

# Speech Recognition : Statistical Methods



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# Agenda

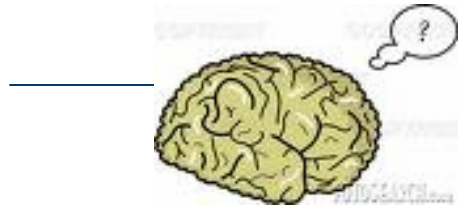
- Motivation.
- Introduction
- Automatic Speech Recognition
- Summary and Discussion.

# Introduction

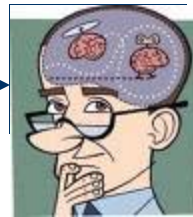
- Human Speech Dialog



Hear



Process

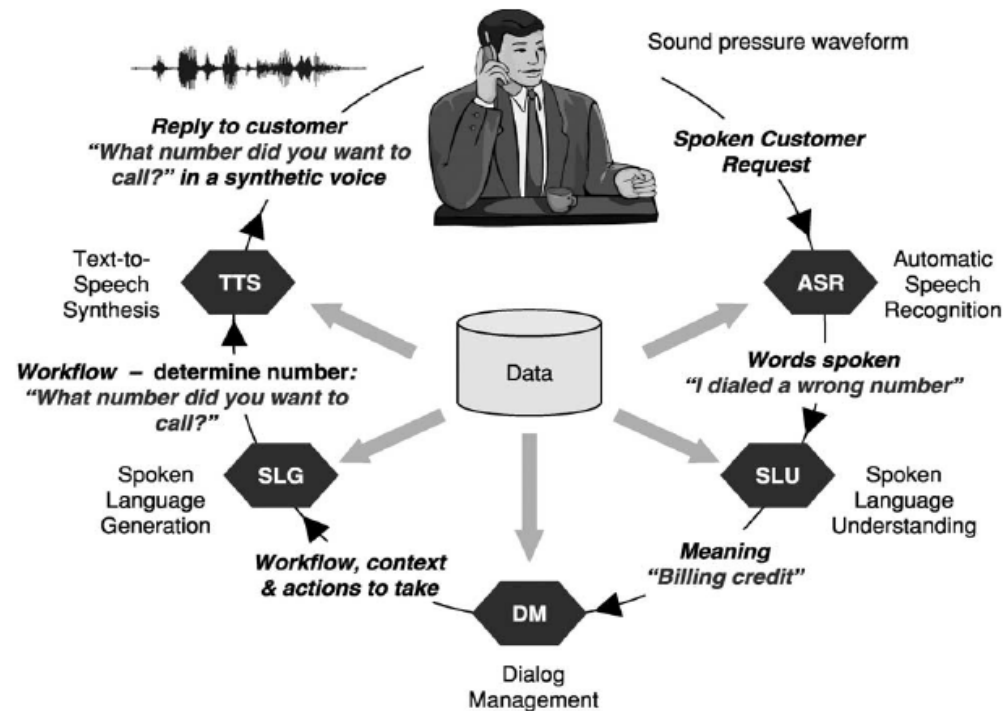


Decide



Act/Talk

# Dialog With a Machine



# What is ASR ?

- The process of recognizing the words in the speech is called *Automatic Speech Recognition* or *ASR*.

# How it works?

- ASR attempts to decode speech into best estimate of sentences using two steps:
  1. *Convert the Speech signal into Spectral feature vectors, which are usually measured in the span of 10 ms.*
  2. *Use syntactic decoder to generate every possible valid sentence , and select the best sentence .*

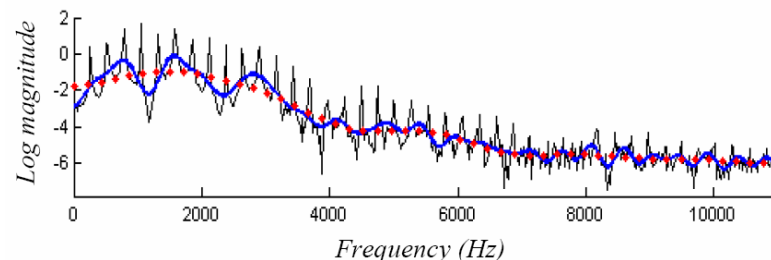
## Step -1 Deriving the Features

- MFCC are chosen :

**Reason:** They match certain characteristics of human perception.

- Feature vector:

- 13 MFCC.
- 13 MFCC First derivatives.
- 13 MFCC second derivatives



## Step-2 - Selection of the best sentence

- Mathematically: It can be presented as a MAP estimation problem.

