13.0 Speaker Variabilities: Adaption and Recognition

References: 1. 9.6 of Huang

- 2. "Maximum A Posteriori Estimation for Multivariate Gaussian Mixture Observations of Markov Chains", IEEE Trans. on
- Maximum Likelihood Linear Regression for Speaker Adaptation of Continuous Density Hidden Markov Models", Computer Speech and Language, Vol.9 ,1995
- 4. Jolliffe, "Principal Component Analysis", Springer-Verlag, 1986
- 5. "Rapid Speaker Adaptation in Eigenvoice Space", IEEE Trans. on Speech and Audio Processing, Nov 2000
- 6. "Cluster Adaptive Training of Hidden Markov Models", IEEE Trans. on Speech and Audio Processing, July 2000
- 7. "A Compact Model for Speaker-adaptive Training", International Conference on Spoken Language Processing, 1996
- 8. "A Tutorial on Text-independent Speaker Verification", EURASIP Journal on Applied Signal Processing 2004
- 9. "An Overview of Text-independent Speaker Recognition: from Features to Supervectors", Speech Communication, Jan 2010
- 10. "VoxCeleb2: Deep Speaker Recognition", Interspeech 2018

Speaker Dependent/Independent/Adaptation

• Speaker Dependent (SD)

- trained with and used for 1 speaker only, requiring huge quantity of training data, best accuracy
- practically infeasible
- Multi-speaker
 - trained for a (small) group of speakers

• Speaker Independent (SI)

- trained from large number of speakers, each speaker with limited quantity of data
- good for all speakers, but with relatively lower accuracy

• Speaker Adaptation (SA)

- started with speaker independent models, adapted to a specific user with limited quantity of data (adaptation data)
- technically achievable and practically feasible

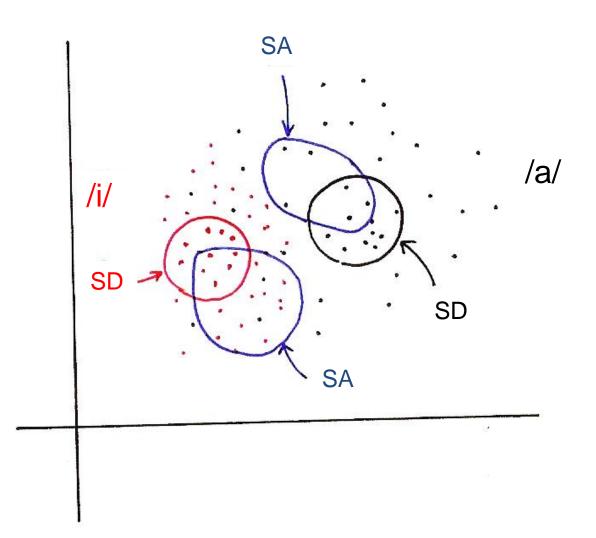
Supervised/Unsupervised Adaptation

- supervised: text (transcription) of the adaptation data is known
- unsupervised: text (transcription) of the adaptation data is unknown, based on recognition results with speaker-independent models, may be performed iteratively

Batch/Incremental/On-line Adaptation

- batch: based on a whole set of adaptation data
- incremental/on-line: adapted step-by-step with iterative re-estimation of models e.g. first adapted based on first 3 utterances, then adapted based on next 3 utterances or first 6 utterances,...

Speaker Dependent/Independent/Adaptation



MAP (Maximum A Posteriori) Adaptation

• Given Speaker-independent Model set $\Lambda = \{\lambda_i = (A_i, B_i, \pi_i), i=1, 2,...M\}$ and A set of Adaptation Data $\overline{O} = (o_1, o_2,...o_t,...o_T)$ for A Specific Speaker

$$\Lambda^{*} = {}^{\operatorname{arg\,max}}_{\Lambda} \operatorname{Prob}[\Lambda | \overline{O}] = {}^{\operatorname{arg\,max}}_{\Lambda} \frac{\operatorname{Prob}[\overline{O} | \Lambda] \operatorname{Prob}[\Lambda]}{\operatorname{Prob}[\overline{O}]} = {}^{\operatorname{arg\,max}}_{\Lambda} \operatorname{Prob}[\overline{O} | \Lambda] \operatorname{Prob}[\Lambda]$$

