Acoustic Modeling for Speech Recognition

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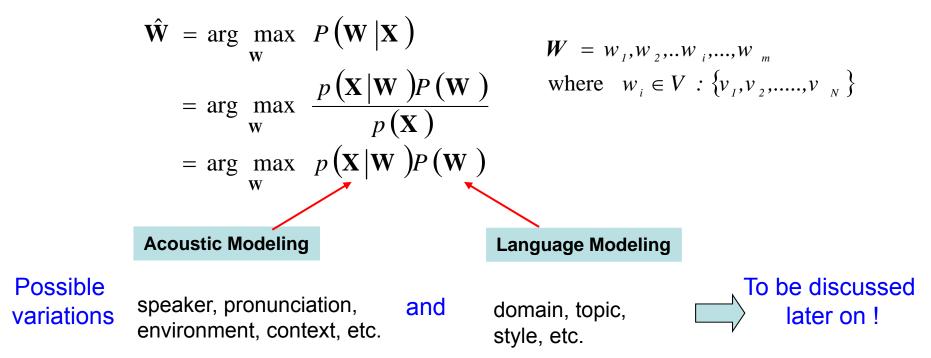
References:

1. X. Huang et. al. Spoken Language Processing. Chapters 8, 4

2. S. Young. The HTK Book (HTK Version 3.4)

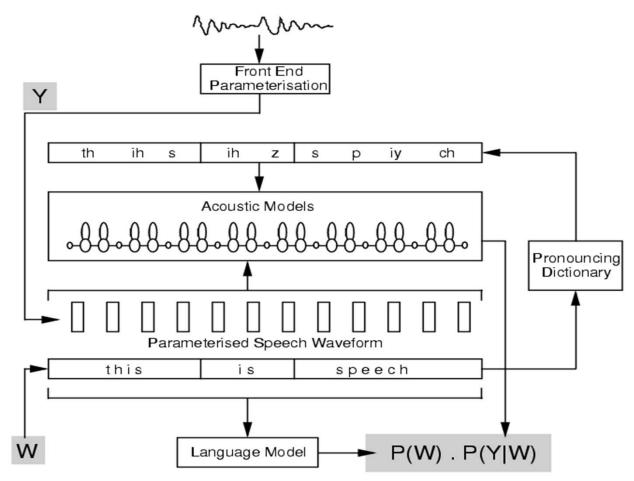
Introduction

• For an given acoustic observation $X = x_1, x_2, ..., x_n$, the goal of speech recognition is to find out an corresponding word sequence $W = w_1, w_2, ..., w_m$ that has the maximum posterior probability P(W | X)



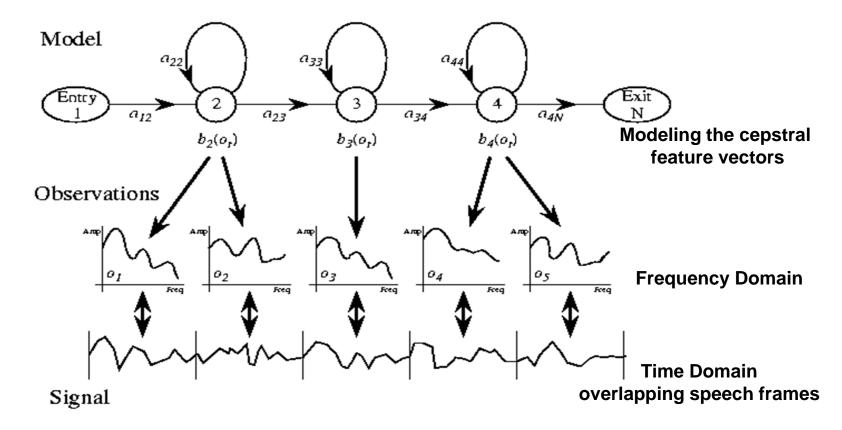
Introduction (cont.)

• An inventory of phonetic HMM models can constitute any given word in the pronunciation lexicon



Review: HMM Modeling

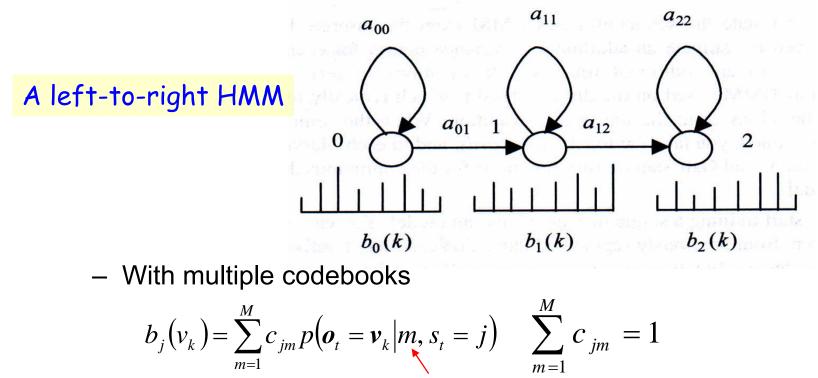
• Acoustic modeling using HMMs



Three types of HMM state output probabilities are frequently used

1. Discrete HMM (DHMM): $b_j(\mathbf{v}_k) = P(\mathbf{o}_t = \mathbf{v}_k | s_t = j)$

- The observations are quantized into a number of symbols
- The symbols are normally generated by a vector quantizer
 - · Each codeword is represented by a distinct symbol



codebook index

- 2. Continuous HMM (CHMM)
 - The state observation distribution of HMM is modeled by multivariate Gaussian mixture density functions (*M* mixtures)