$$W_{\max} = \arg \max_w P(W | X)$$

Where  $W$  is the word string



- Since by Baye's Rule

$$P(W | X) = \frac{P(X | W)P(W)}{P(X)}$$

Since  $P(X)$  doesn't depend on  $W$  :

$$W_{\max} = \arg \max_w P(W | X) \longrightarrow W_{\max} = \arg \max_w P(X | W)P(W)$$

Acoustic Model      Language Model

# Acoustic Modeling: *Probability Measures*

- Acoustic modeling uses probability measures to characterize sound realization using statistical models.
- Statistical Model called HMM is used for the solution.  
HMM- Hidden Markov Models.
- HMM model the spectral variability of each of the basic sounds using Gaussian distribution which is aligned with the speech training **set** and iteratively updated and improved until alignment is achieved.

# Markov Property

$$P(q_{t+1}=s_1|q_t=s_2) = 1/2$$

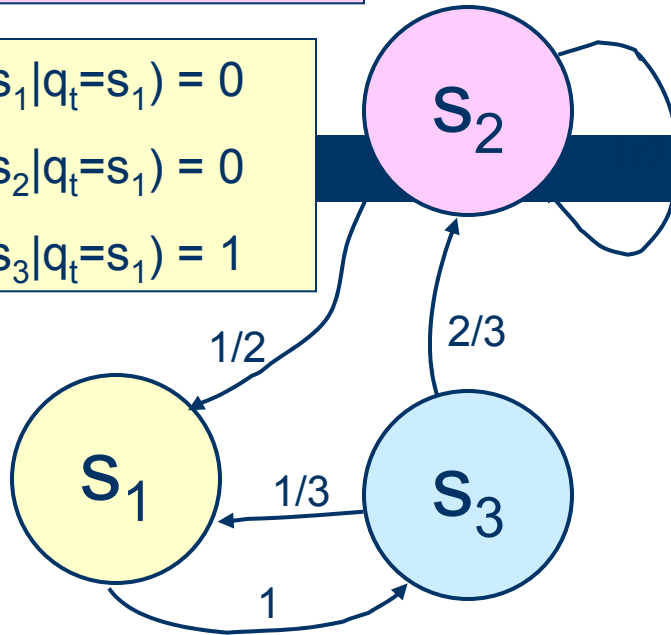
$$P(q_{t+1}=s_2|q_t=s_2) = 1/2$$

$$P(q_{t+1}=s_3|q_t=s_2) = 0$$

$$P(q_{t+1}=s_1|q_t=s_1) = 0$$

$$P(q_{t+1}=s_2|q_t=s_1) = 0$$

$$P(q_{t+1}=s_3|q_t=s_1) = 1$$



$N = 3$

$t=1$

$q_t=q_1=s_2$

$$P(q_{t+1}=s_1|q_t=s_3) = 1/3$$

$$P(q_{t+1}=s_2|q_t=s_3) = 2/3$$

$$P(q_{t+1}=s_3|q_t=s_3) = 0$$

$q_{t+1}$  is conditionally independent of  $\{q_{t-1}, q_{t-2}, \dots, q_1, q_0\}$  given  $q_t$ .

In other words:

$$P(q_{t+1} = s_j | q_t = s_i) =$$

$$P(q_{t+1} = s_j | q_t = s_i, \text{any earlier history})$$

The sequence of  $q$  is said to be a Markov chain, or to have the **Markov property** if the next state depends only upon the current state and not on any past states