- With Some Assumptions on the Prior Knowledge Prob [Λ] and some Derivation (EM Theory)
 - example adaptation formula

$$\mu_{jk}^{*} = \frac{\tau_{jk}\mu_{jk} + \sum_{t=1}^{T} [\gamma_{t}(j,k)o_{t}]}{-\tau_{t}}$$

$$\tau_{jk} + \Sigma_{t=1}^T \gamma_t(j,k)$$

 μ_{jk} : mean of the k - th Gaussian in the j - th state for a certain λ_i

 μ_{jk}^* : adapted value of μ_{jk}

$$\gamma_t(j,k) = \left[\frac{\alpha_t(j)\beta_t(j)}{\sum_{j=1}^N \alpha_t(j)\beta_t(j)}\right] \left[\frac{c_{jk}N(o_t;\mu_{jk},U_{jk})}{\sum_{m=1}^L c_{jm}N(o_t;\mu_{jm},U_{jm})}\right]$$

$$\gamma_t(j) = P(q_t = j | \overline{O}, \lambda_i)$$

 τ $_{\rm jk}$: a parameter having to do with the prior knowledge about $\,\mu_{\,\rm jk}$

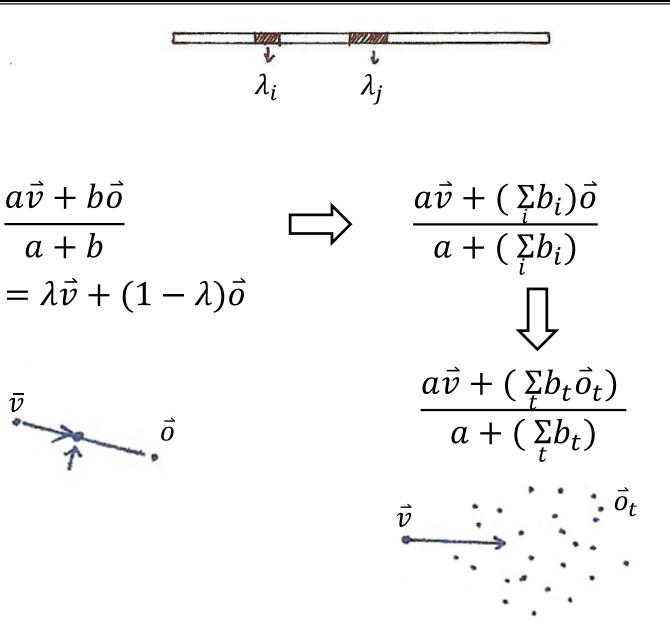
may have to do with number of samples used to train μ_{jk}

- a weighted sum shifting μ_{jk} towards those directions of o_t (in j-th state and k-th Gaussian) larger τ_{ik} implies less shift

• Only Those Models with Adaptation Data will be Modified, Unseen Models remain Unchanged — MAP Principle

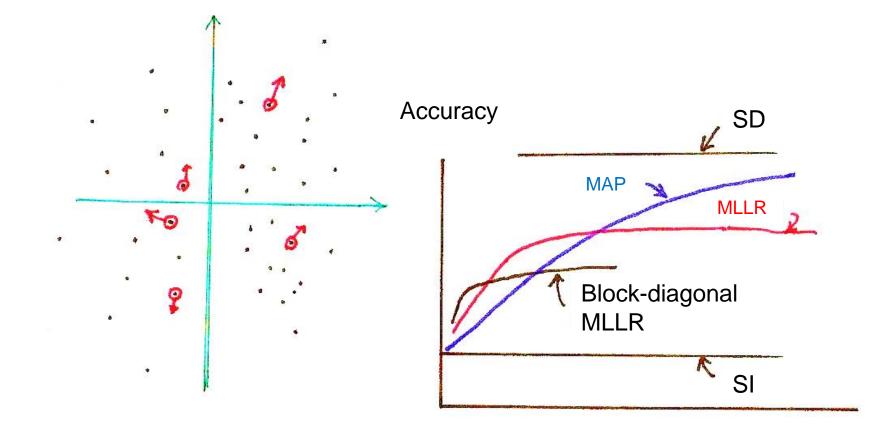
- good with larger quantity of adaptation data
- poor performance with limited quantity of adaptation data