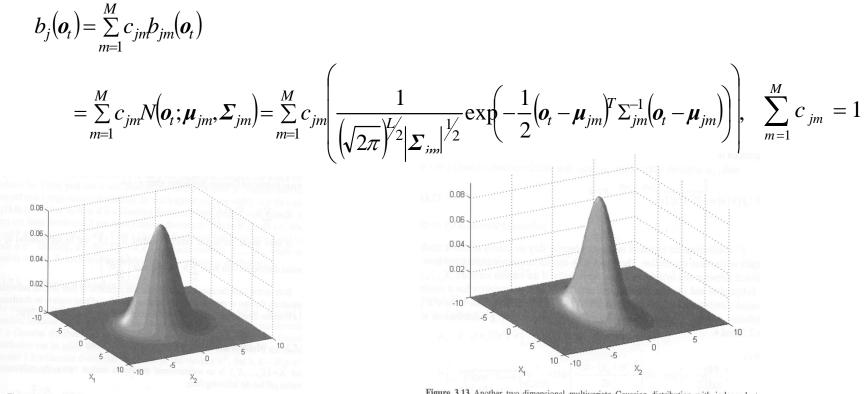
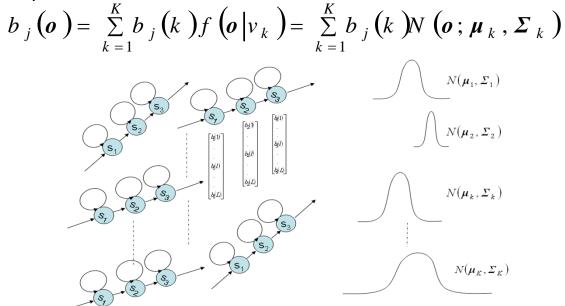


Figure 3.12 A two-dimensional multivariate Gaussian distribution with independent random variables x_1 and x_2 that have the same variance.

Figure 3.13 Another two-dimensional multivariate Gaussian distribution with independent random variable x_1 and x_2 which have different variances.

- 3. Semicontinuous or tied-mixture HMM (SCHMM)
 - The HMM state mixture density functions are tied together across all the models to form a set of shared kernels (shared Gaussians)



With multiple sets of shared Gaussians (or multiple codebooks)

$$b_j(\boldsymbol{o}) = \sum_{m=1}^M c_m \sum_{k=1}^K b_{jm}(k) f(\boldsymbol{o} | \boldsymbol{v}_{m,k}) = \sum_{m=1}^M c_m \sum_{k=1}^K b_{jm}(k) N(\boldsymbol{o}; \boldsymbol{\mu}_{m,k}, \boldsymbol{\Sigma}_{m,k})$$

Comparison of Recognition Performance

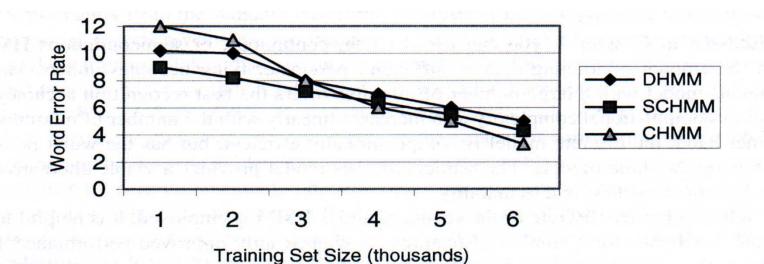


Figure 9.8 Continuous speaker-independent word recognition error rates of the discrete HMM (DHMM), SCHMM, and the continuous HMM (CHMM) with respect to the training set sizes (thousands of training sentences). Both the DHMM and SCHMM have multiple codebooks. The CHMM has 20 mixture diagonal Gaussian density functions.

The phenomena of "curse of dimensionality" occur when training data is small but the model complexity is high.

Choice of Appropriate Units for HMMs

- Issues for HMM Modeling units
 - Accurate: accurately represent the acoustic realization that appears in different contexts
 - Trainable: have enough data to estimate the parameters of the unit (or HMM model)
 - Generalizable: the model of any new word can be derived from a predefined unit inventory for task-independent speech recognition

Choice of Appropriate Units for HMMs (cont.)

- Comparison of different units
 - Word
 - Semantic meaning; capturing within-word coarticulation; can be accurately trained for small-vocabulary speech recognition, but is not generalizable for modeling unseen words and interword coarticulation

- Phone

subword

- More trainable and generalizable, but less accurate
- There are only about 50 context-independent phones in English and 30 in Mandarin Chinese
- Drawbacks: the realization of a phoneme is strongly affected by immediately neighboring phonemes (e.g., /t s/ and /t r/)

