## Average-Voice-Based Speech Synthesis Using HSMM-Based Speaker Adaptation and Adaptive Training

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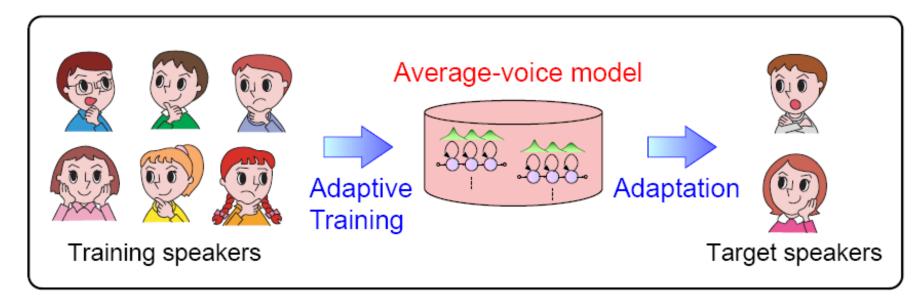
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# Speaker Adaptation

- Originally developed in ASR, but works very well in TTS
- Average voice-based speech synthesis (AVSS) [Yamagishi;'06]

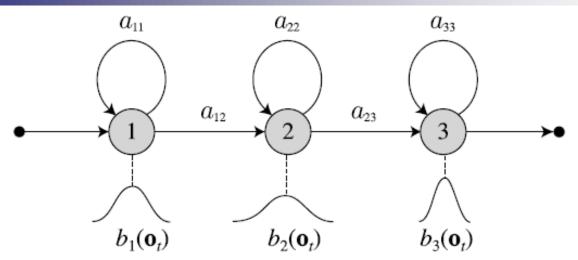


- Require small data of target speaker/speaking style
  - ⇒ Smal cost to create new voices



### Introduction

- In the previous work, the <u>HMM-based</u> speaker adaptation and speaker adaptive training were conducted for transforming and normalizing only state output probability distributions corresponding to <u>spectrum</u> and <u>F0</u> parameters of the speech data.
- The <u>HSMM</u> is an HMM with explicit <u>state-duration</u> probability distributions and enables us to conduct simultaneous adaptation of the output distributions and state duration distributions.



 $a_{ij}$ : Transition Probability  $b_i(\mathbf{o}_t)$ : Output Probability

Fig. 1 Hidden Markov model.

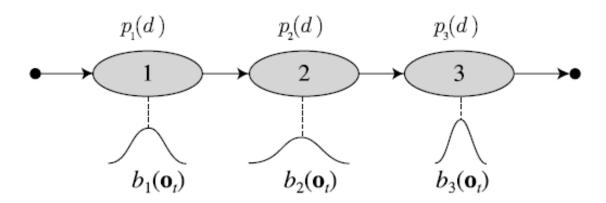
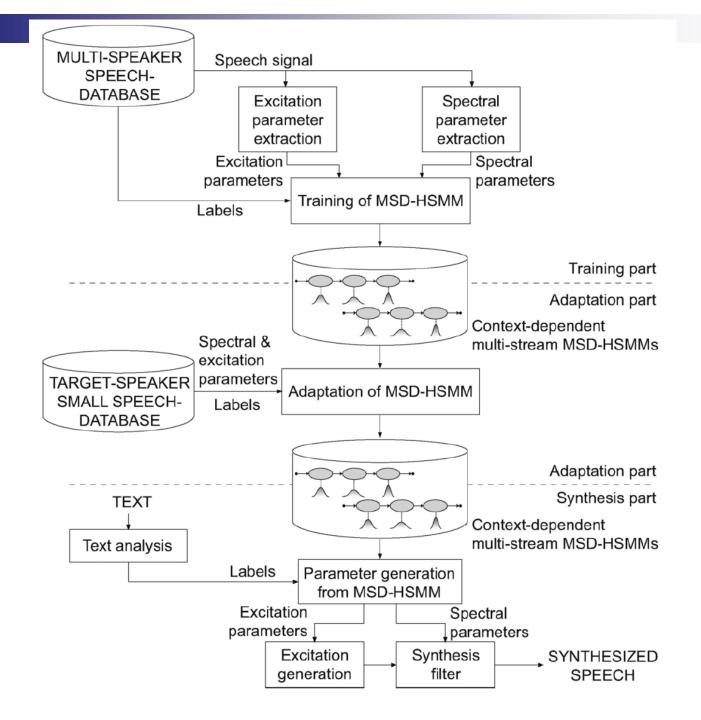


Fig. 2 Hidden semi-Markov model.