MAP Adaptation



5

MAP Adaptation



Adaptation Data

Maximum Likelihood Linear Regression (MLLR)

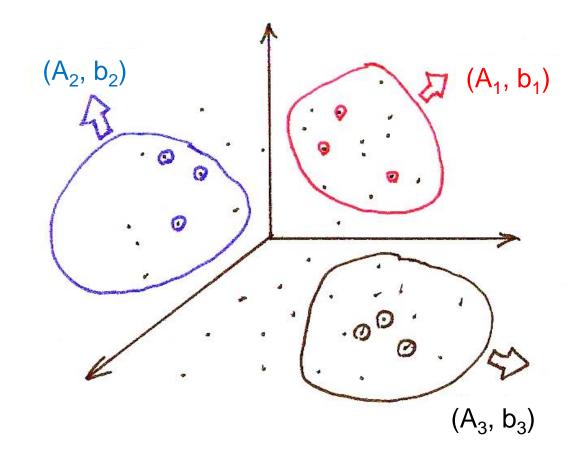
• Divide the Gaussians (or Models) into Classes C₁, C₂,...C_L, and Define Transformation-based Adaptation for each Class

 $\mu_{jk}^* = A \mu_{jk} + b$, μ_{jk} : mean of the k - th Gaussian in the j - th state

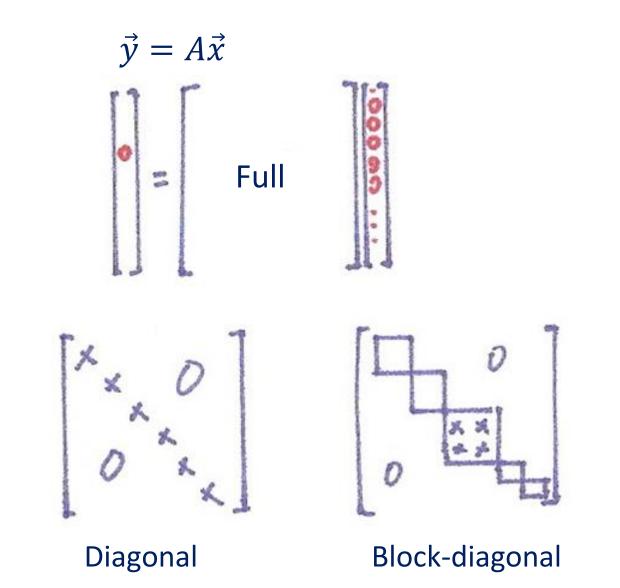
- linear regression with parameters A, b estimated by maximum likelihood criterion

 $[A_i, b_i] = \stackrel{\text{arg max}}{A, b} \operatorname{Prob}[\overline{O} | \Lambda, A_i, b_i] \text{ for a class } C_i$

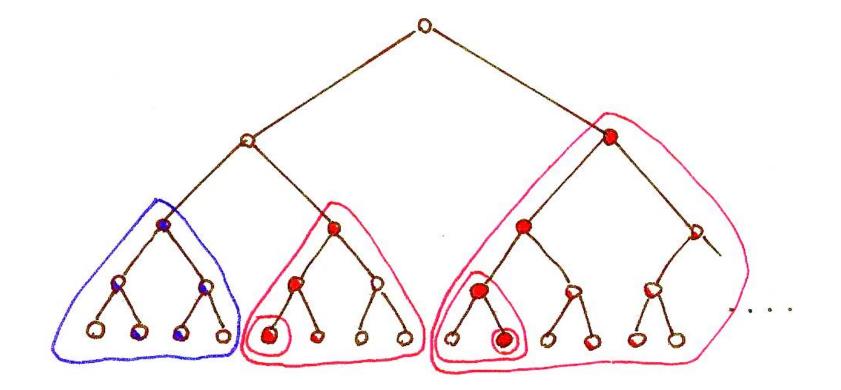
- A_i, b_i estimated by EM algorithm
- All Gaussians in the same class updated with the same A_i , b_i : parameter sharing, adaptation data sharing
- unseen Gaussians (or models) can be adapted as well
- A_i can be full matrices, or reduced to diagonal or block-diagonal to have less parameters to be estimated
- faster adaptation with much less adaptation data needed, but saturated at lower accuracy with more adaptation data due to the less precise modeling
- Clustering the Gaussians (or Models) into L Classes
 - too many classes requires more adaptation data, too less classes becomes less accurate
 - basic principle: Gaussian (or models) with similar properties and "just enough" data form a class
 - data-driven (e.g. by Gaussian distances) primarily, knowledge driven helpful
- Tree-structured Classes
 - the node including minimum number of Gaussians (or models) but with adequate adaptation data is a class
 - dynamically adjusting the classes as more adaptation data are observed
- Feature-based MLLR (fMLLR)