- Syllable

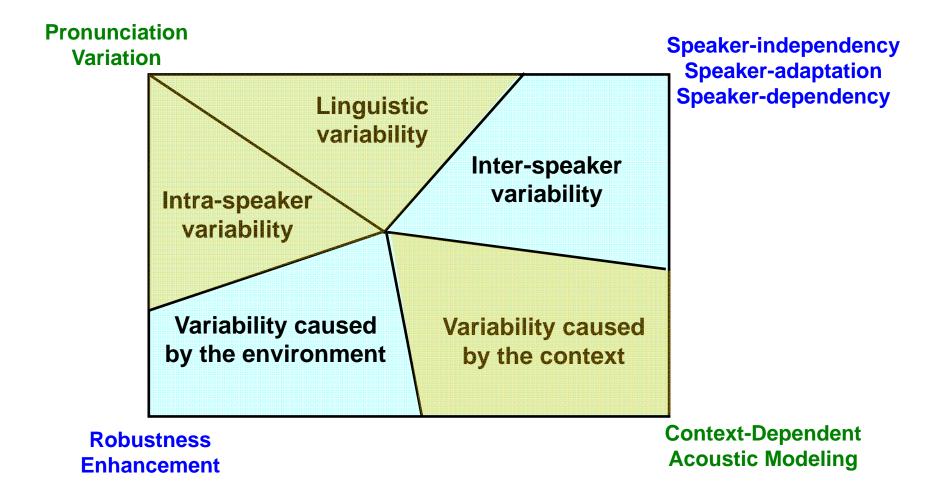
- A compromise between the word and phonetic models; syllables are larger than phone
- There only about 1,300 tone-dependent syllables in Chinese and 50 in Japanese. However, there are over 30,000 in English

Choice of Appropriate Units for HMMs (cont.)

• Phonetic Structure of Mandarin Syllables

	Syllables (1,345)	
Base	e-syllables (408)	
INITIAL's (21)	FINAL's (37)	Tones (4+1)
Phone-lil	ke Units/Phones (33)	_

Variability in the Speech Signals



- Context Variability
 - Context variability at word/sentence level
 - E.g., "*Mr. <u>Wright</u> should <u>write</u> to Ms. <u>Wright right</u> away about his <u>Ford or four door</u> Honda"*
 - Same pronunciation but different meaning (Wright , write , right)
 - Phonetically identical and semantically relevant (Ford or, four door)
 Pause or intonation information is needed
 - Context variability at phonetic level
 - The acoustic realization of phoneme /ee/ for word peat and wheel depends on its left and right context

the effect is more important in fast speech or spontaneous conversations, since many phonemes are not fully realized!

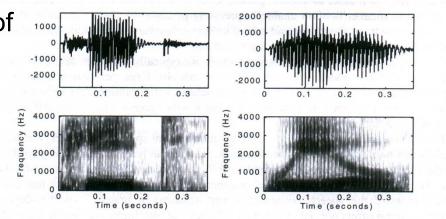


Figure 9.1 Waveforms and spectrograms for words *peat* (left) and *wheel* (right). The phoneme /*ee*/ is illustrated with two different left and right contexts. This illustrates that different contexts may have different effects on a phone.

- Style Variability (also including intra-speaker and linguistic variability)
 - Isolated speech recognition
 - Users have to pause between each word (a clear boundary between words)
 - Ambiguity such as that between "Ford or" and "four door" can be eliminated
 - But is unnatural to most people
 - Continuous speech recognition
 - Casual, spontaneous, and conversational
 - Higher speaking rate and coarticulation effects
 - Emotional changes also introduce more significantly variations

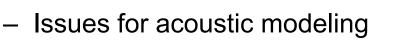
TABLE I

Speaking Rates of the Broadcast News Speech Used in This Study

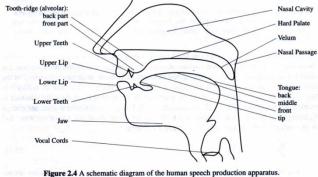
Data Types	Training				Testing	
Sources	PRS	UFO	VOH	VOT	Average	BCC
Average Speaking	4.9	6.2	4.8	5.7	5.4	5.8
Rates (characters/sec)		ĺ				

Statistics of the speaking rates of various broadcast new speech sources collected in Taiwan

- Speaker Variability •
 - Interspeaker
 - Vocal tract size, length and width of the neck and a range of physical characteristics
 - E.g., gender, age, dialect, health, education, and personal style
 - Intraspeaker
 - The same speaker is often unable to precisely produce the same utterance Tooth-ridge (alveolar):
 - The shape of the vocal tract movement and rate of delivery may vary from utterance to utterance



- Speaker-dependent (SD), speaker-independent (SI) and speaker-adaptive (SA) modeling
- Typically an SD system can reduce WER by more than 30% as compared with a comparable SI one

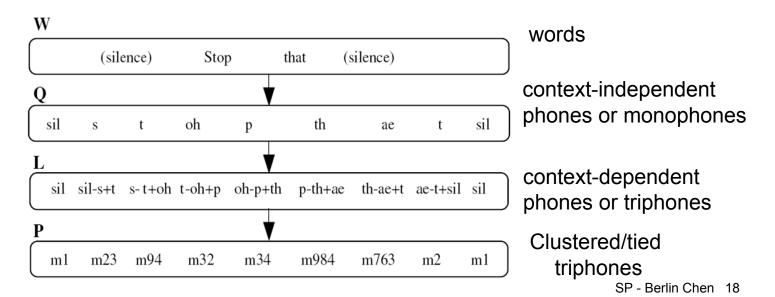


- Environment Variability
 - The world we live in is full of sounds of varying loudness from different sources
 - Speech recognition in hands-free or mobile environments remain one of the most severe challenges
 - The spectrum of noises varies significantly
 - Noise may also be present from the input device itself, such as microphone and A/D interface noises
 - We can reduce the error rates by using multi-style training or adaptive techniques
 - Environment variability remains as one of the most critical challenges facing today's state-of-the-art speech systems

