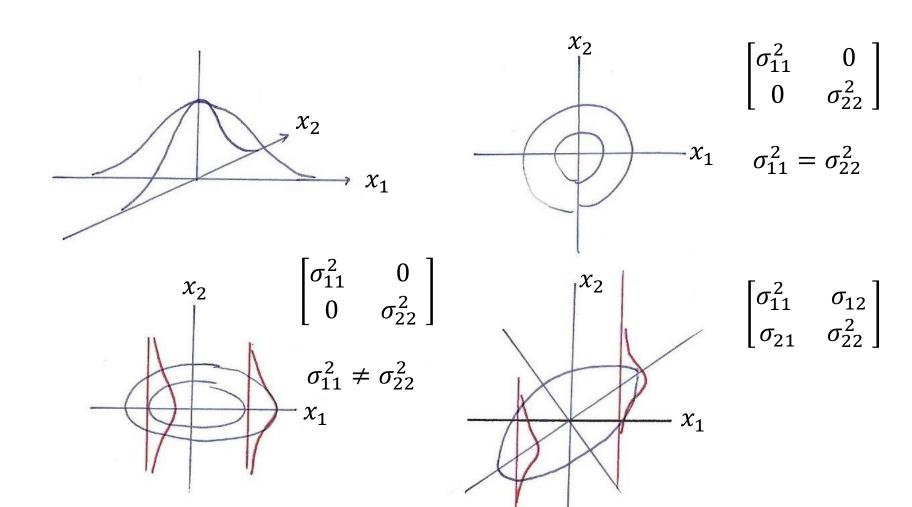
$$= \begin{bmatrix} \chi_{1} - M_{1} & \chi_{2} - M_{2} & \cdots & \chi_{n} - M_{n} \end{bmatrix} \sum^{-1} \begin{bmatrix} \chi_{1} - M_{1} \\ \chi_{2} - M_{2} \\ \vdots \\ \chi_{n} - M_{n} \end{bmatrix}$$

$$= (\chi_{1} - M_{1})^{2} + (\chi_{2} - M_{2})^{2} + \cdots , \qquad \text{if } \sum \begin{bmatrix} \chi_{1} - M_{1} \\ \chi_{2} - M_{2} \\ \vdots \\ \chi_{n} - M_{n} \end{bmatrix}$$

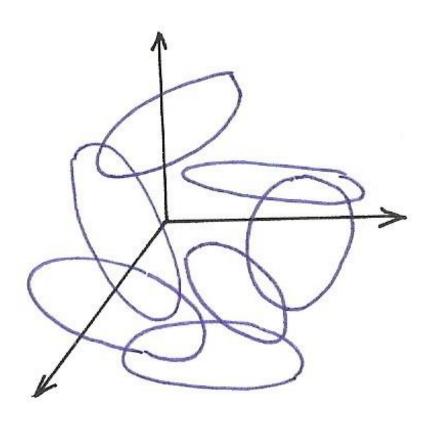
$$= (\chi_{1} - M_{1})^{2} + (\chi_{2} - M_{2})^{2} + \cdots , \qquad \text{if } \sum \begin{bmatrix} \chi_{1} - M_{1} \\ \chi_{2} - M_{2} \\ \vdots \\ \chi_{n} - M_{n} \end{bmatrix}$$

$$\sigma_{ij} = E[(\chi_{i} - \mu_{\chi_{i}}) (\chi_{j} - \mu_{\chi_{j}})]$$

2-dim Gaussian



N-dim Gaussian Mixtures



Hidden Markov Models (HMM)

• Double Layers of Stochastic Processes

- hidden states with random transitions for time warping
- random output given state for random acoustic characteristics

• Three Basic Problems

(1) Evaluation Problem:

Given
$$\overline{O} = (\overline{o_1}, \overline{o_2}, ... \overline{o_t}... \overline{o_T})$$
 and $\lambda = (A, B, \pi)$ find Prob [$\overline{O} \mid \lambda$]