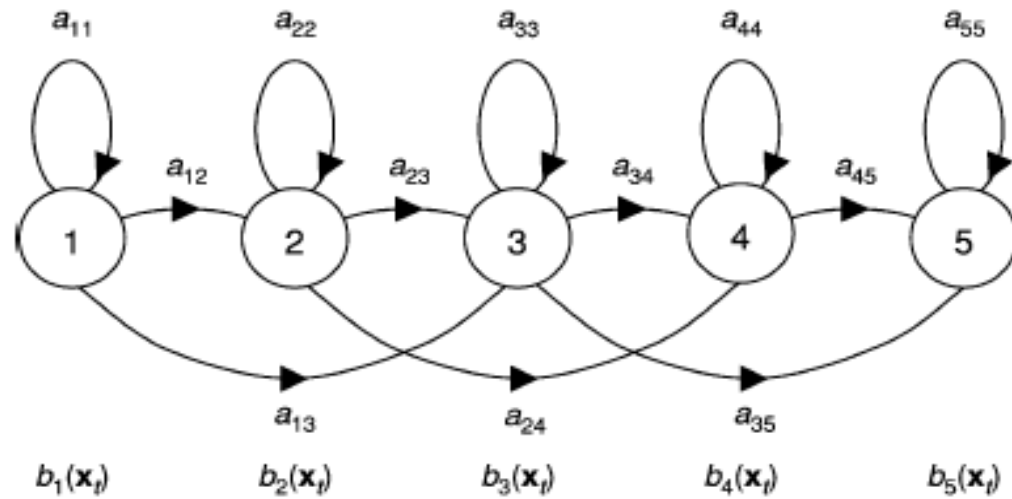
# *From Markov To Hidden Markov*

- The previous model assumes that each state can be **uniquely** associated with an observable event.
- To make the model more flexible, we will assume that the outcomes or observations of the model are a **probabilistic function** of each state

## **Why Hidden???**

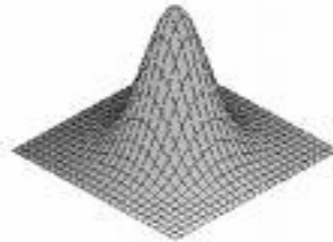
These are known as Hidden Markov Models (HMM), because the state sequence is not directly observable, it can only be approximated from the sequence of observations produced by the system

# HMM Model



**Figure 8** HMM for whole word model with five states.

- Each HMM state in our case is represented by a *Gaussians probability* density function.



▪

# Requirements for HMM

HMM is specified by a five-tuple  $(S, O, \Pi, A, B)$

1)  $S = \{1, 2, \dots, N\}$

Set of hidden states

$N$ : the number of states  $s_t$ : the state at time  $t$

2)  $O = \{o_1, o_2, \dots, o_M\}$

Set of observation symbols

$M$ : the number of observation symbols

3)  $\pi = \{\pi_i\}$   $\pi_i = P(s_0 = i)$   $1 \leq i \leq N$

The initial state distribution

4)  $A = \{a_{ij}\}$   $a_{ij} = P(s_t = j | s_{t-1} = i)$ ,  $1 \leq i, j \leq N$

State transition probability distribution

5)  $B = \{b_j(k)\}$   $b_j(k) = P(X_t = o_k | s_t = j)$   $1 \leq j \leq N, 1 \leq k \leq M$

Observation symbol probability distribution in state  $j$

# Creation of the Model

- Once the set of state transitions and state probability densities are specified we say that model  $\varphi$  has been created for the word or sub word unit.



# Three Problems

- **1. The Evaluation Problem** – Given a model  $\Phi$  and a sequence of observations  $X = (X_1, X_2, \dots, X_T)$ , what is the probability  $P(X | \Phi)$ ; i.e., the probability of the model that generates the observations?
- **2. The Decoding Problem** – Given a model  $\Phi$  and a sequence of observation  $X = (X_1, X_2, \dots, X_T)$ , what is the most likely state sequence  $S = (s_0, s_1, \dots, s_T)$  in the model that produces the observations?
- **3. The Learning Problem** – Given a model  $\Phi$  and a set of observations, how can we adjust the model parameter  $\hat{\Phi}$  to maximize the joint probability  $\prod_X P(X | \Phi)$ ?

So much Problems



## Solution-Problem 1

To calculate the probability (likelihood)  $P(\mathbf{X}|\Phi)$  of the observation sequence  $\mathbf{X} = (X_1, X_2, \dots, X_T)$ , given the HMM  $\Phi$ , the most intuitive way is to sum up the probabilities of all possible state sequences:

$$P(\mathbf{X} | \Phi) = \sum_{\text{all } \mathbf{S}} P(\mathbf{S} | \Phi) P(\mathbf{X} | \mathbf{S}, \Phi)$$

Applying Markov assumption:

$$P(\mathbf{S} | \Phi) = P(s_1 | \Phi) \prod_{t=2}^T P(s_t | s_{t-1}, \Phi) = \pi_{s_1} a_{s_0 s_1} \dots a_{s_{T-1} s_T} = a_{s_0 s_1} a_{s_1 s_2} \dots a_{s_{T-1} s_T}$$

# Continued.....

Applying output independent assumption:

$$\begin{aligned}P(\mathbf{X} | \mathbf{S}, \Phi) &= P(X_1^T | S_1^T, \Phi) = \prod_{t=1}^T P(X_t | s_t, \Phi) \\ &= b_{s_1}(X_1) b_{s_2}(X_2) \dots b_{s_T}(X_T) \\ P(\mathbf{X} | \Phi) &= \sum_{\text{all } \mathbf{S}} P(\mathbf{S} | \Phi) P(\mathbf{X} | \mathbf{S}, \Phi) \\ &= \sum_{\text{all } \mathbf{S}} a_{s_0 s_1} b_{s_1}(X_1) a_{s_1 s_2} b_{s_2}(X_2) \dots a_{s_{T-1} s_T} b_{s_T}(X_T)\end{aligned}$$

## Solution- Problem 2 (Viterbi Algorithm)

Viterbi algorithm picks and remembers the best path.

