Doctoral Course in Speech and Speaker Recognition

Part 1

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March - June 2007

March 29, 2007

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1

Introduction

- Course objective
 - deeper insight into basic and specific methods and algorithms
 - understanding not exact details of equations
 - no derivation of theorems and algorithms
 - Not covered
 - Phonetics, linguistics
 - Signal processing relevant parts (short time spectral analysis)
 - theory of probabilistics and pattern recognition overviewed
 - merit 7.5p in GSLT (5p in old system))
- Recommended background
 - GSLT or TMH course in "Speech technology" or equivalent

Recommended Background

- Basic mathematics, statistics and programming
- Acoustic phonetics
- Speech analysis
 - Short Time Spectral Analysis
 - MFCC
- Recognition
 - Dynamic programming and DTW
 - Fundamentals of hidden Markov models
 - Viterbi decoding
 - Phoneme-based speech recognition methods

Literature

- Spoken Language Processing
 - A Guide to Theory, Algorithm and System Development
 - X. Huang, A. Acero and H-W Hon
 - Contains theoretically heavy parts and many equations but it is not necessary to follow all derivations. The verbose explanations of their functions are easier to follow.
- Separate papers
 - Speaker recognition
 - Finite State Transducers
 - Bayesian Networks
 - Articulatory inspired approaches

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Course organization

- 3 lecture days
 - March 29-30, May 11
- Practical and computational exercises
- Write term paper + review + presentation
- Closing seminar day
 - June 8
 - Students' presentation of individual term papers

Course overview

- Day #1 March 29
 - Probability, Statistics and Information Theory (pp 73-131: 59 pages)
 - Pattern Recognition (pp 133-197: 65 pages)
 - Speech Signal Representations (pp 275-336 62 pages)
 - Hidden Markov Models (pp 377-413: 37 pages)
 - HTK tutorial & practical exercise
- Day #2 March 30
 - Acoustic Modeling (pp 415-475: 61 pages)
 - Environmental Robustness (pp 477-544: 68 pages)
 - Computational problems exercise
- Day #3 May 11
 - Language Modeling (pp 545-590: 46 pages)
 - Basic and Large-Vocabulary Search Algorithms (pp 591-685: 94 pages)
 - Applications and User Interfaces (pp 919-956: 38 pages)
 - Speaker recognition
- Day #4 June 8
 - Presentations of term papers & Solutions to exercises

Term paper

- Choose subject from a list or suggest one yourself
- Review each others reports
- Suggested topics
 - Further experiments on the practical exercise corpus
 - Phoneme recognition experiments on larger corpus (e.g. TIMIT or WAXHOLM)
 - Language models for speech recognition
 - Limitations in standard HMM and ways to reduce them
 - Pronunciation variation and their importance for speech recognition
 - New search methods
 - Techniques for robust recognition of speech
 - Speaker recognition topics: impersonation, forensics, channel and score normalisation
 - Own work and experiments after discussion with the teacher

Course Book

- The authors work for Microsoft Research
- Topics
 - Fundamental theory
 - Speech & Language, Statistics, Pattern Recognition, Information Theory
 - Speech processing
 - Speech recognition
 - Text-to-Speech
 - Spoken Language systems
- Historical Perspective and Further Reading in each chapter
- Important algorithms described in step-by-step
- Examples from Microsoft's own research

Book organization 1(2)

- Ch 1 Introduction
- Part I: Fundamental theory
 - Ch 2 Spoken Language Structure
 - Ch 3 Probability, Statistics and Information Theory
 - Ch 4 Pattern Recognition
- Part II: Speech Processing
 - Ch 5 Digital Signal Processing
 - Ch 6 Speech Signal Representation
 - Ch 7 Speech Coding

Book organization 2(2)

- Part III: Speech Recognition
 - Ch 8 Hidden Markov Models
 - Ch 9 Acoustic Modeling
 - Ch 10 Environmental Robustness
 - Ch 11 Language Modeling
 - Ch 12 Basic Search Algorithms
 - 13 Large-Vocabulary Search Algorithms
- Part IV: Text-to-Speech Systems
 - Ch 14 Text and Phonetic Analyses
 - Ch 15 Prosody
 - Ch 16 Speech Synthesis
- Part V: Spoken Language systems
 - Ch 17 Spoken Language Understanding
 - Ch 18 Applications and User Interfaces