MLLR



MLLR



Principal Component Analysis (PCA)

• Problem Definition:

- for a zero mean random vector *x* with dimensionality *N*, *x*∈R^N, E(*x*)=0, iteratively find a set of *k* (*k*≤*N*) orthonormal basis vectors {*e*₁, *e*₂,..., *e*_k} so that (1) var (*e*₁^T*x*)=max (*x* has maximum variance when projected on *e*₁)
 (2) var (*e*₁^T*x*)=max, subject to *e*₁⊥*e*₁, 1⊥....⊥*e*₁, 2≤ *i* ≤*k* (*x* has next maximum variance when projected on *e*₂, etc.)
- Solution: $\{e_1, e_2, ..., e_k\}$ are the eigenvectors of the covariance matrix Σ for x corresponding to the largest k eigenvalues
 - new random vector $y \in \mathbb{R}^k$: the projection of x onto the subspace spanned by $A = [e_1 \ e_2 \ \dots \ e_k], y = A^T x$
 - a subspace with dimensionality k≤N such that when projected onto this subspace, y is "closest" to x in terms of its "randomness" for a given k
 var (e_i^T x) is the eigenvalue associated with e_i
- Proof
 - -var $(e_1^T x) = e_1^T E(x x^T)e_1 = e_1^T \Sigma e_1 = \max$, subject to $|e_1|^2 = 1$ -using Lagrange multiplier

$$J(e_1) = e_1^T E(x x^T)e_1 - \lambda(|e_1|^{2-1}), \quad \frac{\partial J(e_1)}{\partial e_1} = 0$$

$$\Rightarrow E(xx^T) e_1 = \lambda_1 e_1, \quad var(e_1^T x) = \lambda_1 = \max$$

$$- \text{ similar for } e_2 \text{ with an extra constraint } e_2^T e_1 = 0, \text{ etc.}$$