Context Dependency

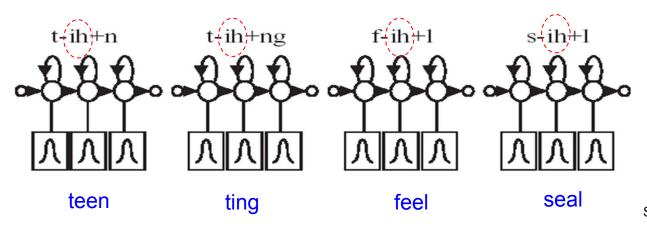
- Review: Phone and Phoneme
 - In speech science, the term phoneme is used to denote any of the minimal units of speech sound in a language that can serve to distinguish one word from another
 - The term phone is used to denote a phoneme's acoustic realization
 - E.g., English phoneme /t/ has two very different acoustic realizations in the word sat and meter
 - We have better treat them as two different phones when building a spoken language system

- Why Context Dependency
 - If we make the unit context dependent, we can significantly improve the recognition accuracy, provided there are enough training data for parameter estimation
 - A context usually refers to the immediately left and/or right neighboring phones
 - Context-dependent (CD) phonemes have been widely used for building LVCSR systems



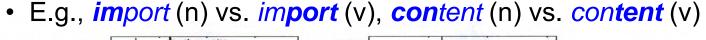
allophones: different realizations of a phoneme is called allophones →**Triphones** are examples of allophones

- Triphone (Intra-word triphone)
 - A triphone model is a phonetic model that takes into consideration both the left and right neighboring phones
 - It captures the most important coarticulatory effects
 - Two phones having the same identity but different left and right context are considered different triphones
 - Challenging issue: Need to balance trainability and accuracy with a number of parameter-sharing techniques



- Modeling inter-word context-dependent phone (like triphones) is complicated
 - Although the juncture effect on word boundaries is one of the most serious coarticulation phenomena in continuous speech recognition
 - E.g., speech /s p iy ch/→ /s/ and /ch/ are depending on the preceding and following words in actual sentences
 - Should be taken into consideration when developing the decoding/search algorithms
- Even with the same left/right context, a phone may have significant different realizations at different word positions
 - E.g., that rock $/t \rightarrow$ extinct!, theatrical $/t \rightarrow /ch/$

- Stress information for context dependency
 - Word-level stress (free stress)
 - The stress information: longer duration, higher pitch and more intensity for stressed vowels



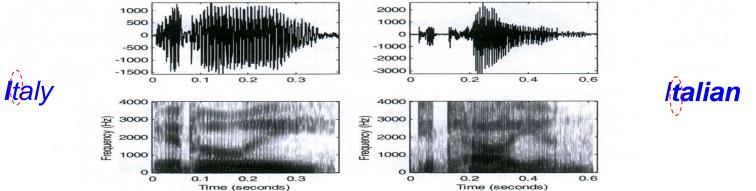


Figure 9.3 The importance of stress is illustrated in *Italy* vs. *Italian* for phone *It*. The realizations are quite different, even though they share the same left and right context.

- Sentence-level stress (including contrastive and emphatic stress)
 - Sentence-level stress is very hard to model without incorporate semantic and pragmatic knowledge
 - Contrastive: e.g., "I said import records not export"
 - Emphatic: e.g., "I did have dinner"

Clustered Acoustic-Phonetic Units

- Triphone modeling assumes that every triphone context is different. Actually, many phones have similar effects on the neighboring phones
 - /b/ and /p/ (labial stops) (or, /r/ and /w/ (liquids)) have similar effects on the following vowel
- It is desirable to find instances of similar contexts and merge them
 - A much more manageable number of models that can be better trained

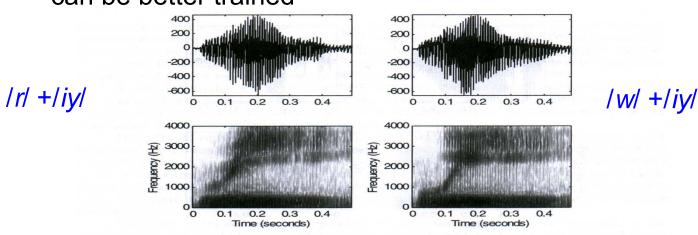
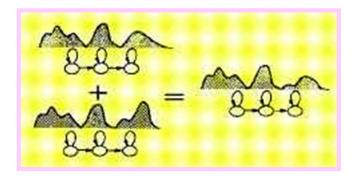
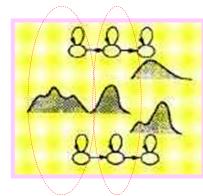


Figure 9.4 The spectrograms for the phoneme /iy/ with two different *left-contexts* are illustrated. Note that /r/ and /w/ have similar effects on /iy/. This illustrates that different left-contexts may have similar effects on a phone.