 $p_i(d)$ : Duration Probability  $b_i(\mathbf{o}_t)$ : Output Probability





## Hidden Semi-Markov Model

■ We assume that the i-th state output and duration distributions are Gaussian distributions characterized by a mean vector  $\boldsymbol{\mu}_i \in \mathcal{R}^L$  and diagonal covariance matrix  $\boldsymbol{\Sigma}_i \in \mathcal{R}^{L \times L}$ , and a scalar mean  $m_i$  and variance  $\sigma_i^2$ .

$$b_i(\mathbf{o}) = \mathcal{N}(\mathbf{o}; \boldsymbol{\mu}_i, \boldsymbol{\Sigma}_i)$$
$$p_i(d) = \mathcal{N}(d; m_i, \sigma_i^2)$$

• where  $o \in \mathcal{R}^{3L}$  is an observation vector and d is the duration in state i.

## м

### Hidden Semi-Markov Model

■ The observation probability of training data  $O = (o_1, \dots, o_T)$  of length T, given the model  $\lambda$ , can be written as

$$P(\mathbf{O}|\lambda) = \sum_{i=1}^{N} \sum_{\substack{j=1 \ j \neq i}}^{N} \sum_{d=1}^{t} \alpha_{t-d}(j) \ p_i(d) \prod_{s=t-d+1}^{t} b_i(\mathbf{o}_s) \ \beta_t(i)$$

• Where  $\forall t \in [1, T]$ .



### Hidden Semi-Markov Model

■ The state occupancy probability  $\gamma_t^d(i)$  of being in the state i at the period of time from t-d+1 to t is defined as

$$\gamma_t^d(i) = \frac{1}{P(\mathbf{O}|\lambda)} \sum_{\substack{j=1\\j \neq i}}^N \alpha_{t-d}(j) \ p_i(d) \prod_{s=t-d+1}^t b_i(\mathbf{o}_s) \ \beta_t(i)$$



■ In the HSMM-based MLLR adaptation, the mean vectors of the state output and duration distributions for the target speaker are obtained by linearly transforming the mean vectors of the state output and duration distributions of the average voice model (Fig. 3) as follows:

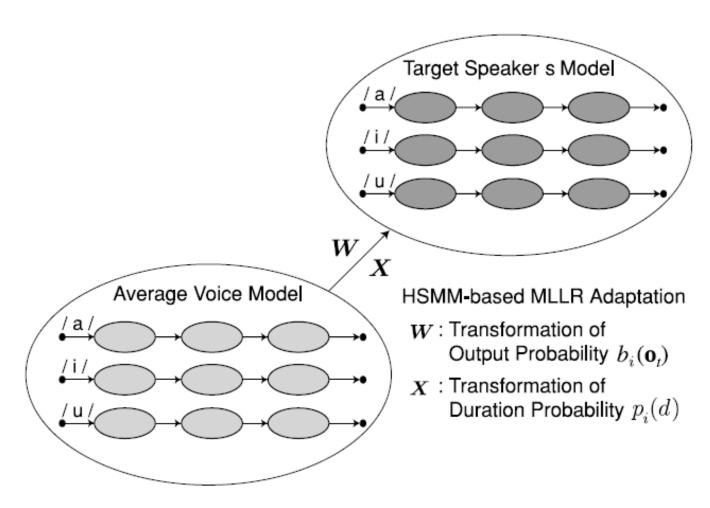


Fig. 3 HSMM-based MLLR adaptation.

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#### Maximum Likelihood Linear Regression Based on Hidden Semi-Markov Model

$$b_{i}(\boldsymbol{o}) = \mathcal{N}(\boldsymbol{o}; \boldsymbol{\zeta}\boldsymbol{\mu}_{i} + \boldsymbol{\epsilon}, \boldsymbol{\Sigma}_{i})$$

$$= \mathcal{N}(\boldsymbol{o}; \boldsymbol{W}\boldsymbol{\xi}_{i}, \boldsymbol{\Sigma}_{i})$$

$$p_{i}(d) = \mathcal{N}(d; \boldsymbol{\chi}\boldsymbol{m}_{i} + \boldsymbol{\nu}, \sigma_{i}^{2})$$

$$= \mathcal{N}(d; \boldsymbol{X}\boldsymbol{\phi}_{i}, \sigma_{i}^{2})$$

■  $W = [\zeta, \epsilon] \in \mathcal{R}^{L \times (L+1)}$  and  $X = [\chi, \nu] \in \mathcal{R}^{1 \times 2}$  are the transformation matrices which transform extended mean vectors  $\boldsymbol{\xi}_i = [\boldsymbol{\mu}_i^{\mathsf{T}}, 1]^{\mathsf{T}} \in \mathcal{R}^{L+1}$  and  $\boldsymbol{\phi}_i = [m_i, 1]^{\mathsf{T}} \in \mathcal{R}^2$ , respectively.