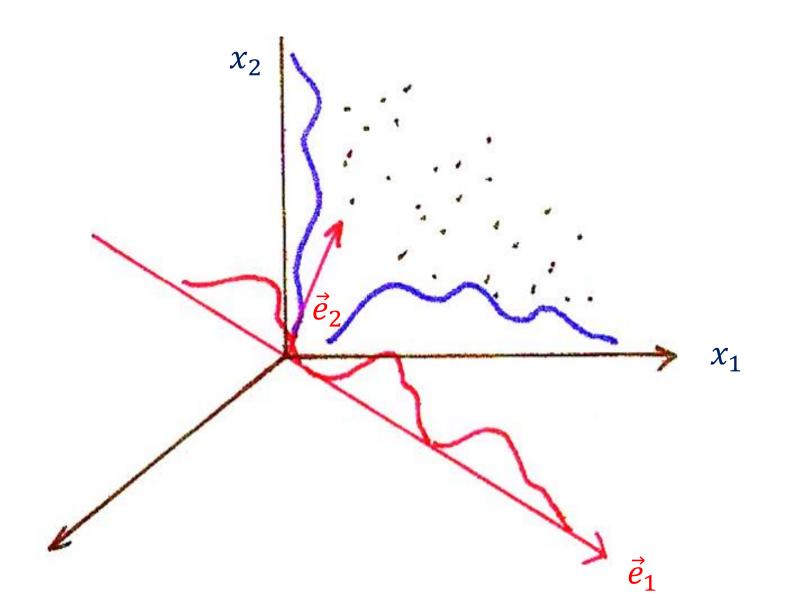
$$Av = u$$

$$Av = \lambda v$$

$$f \land \delta$$

$$eigenvector eigenvalue$$

PCA

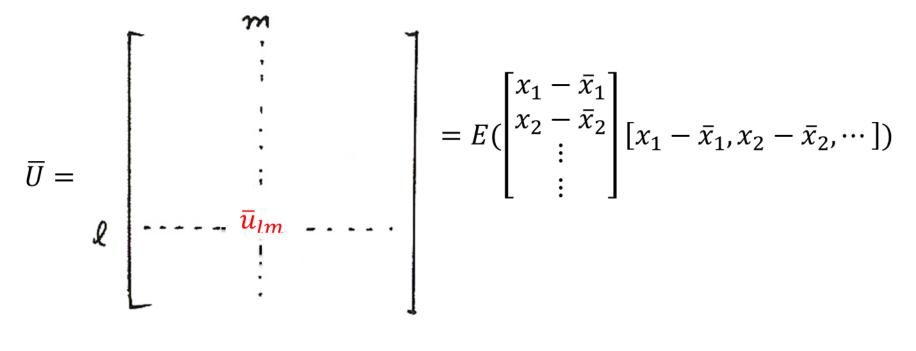


PCA

$$\cdots e_{1k}^{T} \cdots \left[\begin{matrix} \vdots \\ x_{k} \\ \vdots \end{matrix} \right] = \vec{e}_{1} \cdot \vec{x}$$

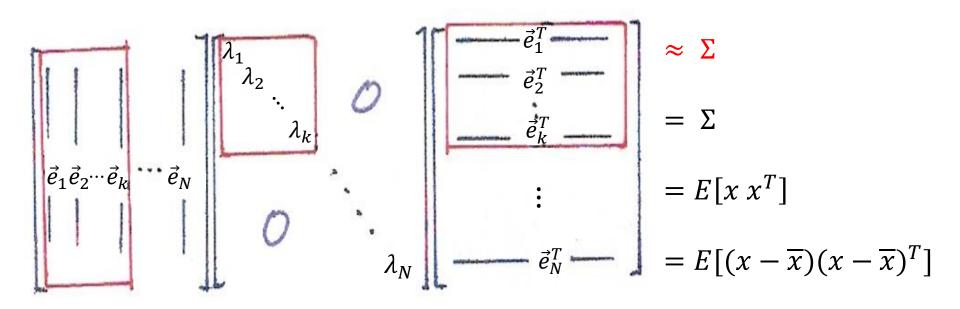
$$= \frac{|\vec{e}_{1}||\vec{x}| \cos \theta}{\left\| \begin{matrix} \\ \\ \\ \\ 1 \end{matrix} \right|}$$

$$\vec{y} = A^{T} \vec{x} = \begin{bmatrix} \vec{e}_{1}^{T} \\ \vec{e}_{2}^{T} \\ \vdots \\ \vec{e}_{k}^{T} \end{bmatrix} \vec{x}$$



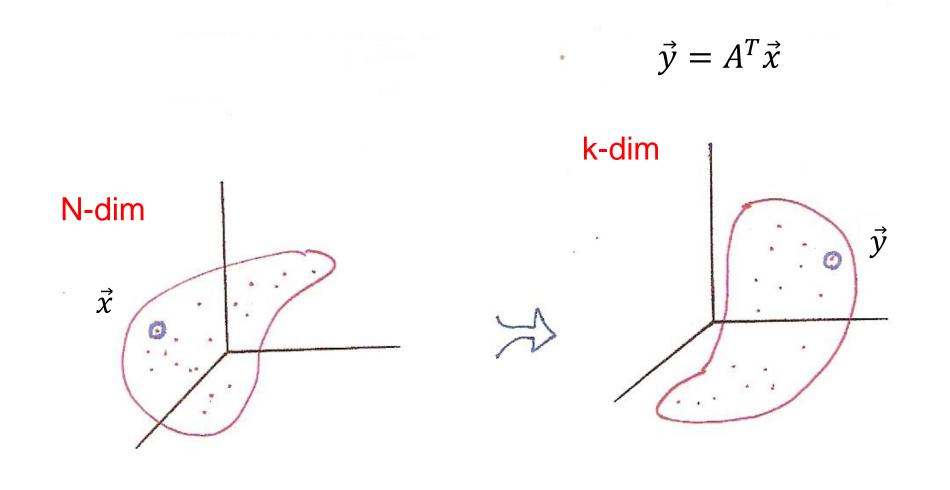
 $\bar{u}_{lm} = E[(x_l - \bar{x}_l)(x_m - \bar{x}_m)]$