Define the best-path probability:

$$V_t(i) = P(X_1^t, S_1^{t-1}, s_t = i | \Phi)$$

$V_t(i)$  is the **probability of the most likely state sequence** at time **t**, which has generated the observation  $X_1^t$  (until time t) and ends in state **i**.

## Solution - Problem 3

- In order to train the HMM for each sub word unit a labeled training set of words and sentences is used.
- An efficient Training Algorithm Known as *Baum-Welch Algorithm* is used .
- It *aligns* various *sub word* units with *spoken inputs* and then *estimate* the appropriate *means, Covariance* and *mixture gains* for the distribution.
- The algorithm is a form of *hill climbing algorithm* and is *iterated* until a *stable alignment* of unit model and speech is obtained.

# Language Model

- **Purpose:** *To provide task syntax that defines acceptable spoken input sentences and enable computation of probability of the word string.*

# N- Gram model

- "*n*-gram" or " $(n - 1)$ -order Markov model".
- As per wikipedia: “An *n*-gram model predicts  $x_i$  based on  $x_{i-1}, x_{i-2}, \dots, x_{i-n}$ .” When used for language modeling independence assumptions are made so that each word depends only on the last *n* words”.

# Trigram model

Frequency Count of word triplet occurring in training data

$$P(w_i | w_{i-1}, w_{i-2}) = \frac{C(w_{i-2}, w_{i-1}, w_i)}{C(w_{i-2}, w_{i-1})}$$

Frequency Count of word duplet  $(w_{i-2}, w_{i-1})$  occurring in training data



# Problem of the Model

*N-grams are often highly in error due to problems of data sparseness in the training set. Hence for a text training set of millions of words, and a word vocabulary of several thousand words, more than 50% of word trigrams are likely to occur either once or not at all in the training set*

# Alternative provided.

$$\hat{P}(w_i|w_{i-1}, w_{i-2}) = p_3 \frac{C(w_{i-2}, w_{i-1}, w_i)}{C(w_{i-2}, w_{i-1})} + p_2 \frac{C(w_{i-1}, w_i)}{C(w_{i-1})} + p_1 \frac{C(w_i)}{\sum_i C(w_i)}$$

$$p_3 + p_2 + p_1 = 1$$

$$\sum_i C(w_i) = \text{size of text training corpus}$$

Where  $p_3$ ,  $p_2$ ,  $p_1$  are Smoothing Probabilities

# Language Complexity

- Defined as the average number of words that follow any given word of language
- **Mathematically** If  $P(W)$  is a language model where  $W = (w_1, w_2, \dots, w_Q)$  is a  $Q$  word sequence.
- **Language Perplexity**

$$PP(W) = 2^{H(W)} = P(w_1, w_2, \dots, w_Q)^{-1/Q}$$

as  $Q \rightarrow \infty$ .