- The HSMM-based MLLR adaptation estimates a set of transformation matrices  $\Lambda = (W, X)$  so that the likelihood of the adaptation data O of length T is maximized.
- The problem of the HSMM-based MLLR adaptation based on the ML criterion can be expressed as follows:

$$\tilde{\Lambda} = (\tilde{W}, \tilde{X}) = \underset{\Lambda}{\operatorname{argmax}} P(O|\lambda, \Lambda)$$

• where  $\lambda$  is the parameter set of HSMM.

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#### Maximum Likelihood Linear Regression Based on Hidden Semi-Markov Model

Re-estimation formulas:

$$\overline{\boldsymbol{w}}_{l} = \boldsymbol{y}_{l} \boldsymbol{G}_{l}^{-1}$$

$$\overline{\boldsymbol{X}} = \boldsymbol{z} \boldsymbol{K}^{-1}$$

- where  $w_l \in \mathcal{R}^{L+1}$  is the l-th row vector of W.
- In these equations,  $y_l \in \mathcal{R}^{L+1}$ ,  $G_l \in \mathcal{R}^{(L+1)\times(L+1)}$ ,  $z \in \mathcal{R}^2$ , and  $K \in \mathcal{R}^{2\times 2}$  are given by



$$y_{l} = \sum_{r=1}^{R_{b}} \sum_{t=1}^{T} \sum_{d=1}^{t} \gamma_{t}^{d}(r) \frac{1}{\Sigma_{r}(l)} \sum_{s=t-d+1}^{t} o_{s}(l) \, \boldsymbol{\xi}_{r}^{\top} \qquad \boldsymbol{z} = \sum_{r=1}^{R_{p}} \sum_{t=1}^{T} \sum_{d=1}^{t} \gamma_{t}^{d}(r) \, \frac{1}{\sigma_{r}^{2}} \, d \, \boldsymbol{\phi}_{r}^{\top}$$

$$G_{l} = \sum_{r=1}^{R_{b}} \sum_{t=1}^{T} \sum_{d=1}^{t} \gamma_{t}^{d}(r) \, d \, \frac{1}{\Sigma_{r}(l)} \, \boldsymbol{\xi}_{r} \, \boldsymbol{\xi}_{r}^{\top} \qquad \boldsymbol{K} = \sum_{r=1}^{R_{p}} \sum_{t=1}^{T} \sum_{d=1}^{t} \gamma_{t}^{d}(r) \, \frac{1}{\sigma_{r}^{2}} \, \boldsymbol{\phi}_{r} \, \boldsymbol{\phi}_{r}^{\top},$$

• where  $\Sigma_r(l)$  is the l-th diagonal element of the diagonal covariance matrix  $\Sigma_r$ , and  $o_s(l)$  is the l-th element of the observation vector  $o_s$ .



- It is not always possible to estimate w and x for every distribution, because the amount of adaptation data of a target style is <u>limited</u>.
- Therefore, we use <u>tree structures</u> to group the distributions in the model and to tie the transformation matrices in each group in the same manner as HMMbased techniques.
- $R_b$  and  $R_p$  is respectively the <u>number of the</u> <u>distributions</u> of the state output and duration distributions belonging to this group.