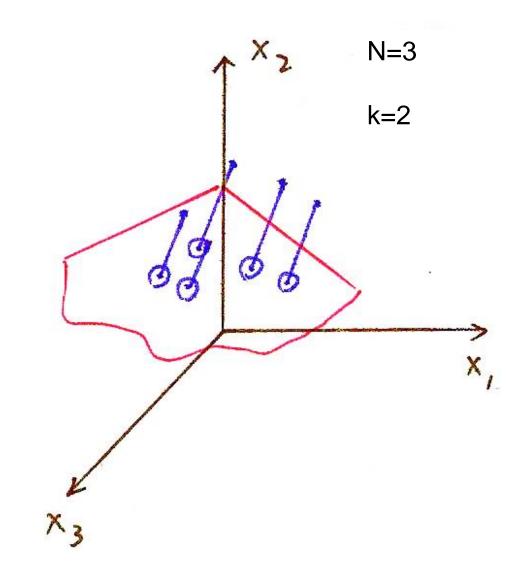
PCA



$$\sum_{\substack{\underline{\vec{e}_i} \\ \uparrow}} = \frac{\lambda_i}{\uparrow} \frac{\vec{e}_i}{\stackrel{eigenvalue}{\longrightarrow}}$$

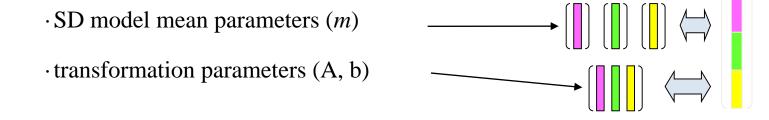
Av = u $Av = \lambda v$ $\uparrow \quad \uparrow$ eigenvector eigenvalue





Eigenvoice

- A Supervector x constructed by concatenating all relevant parameters for the speaker specific model of a training speaker
 - concatenating the mean vectors of Gaussians in the speaker-dependent phone models
 - -concatenating the columns of A, b in MLLR approach
 - x has dimensionality N (N = $5,000 \times 3 \times 8 \times 40 = 4,800,000$ for example)



- A total of L (L = 1,000 for example) training speakers gives L supervectors x₁,x₂,...x_L
 - $-x_1, x_2, x_3, \dots, x_L$ are samples of the random vector x
 - -each training speaker is a point (or vector) in the space of dimensionality N
- Principal Component Analysis (PCA)
 - -x'=x-E(x), $\Sigma = E(x' x'^T)$,

 $\Sigma \approx [e_1, e_2, \dots, e_K] [\lambda_i] [e_1, e_2, \dots, e_k]^T$, $[\lambda_i]$: diagonal with λ_i as elements

 $\{e_1, e_2, \dots, e_k\}$: eigenvectors with maximum eigenvalues $\lambda_1 > \lambda_2 \dots > \lambda_k$ k is chosen such that λ_j , j>k is small enough (k=250 or 50 for example)

Eigenvoice

• Principal Component Analysis (PCA)

-x' = x - E(x), $\Sigma = E(x' x'^T)$, $\Sigma \approx [a, a, b] [\lambda, b] [a, a, c, b] T [\lambda, b] diagonal with$

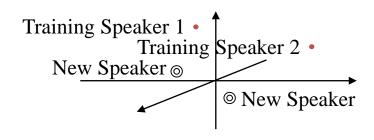
 $\Sigma \approx [e_1, e_2, \dots, e_K] [\lambda_i] [e_1, e_2, \dots, e_K]^T, [\lambda_i]: \text{diagonal with } \lambda_i \text{ as elements}$

 $\{e_1, e_2, \dots, e_k\}$: eigenvectors with maximum eigenvalues $\lambda_1 > \lambda_2 \dots > \lambda_k$

k is chosen such that λ_j , j>k is small enough (k=50 for example)