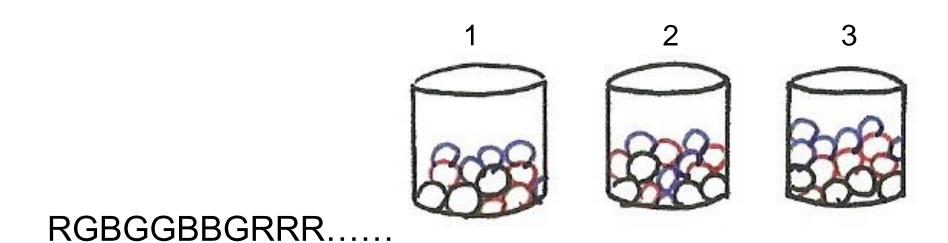
(2) Decoding Problem:

Given
$$\overline{O} = (\overline{o_1}, \overline{o_2}, ... \overline{o_t}... \overline{o_T})$$
 and $\lambda = (A, B, \pi)$ find a best state sequence $\overline{q} = (q_1, q_2, ... q_t, ... q_T)$

(3) Learning Problem:

Given O, find best values for parameters in λ such that Prob [$\overline{O} \mid \lambda$] = max

Simplified HMM



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Feature Extraction (Front-end Signal Processing)

• Pre-emphasis

$$H(z) = 1 - az^{-1}, 0 << a < 1$$

 $x[n] = x'[n] - ax'[n-1]$

- pre-emphasis of spectrum at higher frequencies

• Endpoint Detection (Speech/Silence Discrimination)

- short-time energy

$$E_{\mathbf{n}} = \sum_{m=-\infty}^{\infty} (\mathbf{x}[m])^2 \mathbf{w}[m-n]$$

- adaptive thresholds

• Windowing

$$Q_{\mathbf{n}} = \sum_{m=-\infty}^{\infty} T\{x[m]\}w[m-n]$$

T{ • } : some operator

w[m]: window shape

- Rectangular window

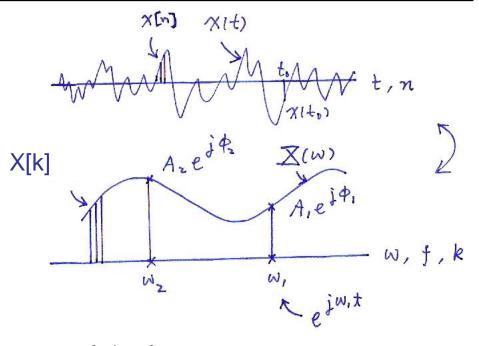
$$w[m] = \begin{cases} 1, & 0 < m \le L-1 \\ 0, & else \end{cases}$$

Hamming window

$$w[m] = \begin{cases} 0.54 - 0.46 \cos[\frac{2\pi m}{L}], 0 \le m \le L-1 \\ 0, \text{ else} \end{cases}$$

window length/shift/shape

Time and Frequency Domains



$$Re\{e^{j\omega_1 t}\} = \cos(\omega_1 t)$$

$$Re\{(A_1 e^{j\phi_1})(e^{j\omega_1 t})\} = A_1 \cos(\omega_1 t + \phi_1)$$

$$\vec{X} = \vec{a_1}\vec{i} + a_2\vec{j} + a_3\vec{k}$$

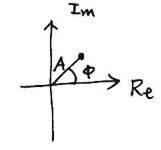
$$\vec{X} = \sum_{i} a_i x_i$$

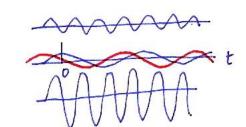
$$x(t) = \sum_{i} a_{i} (x_{i}(t))$$

time domain

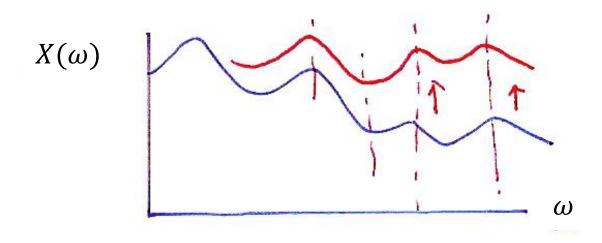
1-1 mapping
Fourier Transform
Fast Fourier Transform (FFT)