Model-based clustering



- State-based clustering (state-tying)
 - Keep the dissimilar states of two models apart while the other corresponding states are merged



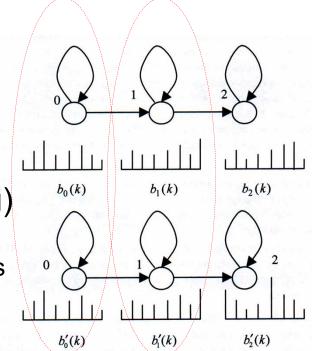
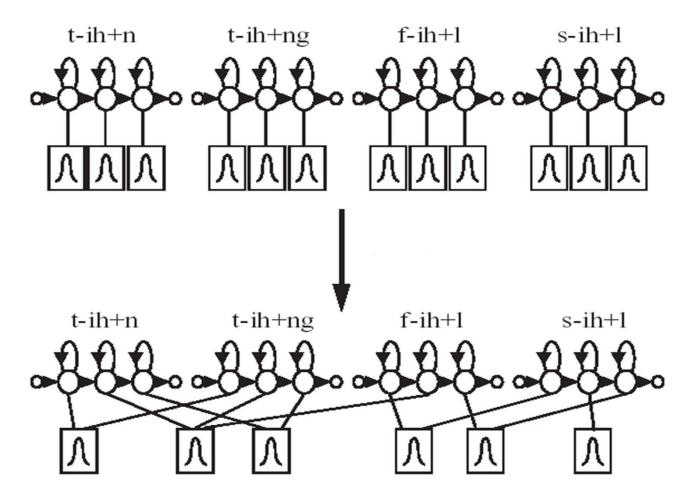


Figure 9.5 State-based vs. model-based clustering. These two models are very similar, as both the first and the second output distributions are almost identical. The key difference is the output distribution of the third state. If we measure the overall model similarity, which is often based on the accumulative output distribution similarities of all states, these two models may be clustered, leading to a very inaccurate distribution for the last state. If we cluster output distributions at state level, we can cluster the first two output distributions while leaving the last ones intact, leading to more accurate representations.

• State-tying of triphones



- Two key issues for CD phonetic or subphonetic modeling
 - Tying the phones with similar contexts to improve trainability and efficiency
 - Enable better parameter sharing and smoothing
 - Mapping the unseen triphones (in the test) into appropriately trained triphones is important
 - Because the possible of triphones could be very large
 - E.g., English has over 100,000 triphones

- Microsoft's approach State-based clustering
 - Generate clustering to the state-dependent output distributions across different phonetic models
 - Each cluster represents a set of similar HMM states and is called senone
 - A subword model is composed of a sequence of senones

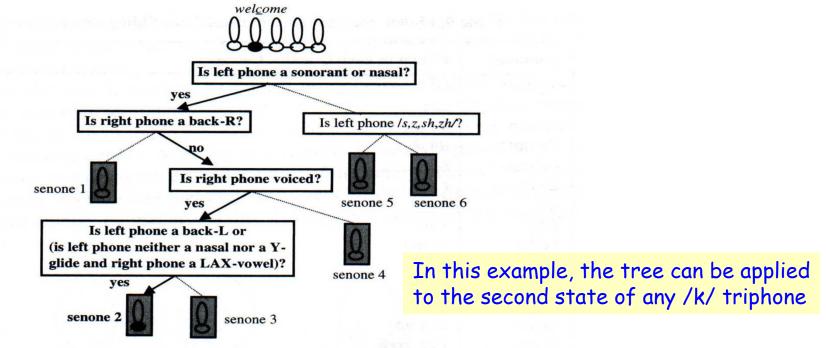


Figure 9.6 A decision tree for classifying the second state of *K*-triphone HMMs [48].

• Some example questions used in building senone trees

Questions	Phones in Each Question Category	
Aspseg	hh	1
Sil	sil	gh
Alvstp	dt	
Dental	dh th •iy yuw	• uw
Labstp	bp	station portaging the second statistical
Liquid	lr	[5] M. Line and A. Mitchell, 1997, 2018.
Lw	l w • ih	section and the section of the
S/Sh	s sh	• uh
Sylbic	er axr	
Velstp	g k ey	ax ba
Affric	ch jh	1 Jow
Lggl-B	front oy	-H
Nasal	mnng eh ay m	• ao
Retro	r er axr	aw aw
Schwa	ax ix axr	
Velar	ng g k • ae	× • aa
Fric2	th s sh f	• aa
Fric3	dh z zh v	ow
Lggl	lrwy	ow
S/Z/Sh/Zh	s z sh zh	0
Wglide	uw aw ow w	
Labial	wmbpv	
Palatl	y ch jh sh zh	
Yglide	iy ay ey oy y	1-1-1-1-1-1-1-1-1-1-1-1-1-1-1-1-1-1-1-
High	ih ix iy uh uw y	
Lax	eh ih ix uh ah ax	
Low	ae aa ao aw ay oy	
Orstp2	ptk	
Orstp3	b d g	
Alvelr	ndtsz	
Diph	uw aw ay ey iy ow oy	
Fric1	dh th s sh z zh v f	
Round	uh ao uw ow oy w axr er	
Frnt-R	ae eh ih ix iy ey ah ax y aw	
Tense	iy ey ae uw ow aa ao ay oy aw	
Back-L	uh ao uw ow aa er axr l r w aw	
Frnt-L	ae eh ih ix iy ey ah ax y oy ay	
Back-R	uh ao uw ow aa er axr oy l r w ay	
Orstp1	b d g p t k ch jh	6
Vowel	ae eh ih ix iy uh ah ax aa ao uw aw ay ey ow oy er axr	
Son	ae eh ih ix iy ey ah ax oy ay uh ao uw ow aa er axr aw l r w y	
Voiced	ae eh ih ix iy uh ah ax aa ao uw aw ay ey ow oy l r w y er axr m	
. Steed	n ng jh b d dh g v z zh	SP - Berlin Chen