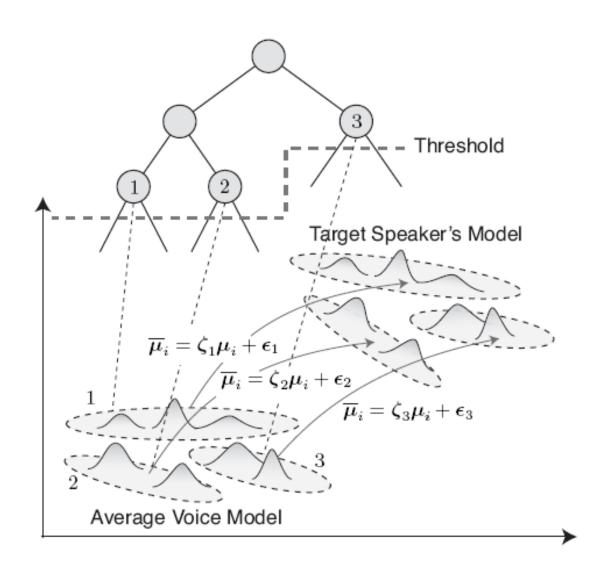


Figure 7.4: Maximum Likelihood Linear Regression



## **Experimental Conditions**

- We used the ATR Japanese speech database (Set B), which contains a set of <u>503 phonetically balanced</u> <u>sentences</u> uttered by <u>6 male speakers</u> and <u>4 female</u> <u>speakers</u>.
- We chose a <u>male speaker MTK</u> and a <u>female speaker</u> <u>FTK</u> as <u>target speakers</u> of the speaker adaptation and used the rest as training speakers for the average voice model.



## **Experimental Conditions**

- In the system, gender-dependent average voice models were trained using 453 sentences for each training speaker.
- The total numbers of training sentences were <u>2265</u> <u>sentences for male-speaker</u> average voice model and <u>1812 sentences for female-speaker</u> average voice model, respectively.



# Objective Evaluation of the HSMM-Based MLLR Adaptation

- We calculated the target speakers' average <u>mel-cepstral</u> <u>distance</u> and <u>root-mean-square</u> (RMS) error of log F0 and vowel duration as the objective measure.
- The adaptation data was from 5 sentences to 450 sentences.
- Fifty test sentences were used for the evaluation, and these were included in neither the training nor the adaptation data.

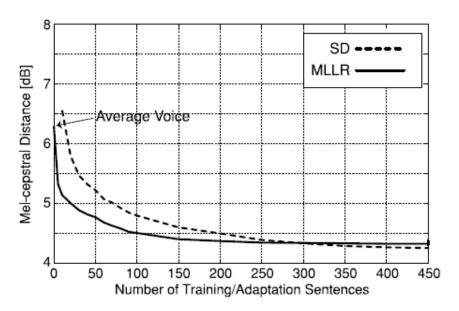


Fig. 11 Average mel-cepstral distance of male speaker MTK.

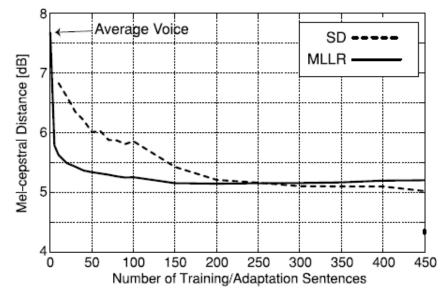


Fig. 14 Average mel-cepstral distance of female speaker FTK.

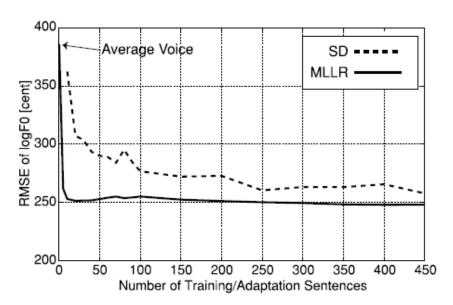


Fig. 12 RMS log F0 error of male speaker MTK.

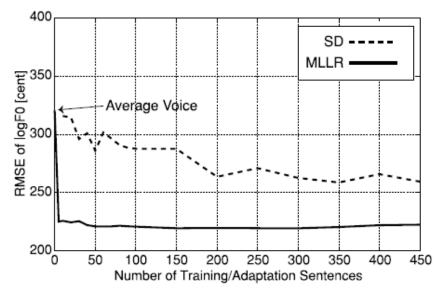


Fig. 15 RMS log F0 error of female speaker FTK.

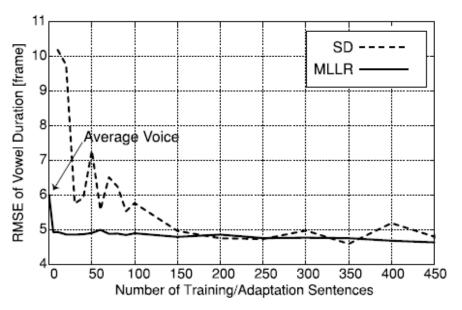


Fig. 13 RMS error of vowel duration of male speaker MTK.