• Eigenvoice Space: spanned by $\{e_1, e_2, \dots, e_k\}$

- each point (or vector) in this space represents a whole set of tri-phone model parameters
- $-\{e_1, e_2, \dots, e_k\}$ represents the most important characteristics of speakers extracted from huge quantity of training data by large number of training speakers
- -each new speaker as a point (or vector) in this space, $y = \sum_{i=1}^{k} a_i e_i$
- a_i estimated by maximum likelihood principle (EM algorithm) $\overline{a}^* = \frac{\arg \max}{\overline{a}} \operatorname{Prob}[\overline{O} \Big| \sum_{i=1}^{k} a_i e_i]$



• Features and Limitations

- only a small number of parameters $a_1 \dots a_k$ is needed to specify the characteristics of a new speaker
- -rapid adaptation requiring only very limited quantity of training data
- -performance saturated at lower accuracy (because too few free parameters)
- -high computation/memory/training data requirements

Speaker Adaptive Training (SAT) and Cluster Adaptive Training (CAT)

• Speaker Adaptive Training (SAT)

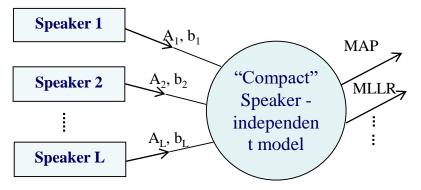
- trying to decompose the phonetic variation and speaker variation
- removing the speaker variation among training speakers as much as possible
- obtaining a "compact" speaker-independent model for further adaptation
- -y=Ax+b in MLLR can be used in removing the speaker variation

Clustering Adaptive Training (CAT)

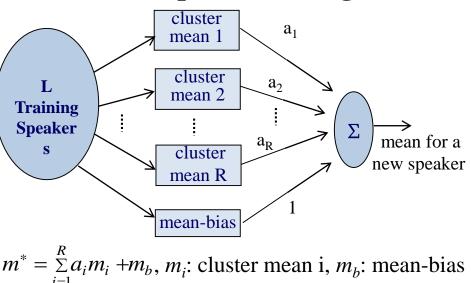
- dividing training speakers into R clusters by speaker clustering techniques
- obtaining mean models for all clusters(may include a mean-bias for the "compact" model in SAT)
- models for a new speaker is interpolated from the mean vectors

• Speaker Adaptive Training (SAT)

Training Speakers



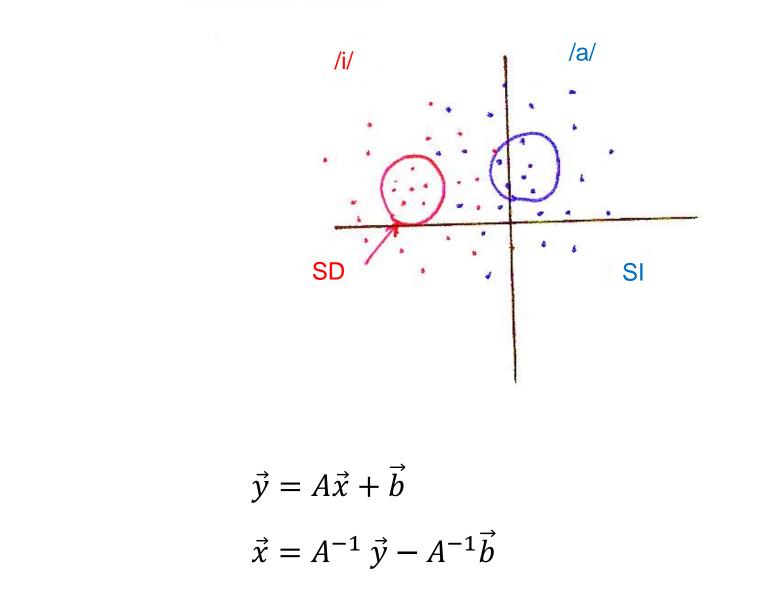
Original SI : $\Lambda^* = {}^{\arg \max}_{\Lambda} \operatorname{Prob}(\overline{o}_{1,2...L} | \Lambda)$ SAT : $[\Lambda_c^*, (A, b)_{1,\dots L}^*] = \underset{\Lambda_c, (A, b)_{1,\dots L}}{\operatorname{argmax}} \operatorname{Prob}(\overline{o}_{1,2\dots L} | \Lambda_c, (A, b)_{1,\dots L}) \quad m^* = \underset{i=1}{\overset{R}{\Sigma}} a_i m_i + m_b, m_i: \text{ cluster mean } i, m_b: \text{ mean-bias}$ EM algorithm used



 a_i estimated with maximum likelihood criterion

• Cluster Adaptive Training (CAT)