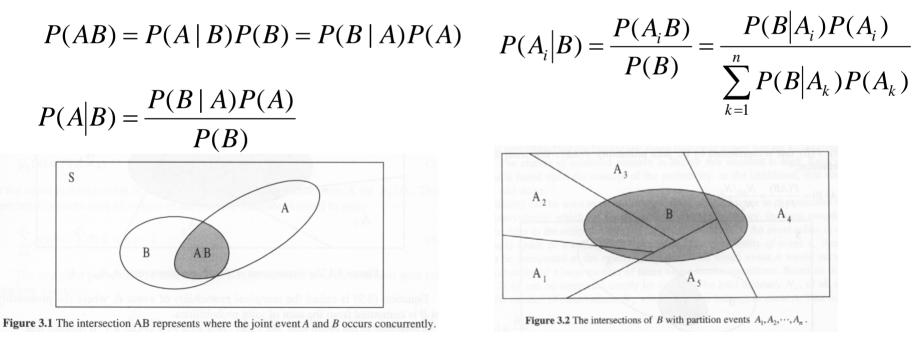
Ch 3. Probability, Statistics and Information Theory

- Conditional Probability and Bayes' Rule
- Covariance and Correlation
- Gaussian Distributions
- Bayesian Estimation and MAP Estimation
- Entropy
- Conditional Entropy
- Mutual Information and Channel Coding

Conditional Probability and Bayes' Rule

• Bayes' rule - the common basis for all pattern recognition

$$P(A|B) = \frac{P(AB)}{P(B)} = \frac{N_{AB} / N_S}{N_B / N_S}$$



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Bayes' rule in ASR

 $P(Word \mid Acoustics) = \frac{P(Acoustics \mid Word) \times P(Word)}{P(Acoustics)}$

- *P(Word | Acoustics)* is the *a posteriori probability* for a word sequence given the acoustic information
- *P*(*Acoustics | Word*) is the *probability* that the word sequence generates the acoustic information and is calculated from the training data
- *P(Word)* is given by the language model and is the *a priori probability* for the word sequence
- *P*(*Acoustics*) may be seen as *constant* since it is independent of the word sequence and may be ignored
- A combination of acoustic and language knowledge!

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Mean, Covariance and Correlation

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• Mean
$$\mu_x = E(X) = \sum x f(x)$$

- Variance $Var(X) = \sigma_x^2 = E[(X \mu_x)^2] = \frac{\sum (x_i \mu_x)^2}{n 1}$
- Covariance $Cov(X,Y) = E[(X \mu_x)(Y \mu_y)]$
- Correlation $\rho_{XY} = \frac{Cov(X,Y)}{\sigma_X \sigma_Y}$
- Multidimensional (Mean and variance vectors, covariance matrix)

$$E(\mathbf{X}) = \begin{bmatrix} E(X_1) \\ \vdots \\ E(X_n) \end{bmatrix} \qquad \qquad \mathbf{\Sigma}_{\mathbf{X}} = Cov(\mathbf{X}) = \begin{bmatrix} Cov(X_1, X_1) & \cdots & Cov(X_1, X_n) \\ \vdots & & \vdots \\ Cov(X_n, X_1) & \cdots & Cov(X_n, X_n) \end{bmatrix}$$

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14

Gaussian Distributions

• One-dimensional

$$f(x \mid \mu, \sigma^2) = N(\mu, \sigma^2) = \frac{1}{\sqrt{2\pi\sigma}} \exp\left[-\frac{(x-\mu)^2}{2\sigma^2}\right]$$

• Multivariate n-dimensional

$$f(\mathbf{X} = \mathbf{x} \mid \boldsymbol{\mu}, \boldsymbol{\Sigma}) = N(x; \boldsymbol{\mu}, \boldsymbol{\Sigma}) = \frac{1}{(2\pi)^{n/2}} \exp\left[-\frac{1}{2}(\mathbf{x} - \boldsymbol{\mu})^{'}\boldsymbol{\Sigma}^{-1}(\mathbf{x} - \boldsymbol{\mu})\right]$$

3.2 Estimation theory

- The basis for training a speech recogniser
- Estimate parameters of a probability distribution function
 - Minimum/Least Mean Squared Error Estimation
 - Minimize the difference between the distribution of the data and the model
 - Maximum Likelihood Estimation
 - Find the distribution with the maximum likelihood of generating the data
 - Bayesian Estimation and MAP Estimation
 - Assumes that we have a prior distribution that is modified by the new data