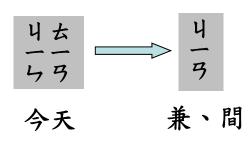
Comparison of recognition performance for different acoustic modeling

Table 9.4 Relative error reductions for different modeling units. The reduction is relative to that of the preceding row.

Units	Relative Error Reductions	181
Context-independent phone	Baseline	
Context-dependent phone	+25%	
Clustered triphone	+15%	model-based clusterin
Senone	+24%	state-based clusterin

Pronunciation Variation

- We need to provide alternative pronunciations for words that may have very different pronunciations
 - In continuous speech recognition, we must handle the modification of interword pronunciations and reduced sounds
- Variation kinds
 - Co-articulation (Assimilation)
 - "did you" /d ih jh y ah/, "set you" /s eh ch er/
 - Assimilation: a change in a segment to make it more like a neighboring segment
 - Deletion
 - /t/ and /d/ are often deleted before a consonant
- Variation can be drawn between
 - Inter-speaker variation (social)
 - Intra-speaker variation (stylistic)



Pronunciation Variation (cont.)

• Pronunciation Network (a probabilistic finite state machine)

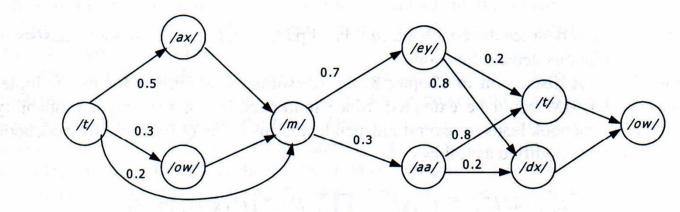


Figure 9.7 A possible pronunciation network for word *tomato*. The vowel |ey| is more likely to flap, thereby having a higher transition probability into |dx|.

- Examples:
 - E. g., word "*that*" appears 328 times in one corpus, with 117 different tokens of the 328 times (only 11% of the tokens are most frequent)

Greenberg, 1998

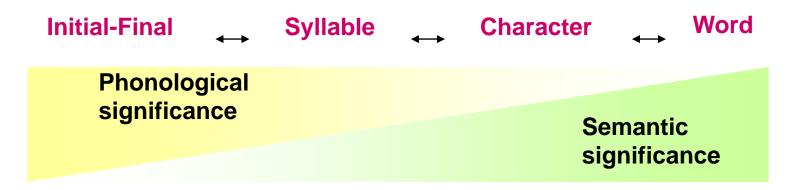
Cheating experiments show big performance improvements achieved if the tuned pronunciations were applied to those in test data (e.g. Switchboard WER goes from 40% to 8%)

Pronunciation Variation (cont.)

- Adaptation of Pronunciations
 - Dialect-specific pronunciations
 - Native vs. non-native pronunciations
 - Rate-specific pronunciations
- Side Effect
 - Adding more and more variants to the pronunciation lexicon increases size and confusion of the vocabulary
 - Lead to increased ASR WER

Characteristics of Mandarin Chinese

Four levels of linguistic units



- A monosyllabic-structure language
 - All characters are monosyllabic
- Most characters are morphemes (詞素)
- A word is composed of one to several characters
- Homophones
 - Different characters sharing the same syllable

Characteristics of Mandarin Chinese (cont.)

- Sub-syllable HMM Modeling
 - INITIALs

Context -Independent INITIALs b(ク), p(タ), m(ロ), f(こ), d(カ), t(よ), n(ろ), l(カ), g(巜), k(丂), h(厂), ji(ㄐ), chi(ㄑ), shi(丅), j(屮), ch(犭), sh(ㄕ), r(☑), tz(ℙ), ts(歬), s (᠘)

Table 2.2.1a 21 context-independent INITIALs.

Sub-Syllable HMM Modeling (cont.)

Sub-syllable HMM Modeling

– FINALs

	n a in a in the term
Cluster	FINALS
1.(empty)	empty
2.(a)	a(Y), ai(另), au(么), an(弓), ang(九)
3.(0)	o(こ), ou(又)
4.(e)	e(さ), en(ケ), eng(ム), er(ル)
5.(i)	i(一), ia(一Y), ie(一せ), iai(一刃),
[iau(-4), $ian(-3)$, $in(-4)$, $ing(-4)$,
ch_em	iang(ーナ), iou(ーヌ) <mark>, io</mark> (ーこ, e.g., for 南 was ignored here)
6.(u)	u(メ), ua(メY), uo(メT), uai(メ所),
iz emp	uei(X 1), uan(X 3), uen(X 4),
ts_emp	ueng(メム), uang(メた)
7.(iu) able	iu(ロ), iue(ロせ), iuan(ロラ), iun(ロム),
	iung(니ム)
8.(E)	$ei(\sim)$

Table 2.2.1b37 FINALs, which can be divided into 8clusters according to the beginning phone.