frequency domain





Pre-emphasis

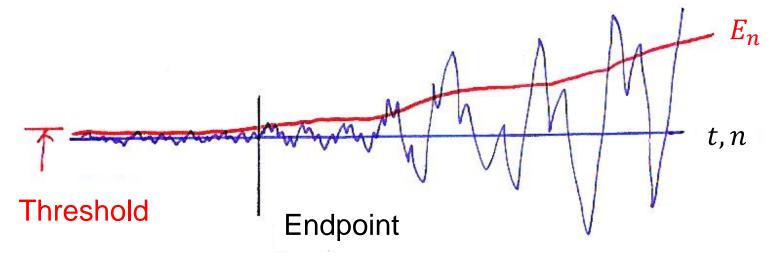


Pre-emphasis

$$H(z) = 1 - az^{-1},$$
 $0 << a < 1$
 $x[n] = x'[n] - ax'[n-1]$

pre-emphasis of spectrum at higher frequencies

Endpoint Detection



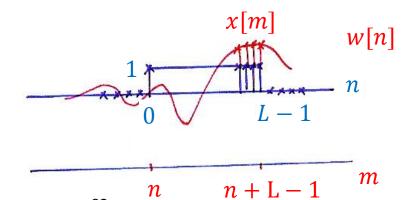
- Endpoint Detection (Speech/Silence Discrimination)
 - short-time energy

$$E_n = \sum_{m=-\infty}^{\infty} (x[m])^2 w[m-n]$$

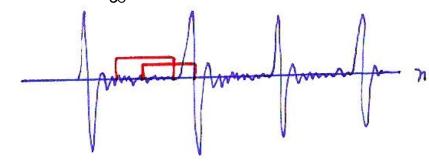
adaptive thresholds

Endpoint Detection

Rectangular Window

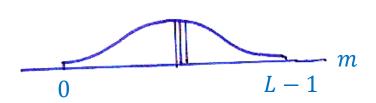


$$E_n = \sum_{-\infty}^{\infty} (x[m])^2 w[m-n]$$



Hamming Window

w[m]



Hamming window

$$w[m] = \begin{cases} 0.54 - 0.46 \cos[\frac{2\pi m}{L}], 0 \le m \le L - 1\\ 0, \text{ else} \end{cases}$$

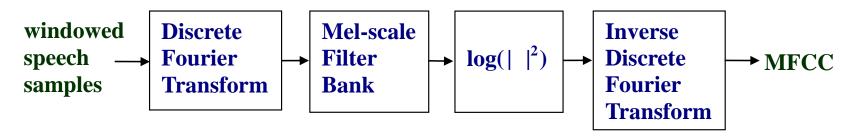
$$Q_n = \sum_{m = -\infty}^{\infty} T\{x[m]\} w[m - n]$$

$$T\{\bullet\} : \text{some operator}$$

W[m]: window shape

Feature Extraction (Front-end Signal Processing)

• Mel Frequency Cepstral Coefficients (MFCC)