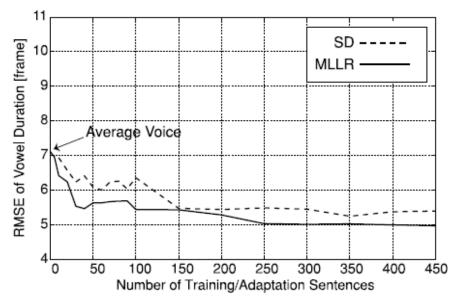


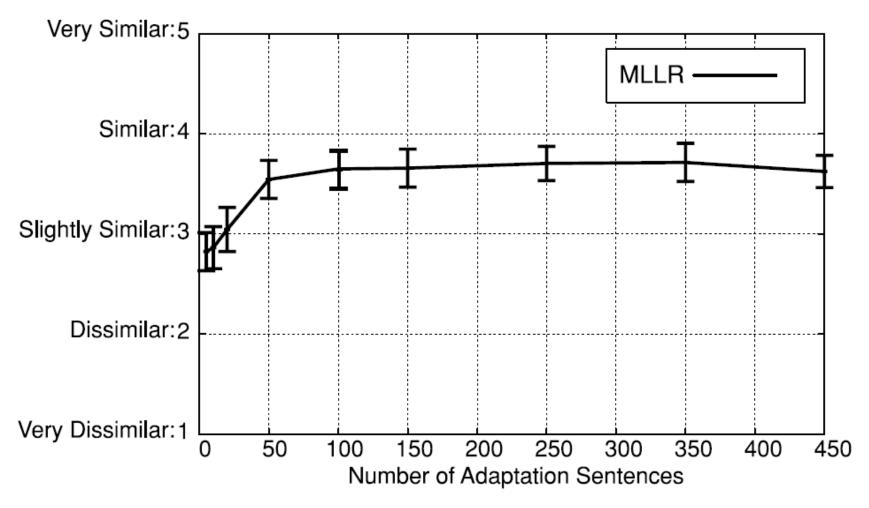
Fig. 16 RMS error of vowel duration of female speaker FTK.



# Subjective Evaluation of the HSMM-Based MLLR Adaptation

- We compared the synthesized speech generated from the adapted models using 5, 10, 20, 50, 100, 150, 250, 350, and 450 sentences of the target speaker.
- For each subject, five test sentences were randomly chosen from a set of 50 test sentences, which were contained in neither the training nor the adaptation data.





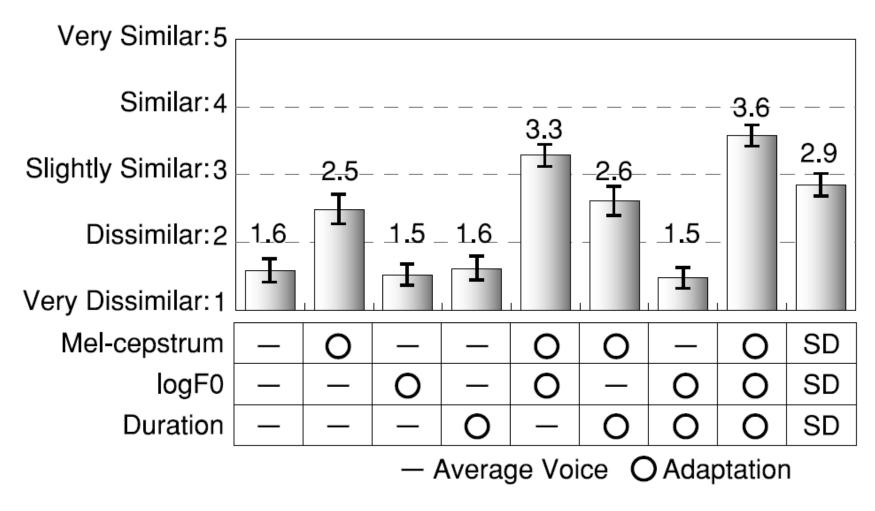
**Fig. 19** Subjective Evaluation of the effect of the number of the adaptation data.



# Subjective Evaluation of the HSMM-Based MLLR Adaptation

- We compared the synthesized speech generated from eight models with or without the adaptation of spectrum, F0, and/or duration.
- The adaptation data comprised 100 sentences.
- For reference, we also compared synthesized speech generated from the SD model using 453 sentences of the target speaker.





**Fig. 20** Subjective evaluation of adaptation effects of each feature.