# Continued.....

- Where  $H(W)$  is Entropy of trigram model defined as

$$H(W) = -\frac{1}{Q} \sum_{i=1}^Q \log_2 P(w_i | w_{i-1}, w_{i-2})$$

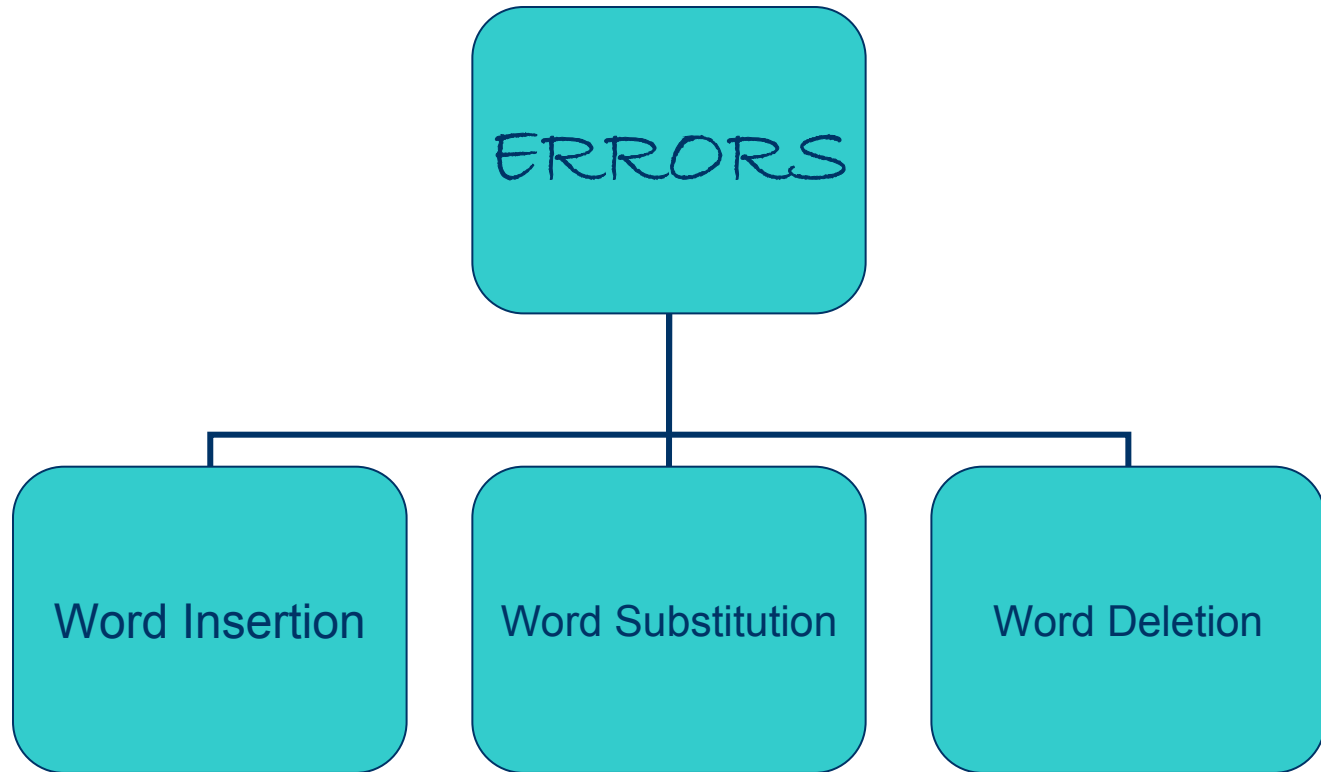
# Pattern Matching

- The task of pattern matching module is to combine probabilities from *acoustic models* , *Language model* and *word Lexicon* to find the optimal word Sequence having highest probability among all possible word sequences.

# Confidence Scoring

- **Goal:** *To identify the recognition error as well as out of vocabulary events.*

# Performance of *Speech Recognizer*



# Word Error Rate



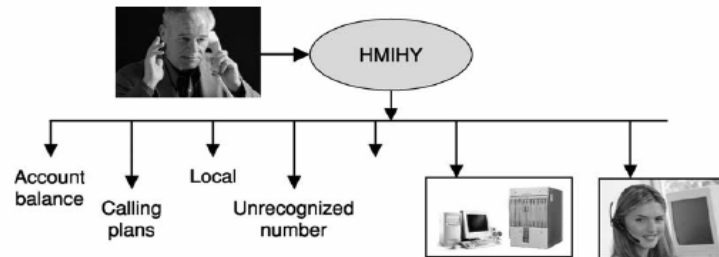
$$WER = \frac{NI + NS + ND}{|W|}$$

$|W|$  is the number of words in sentence



# Spoken Language Understanding

- **Goal:** Interpret the meaning of the key words and phrases in the recognized speech.
- **Example:** "Can I know my bill"



# Mathematics behind SLU

$$P(C|W) = P(W|C)P(C)/P(W) \quad C^* = \arg \max_c P(W|C)P(C)$$

Given the word sequence finding the best conceptual structure (meaning) using a combination of acoustic, linguistic and semantic scores.

Example: *Verizon* Speech recognition assistant.  
SLU is useful where vocabulary set is limited and restricted.

# Dialog Management

- **Role:** To combine meaning of current input speech of user with the current state of the system.
- Decides the next step in the interaction.
- Key tools:

# Spoken Language Generation

- **Role:** To translate the actions of Dialog module into textual representation.

# Text to Speech Module

- **Role:** Converts the text generated by *SLG* into synthetic naturally sounding speech

# Summary

- Paper Introduces us to the **ASR** that we have experiences most often while calling our wireless or phone company.
- However **ASR** is too *sensitive* to noise and till now *no* perfect ASR has been *designed* to show good performance under *noisy conditions*