Classification and Regression Trees (CART)

- CART are binary decision trees, with splitting questions attached to each node
 - Act like a rule-based system where the classification carried out by a sequence of decision rules
- CART provides an easy representation that interprets and predicates the structure of a set of data
 - Handle data with high dimensionality, mixed data type and nonstandard data structure
 e.g., both numeric & symbolic
- CART also provides an automatic and data-driven framework to construct the decision process based on objective criteria, not subjective criteria
 - E.g., the choice and order of rules
- CART is a kind of clustering/classification algorithms

Classification and Regression Trees (cont.)

- Example: height classification
 - Assign a person to one of the following five height classes

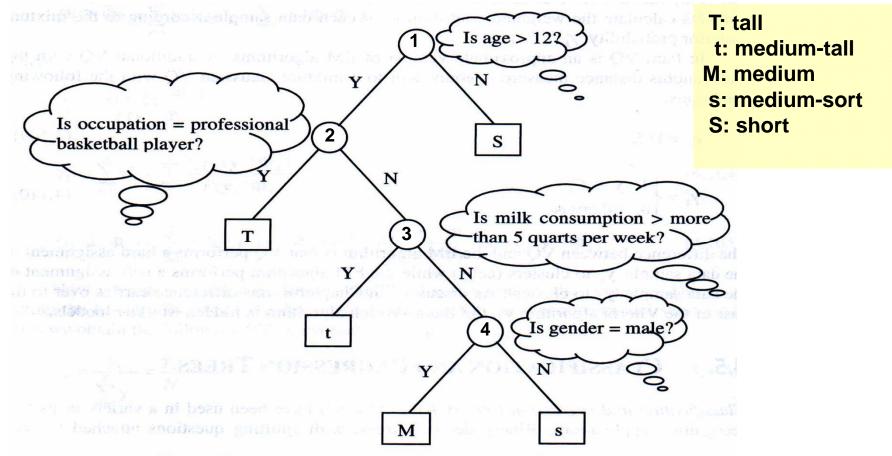


Figure 4.14 A binary tree structure for height classification.

Classification and Regression Trees (cont.)

- Example: height classification (cont.)
 - Can easy predict the height class for any new person with all the measured data (age, occupation, milk-drinking, etc.) but no height information, by traversing the binary tree (based on a set of questions)
 - "No": right branch, "Yes" left branch
 - When reaching a leaf node, we can use its attached label as the height class for the new person
 - Also can use the average height in the leaf node to predict the height of the new person

• Steps

- 1. First, find a set of questions regarding the measured variable
 - E.g., "Is age>12?", "Is gender=male?", etc.
- 2. Then, place all the training samples in the root of the initial tree
- 3. Choose the best question from the question set to split the root into two nodes (need some measurement !)
- 4. Recursively split the most promising node with the best question until the right-sized tree is obtained

How to choose the best question?

- E.g., reduce the uncertainty of the event being decided upon i.e., find the question which gives the **greatest entropy reduction**

- Splitting Criteria (for discrete pf)
 - How to find the best question for a node split ?
 - I.e., find the best split for the data samples of the node
 - Assume training samples have a probability (mass function) $P(\omega|t)$ at each node *t* (modeled as a discrete random variable)
 - E.g., $P(\omega_i | t)$ is the percentage of data samples for class *i* at a node *t* and $\sum_i P(\omega_i | t) = 1$

Review: Fundamentals in Information Theory

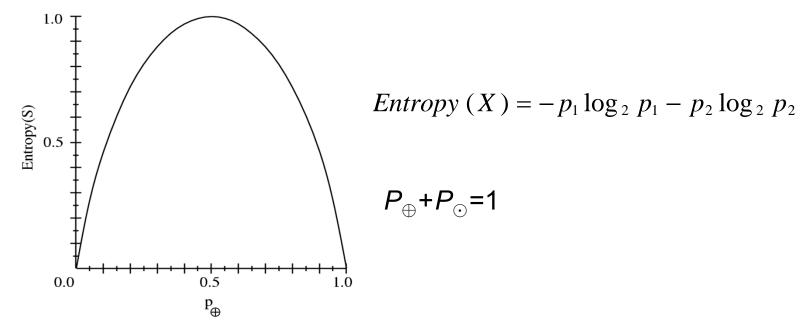
- Three interpretations for quantity of information
 1. The amount of uncertainty before seeing an event
 2. The amount of surprise when seeing an event
 3. The amount of information after seeing an event
- The definition of information: $I(x_i) = \log \frac{1}{P(x_i)} = -\log P(x_i)$

define
$$0\log_2 0 = 0$$

- $P(x_i)$ the probability of an event x_i
- Entropy: the average amount of information $H(X) = E[I(X)]_X = E[-\log P(x_i)]_X = \sum_{x_i} - P(x_i) \cdot \log P(x_i)$ where $S = \{x_1, x_2, ..., x_i, ...\}$
 - Have maximum value when the probability (mass) function is a uniform distribution

Review: Entropy

• For Boolean classification (0 or 1)