Minimum Mean / Least Squared Error Estimation

- Modify a model of the distribution to approximate the data with minimum error
- Find a function that predicts the value of Y from having observed X
- Estimation is made on joint observations of X and Y
- Minimize: $E(Y \hat{Y})^2 = E(Y g(X))^2$
- Minimum Mean Squared Error (MMSE) when the joint distribution is known
- Least Squared Error (LSE) when the distribution is unknown, only observation pairs (Ex. curve fitting)
- MMSE and LSE becomes equivalent with infinite number of samples

Maximum Likelihood Estimation (MLE)

- The most widely used parametric estimation method
- Find the distribution that maximizes the likelihood of generating the observed data

$$\mathbf{\Phi}_{MLE} = \arg\max_{\Phi} p(\mathbf{x} \,|\, \mathbf{\Phi})$$

- Corresponds to intuition
 - Max likelihood is normally achieved when the model has the same distribution as the observed data
- Example: univariate Gaussian pdf

$$\mu_{MLE} = \frac{1}{n} \sum_{k=1}^{n} x_n = E(x) \qquad \qquad \sigma_{MLE}^2 = \frac{1}{n} \sum_{k=1}^{n} (x_k - \mu_{MLE})^2 = E\left[(x_k - \mu_{MLE})^2\right]$$

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18

Bayesian and MAP Estimation

- Assumes that we have a prior distribution that is modified • by the new data
- Use Bayes' rule to find the new posterior distribution Φ

$$\boldsymbol{\Phi}_{MAP} = \arg \max_{\boldsymbol{\Phi}} p(\mathbf{x} \,|\, \boldsymbol{\Phi}) \, p(\boldsymbol{\Phi})$$

- Univariate Gaussian Mean: $\rho = \frac{\sigma^2 \mu + nv^2 \overline{x}_n}{\sigma^2 + nv^2}$ Var: $\tau^2 = \frac{\sigma^2 v^2}{\sigma^2 + nv^2}$
- MAP: Maximum A Posteriori probability is a Bayesian Estimator
- MAP becomes MLE with uniform prior distribution or infinite number of training data
- Valuable for limited training data and for adaptation March 29, 2007 Speech & speaker recognition course 2007

19

3.3 Significance testing

- For practical methods, see Chapter 4
- How certain are the achieved results?
 - The true result is within an interval around the measured value with a certain probability
 - Confidence level and interval
 - Rule of thumb in speech recognition (Doddington, 198x)
 - To assure that the true error rate is within the measured value \pm 30% with a probability of 0.9, requires at least 30 errors to have been made
- Is algorithm A better than B?
 - Matched-Pairs Test
 - Compare results on the same test data,
 - Sign Test
 - Magnitude difference Test
 - McNemar Test (Ch 4)

Entropy and Perplexity

- The information in seeing event x_i with probability $P(x_i)$ is defined as: $I(x_i) = \log \frac{1}{P(x_i)}$
- Entropy is the average information over all possible x values:

$$H(X) = E[I(X)] = \sum_{s} P(x_i)I(x_i) = \sum_{s} P(x_i)^2 \log \frac{1}{P(x_i)} = -\sum_{s} P(x_i)^2 \log(P(x_i))$$

- Perplexity $PP(X) = 2^{H(X)}$
 - The equivalent size of an imaginary list with equi-probable words
 - Perplexity for English letters: 2.39, English words: 130
- Conditional Entropy
 - Input X is distorted by a noisy channel into output Y
 - What is the uncertainty of X after observing Y?
 - Example: Confusion matrix

$$H(X|Y) = -\sum_{X} \sum_{Y} P(X = x_i, Y = y_j)^2 \log P(X = x_i|Y = y_j)$$

– If only diagonal values, the conditional entropy is 0

3.4.4 Mutual Information and Channel Coding

$$\begin{array}{c|c} X & Y \\ \hline & \text{Noisy channel} \end{array} \end{array}$$

- Mutual Information I(X;Y)
 - How much does Y tell us about X?
 - The difference between the entropy of X and the conditional entropy of X given Y

$$I(X;Y) = H(X) - H(X | Y) = \dots = E\left[\log \frac{P(X,Y)}{P(X)P(Y)}\right]$$

- H(X|Y)
 - "the amount of uncertainty remaining about X after Y is known"
 - represents the noise in the channel
- If X and Y are independent: I = 0

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