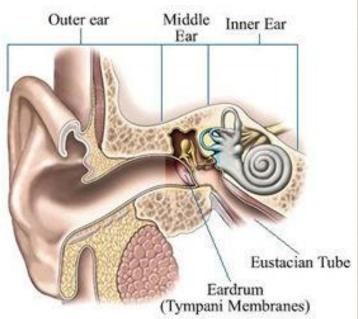


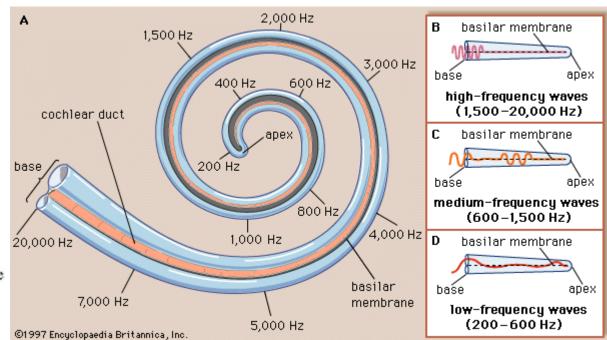
Mel-scale Filter Bank
 triangular shape in frequency/overlapped
 uniformly spaced below 1 kHz
 logarithmic scale above 1 kHz

• Delta Coefficients

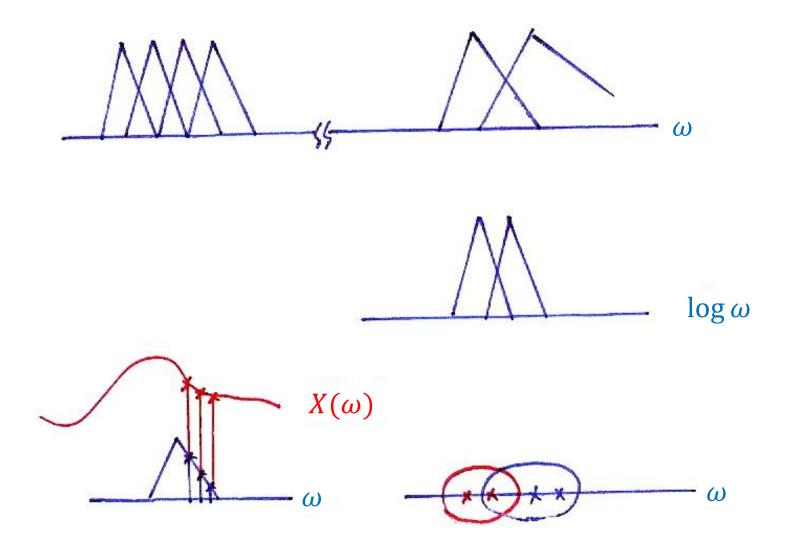
- 1st/2nd order differences

Peripheral Processing for Human Perception (P.34 of 7.0)

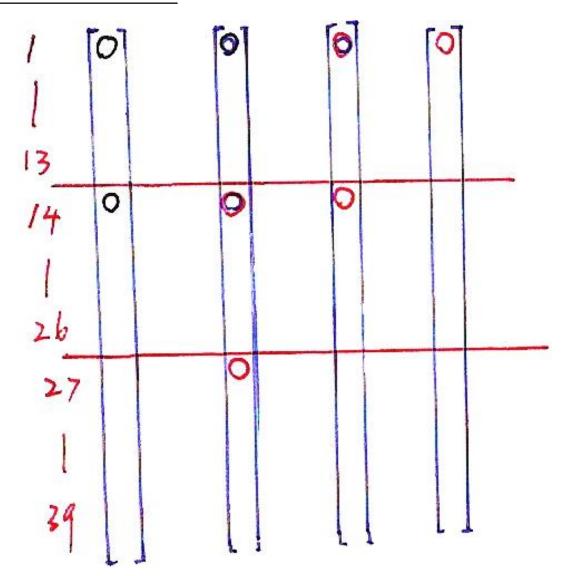




Mel-scale Filter Bank



Delta Coefficients



Language Modeling: N-gram

$$W = (w_1, w_2, w_3, ..., w_i, ..., w_R)$$
 a word sequence

- Evaluation of P(W)

$$P(W) = P(w_1) \prod_{i=2}^{R} P(w_i|w_1, w_2,...w_{i-1})$$

- Assumption:

$$P(w_i|w_1, w_2, ... w_{i-1}) = P(w_i|w_{i-N+1}, w_{i-N+2}, ... w_{i-1})$$

Occurrence of a word depends on previous N-1 words only

N-gram language models

$$N = 2$$
 : bigram $P(w_i | w_{i-1})$

$$N = 3$$
: tri-gram $P(w_i | w_{i-2}, w_{i-1})$

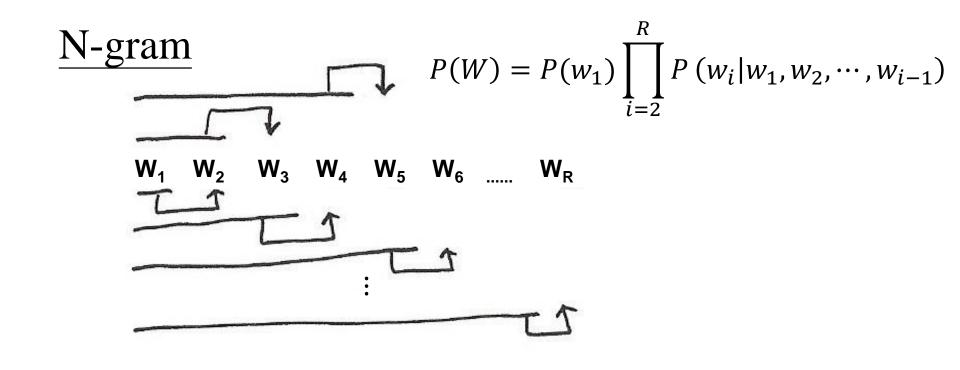
$$N = 4$$
: four-gram $P(w_i | w_{i-3}, w_{i-2}, w_{i-1})$

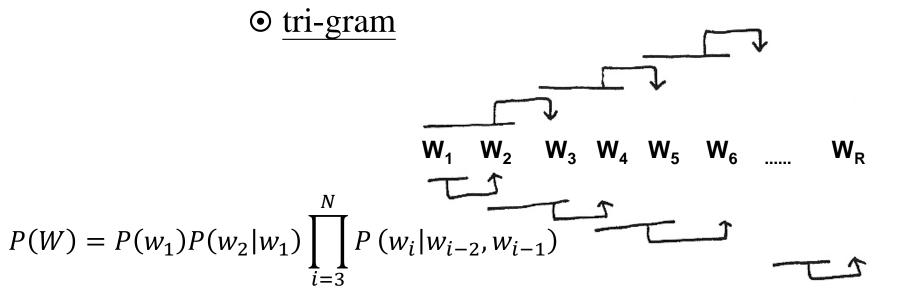
$$N = 1$$
 : unigram $P(w_i)$

probabilities estimated from a training text database

example: tri-gram model

$$P(W) = P(w_1) P(w_2|w_1) \prod_{i=3}^{N} P(w_i|w_{i-2}, w_{i-1})$$





Language Modeling

- Evaluation of N-gram model parameters unigram

$$P(w^{i}) = \frac{N(w^{i})}{\sum_{i=1}^{V} N(w^{j})}$$

wi: a word in the vocabulary

V: total number of different words in the vocabulary $N(\cdot)$ number of counts in the training text database

bigram

$$P(w^{j}|w^{k}) = \frac{N(< w^{k}, w^{j}>)}{N(w^{k})}$$

$$< w^{k}, w^{j} > : a \text{ word pair}$$

trigram

$$P(w^{j}|w^{k},w^{m}) = \frac{N(< w^{k},w^{m},w^{j}>)}{N(< w^{k},w^{m}>)}$$

smoothing – estimation of probabilities of rare events by statistical approaches

Prob [is| this] =
$$\frac{500}{50000}$$

Prob [a| this is] =
$$\frac{5}{500}$$

bigram

$$P(w^{j}|w^{k}) = \frac{N(\langle w^{k}, w^{j} \rangle)}{N(w^{k})}$$
$$\langle w^{k}, w^{j} \rangle: \text{ a word pair}$$