- Entropy can be expressed as the minimum number of bits of information needed to encode the classification of an arbitrary number of examples
 - If *c* classes are generated, the maximum of Entropy can be $Entropy(X) = \log_2 c$

- Splitting Criteria (for discrete pf)
 - Define the weighted entropy for any tree node *t*

 $\overline{H}_{t}(Y) = H_{t}(Y)P(t)$ $H_{t}(Y) = -\sum_{i} P(\omega_{i}|t)\log P(\omega_{i}|t), \text{ Entropy: average amount of information}$

- Y is the discrete random variable for classification decision
- P(t) is the prior probability of visiting node t (ratio of numbers of samples in a node t and the total number of samples)

- Splitting Criteria (for discrete pf)
 - Entropy reduction for a question q to split a node t into nodes I and r
 - Pick the question with the greatest entropy reduction

$$\Delta \overline{H}_{t}(q) = \overline{H}_{t}(Y) - \left(\overline{H}_{l}(Y) + \overline{H}_{r}(Y)\right) = \overline{H}_{t}(Y) - \overline{H}_{t}(Y|q)$$

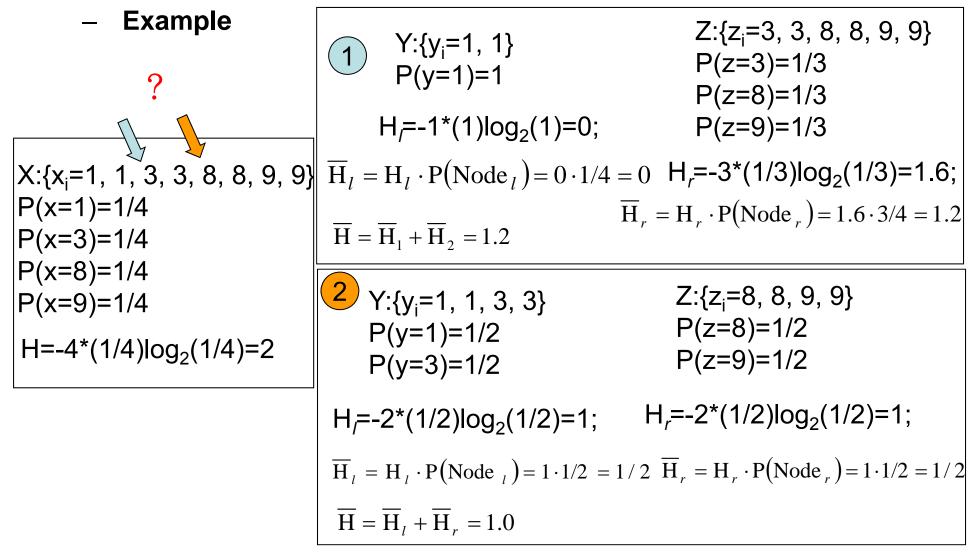
$$q^{*} = \arg \max_{q} \left[\Delta \overline{H}_{t}(q)\right]$$

$$\overline{H}_{t}(Y) = \left[-\sum_{i} P(\omega_{i}|t) \log P(\omega_{i}|t)\right] P(t)$$

$$\overline{H}_{l}(Y) = \left[-\sum_{i} P(\omega_{i}|t) \log P(\omega_{i}|t)\right] P(t)$$

$$\overline{H}_{r}(Y) = \left[-\sum_{i} P(\omega_{i}|t) \log P(\omega_{i}|t)\right] P(t)$$

• Splitting Criteria (for discrete distribution)



- Entropy for a tree
 - the sum of weighted entropies for all terminal nodes

$$\overline{H}(T) = \sum_{t \text{ is terminal}} w_t \cdot \overline{H}_t(Y_t)$$

- It can be show that the above tree-growing (splitting) procedure repeatedly reduces the entropy of the tree
- The resulting tree has a better classification power

- Splitting Criteria (for continuous probability functions)
 - The likelihood gain is often used instead of the entropy measure
 - Suppose one split divides the data X into two groups X_1 and X_2 , which can be respectively represented as two Gaussian distributions $N_1(\mu_1, \Sigma_1)$ and $N_2(\mu_2, \Sigma_2)$

$$L_{1}(\boldsymbol{X}_{1}|\boldsymbol{N}_{1}) = \log \prod_{\boldsymbol{x}_{1}} N(\boldsymbol{x}_{1};\boldsymbol{\mu}_{1},\boldsymbol{\Sigma}_{1})$$

$$L_{2}(\boldsymbol{X}_{2}|\boldsymbol{N}_{2}) = \log \prod_{\boldsymbol{x}_{2}} N(\boldsymbol{x}_{2};\boldsymbol{\mu}_{2},\boldsymbol{\Sigma}_{2}) \quad \boldsymbol{X}_{1}$$

Log likelihood gain at node t $\Delta \overline{L}_t(q) = L_1(\mathbf{X}_1|N_1) + L_2(\mathbf{X}_2|N_2) - L_\mathbf{X}(\mathbf{X}|N)$

 $L_1(\mathbf{X}_1|N_1) = -(d\log 2\pi + |\Sigma_1| + d) \cdot a/2$

$$= (a + b) \log |\Sigma| - a \log |\Sigma_1| - b \log |\Sigma_2| \quad (\ge 0?)$$

a, b are the sample counts for X_1 and X_2