SAT



Speaker Recognition/Verification

• To recognize the speakers rather than the content of the speech

- phonetic variation/speaker variation
- speaker identification: to identify the speaker from a group of speakers
- speaker verification: to verify if the speaker is as claimed
- Gaussian Mixture Model (GMM)

$$\lambda_i = \{(w_j, \mu_j, \Sigma_j), j=1,2,...M\}$$
 for speaker i

for
$$\overline{\mathbf{O}} = \mathbf{O}_1 \mathbf{O}_2 \dots \mathbf{O}_t \dots \mathbf{O}_T$$
, $b_i(o_t) = \sum_{j=1}^M w_j N(o_t; \mu_j, \Sigma_j)$

- maximum likelihood principle

 $i^* = \underset{i}{\operatorname{arg max}} \operatorname{Prob}(\overline{O}|\lambda_i)$

Feature Parameters

- those carrying speaker characteristics preferred
- MFCC
- MLLR coefficients A_i,b_i, eigenvoice coefficients a_i, CAT coefficients a_i

Speaker Verification

- text dependent: higher accuracy but easily broken
- text independent
- likelihood ratio test

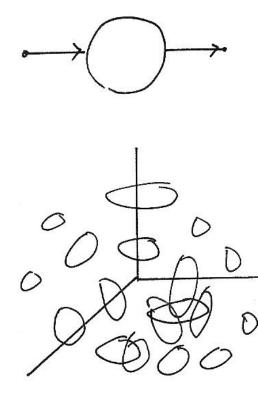
$$\rho(\overline{O}; \lambda_{i}) = \frac{p(O|\lambda_{i})}{p(\overline{O}|\overline{\lambda}_{i})} > th$$

- $\overline{\lambda}_i$: background model or anti model for speaker i, trained by
 - other speakers, competing speakers, or speaker independent model
- th: threshold adjusted by balancing missing/false alarm rates and ROC curre
- speech recognition based verification

Speaker Recognition

Gaussian Mixture Model (GMM)

HMM







Likelihood Ratio Test

- Detection Theory— Hypothesis Testing/Likelihood Ratio Test
 - 2 Hypotheses: H_0 , H_1 with prior probabilities: $P(H_0)$, $P(H_1)$ observation: X with probabilistic law: $P(X | H_0)$, $P(X | H_1)$
 - MAP principle choose H_0 if $P(H_0 | X) > P(H_1 | X)$ choose H_1 if $P(H_1 | X) > P(H_0 | X)$ $= \sum \frac{P(H_0|X)}{P(H_1|X)} \stackrel{\mathbf{n}_0}{\underset{\mathbf{u}}{\gtrsim}} 1$ $P(X|H_1)$ $P(X|H_0)$ X_2 \mathbf{X}_1

- Likelihood Ratio Test $P(H_i | X) = P(X | H_i) P(H_i) / P(X), i=0,1$ $\implies \frac{P(X | H_0)}{P(X | H_1)} \stackrel{H_0}{\underset{H_1}{\overset{\geq}{\underset{H_1}{\overset{\in}{\underset{H_1}{\overset{H_1}{\underset{H_1}{\overset{H_1}{\underset{H_1}{\overset{H_1}{\underset{H_1}{\overset{H_1}{\underset{H_1}{\underset{H_1}{\overset{H_1}{\underset{H_1}{\overset{H_1}{\underset{H_1}{\overset{H_1}{\underset{H_1}{\underset{H_1}{\overset{H_1}{\underset{H}$

Type I error: missing (false rejection)
Type II error: false alarm (false detection)
false alarm rate, false rejection rate, detection rate, recall rate, precision rate
Th: a threshold value adjusted by balancing among different performance rates ²⁴

Receiver Operating Characteristics (ROC) Curve