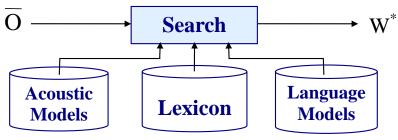
trigram

$$P(w^{j}|w^{k},w^{m}) = \frac{N(\langle w^{k},w^{m},w^{j}\rangle)}{N(\langle w^{k},w^{m}\rangle)}$$

Large Vocabulary Continuous Speech Recognition

$$\begin{array}{ll} \overline{W} = (w_1, w_2, \dots w_R) & \text{a word sequence} \\ \overline{O} = (\overline{o}_1, \overline{o}_2, \dots \overline{o}_T) & \text{feature vectors for a speech utterance} \\ W^* = \frac{\text{Arg Max}}{w} \text{Prob}(W|\overline{O}) & \text{MAP principle} \\ \text{Prob}(W|\overline{O}) = \frac{\text{Prob}(\overline{O}|W) \bullet P(W)}{P(\overline{O})} = \max & \text{A Posteriori Probability} \\ \text{Prob}(\overline{O}|W) \bullet P(W) = \max & \text{A Posteriori (MAP) Principle} \\ \uparrow & \uparrow \\ \text{by HMM} & \text{by language model} \\ \end{array}$$

• A Search Process Based on Three Knowledge Sources



- Acoustic Models : HMMs for basic voice units (e.g. phonemes)
- Lexicon: a database of all possible words in the vocabulary, each word including its pronunciation in terms of component basic voice units
- Language Models : based on words in the lexicon

Maximum A Posteriori Principle (MAP)

$$P(w_1)$$
 $P(w_2)$
+ $P(w_3)$
1.0

$$\vec{O} = (\vec{o}_1, \vec{o}_2, \vec{o}_3, \cdots)$$

weather parameters

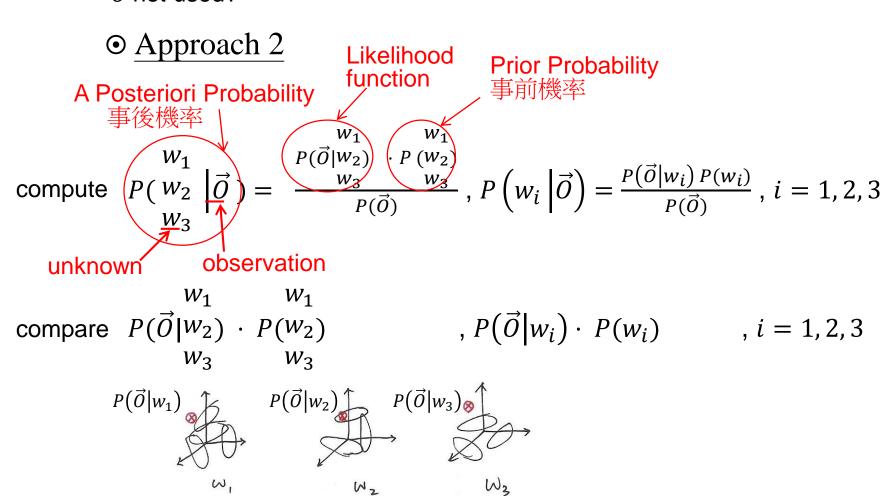
• Problem

given \vec{O} today, to predict W for tomorrow

Maximum A Posteriori Principle (MAP)

Approach 1

Comparing $P(w_1)$, $P(w_2)$, $P(w_3)$ \vec{O} not used?



Syllable-based One-pass Search

- Finding the Optimal Sentence from an Unknown Utterance Using 3 Knowledge Sources: Acoustic Models, Lexicon and Language Model
- Based on a Lattice of Syllable Candidates

