

Structured Discriminative Models for Speech Recognition

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September 2012



Cambridge University Engineering Department

Symposium on Machine Learning in Speech and Language Processing

Overview

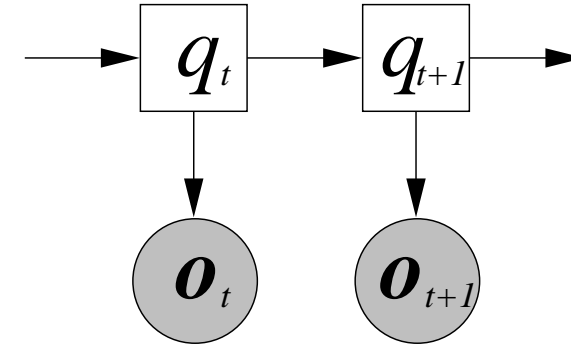
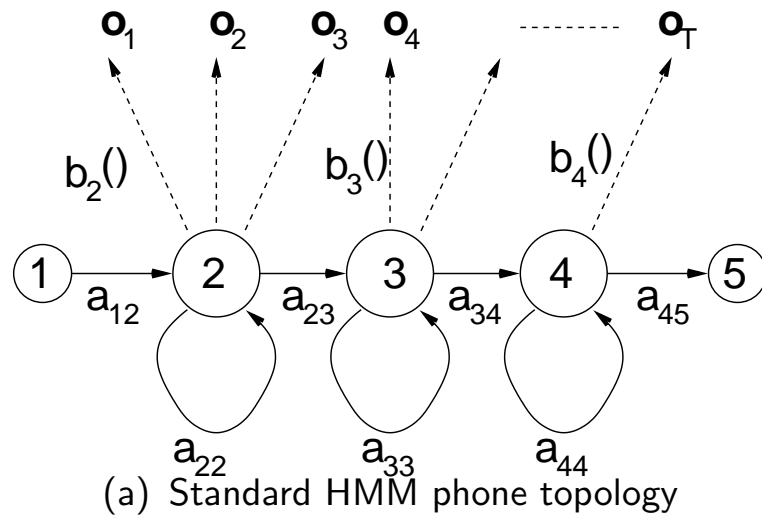
- Acoustic Models for Speech Recognition
 - generative and discriminative models
- Sequence (dynamic) kernels
 - discrete and continuous observation forms
- Combining Generative and Discriminative Models
 - generative score-spaces and log-linear models
 - efficient feature extraction
- Training Criteria
 - large-margin-based training
- Initial Evaluation on Noise Robust Speech Recognition
 - AURORA-2 and AURORA-4 experimental results



Acoustic Models



Hidden Markov Model - a Generative Model



- Conditional independence assumption:
 - observations conditionally independent of other observations given state.
 - states conditionally independent of other states given previous states.

$$p(\mathbf{O}; \lambda) = \sum_{\mathbf{q}} \prod_{t=1}^T P(q_t | q_{t-1}) p(o_t | q_t; \lambda)$$

- Sentence models formed by “glueing” sub-sentence models together

Discriminative Models

- Classification requires class posteriors $P(\mathbf{w}|\mathbf{O})$
 - **Generative model** classification use Bayes' rule e.g. for HMMs

$$P(\mathbf{w}|\mathbf{O}; \boldsymbol{\lambda}) = \frac{p(\mathbf{O}|\mathbf{w}; \boldsymbol{\lambda})P(\mathbf{w})}{\sum_{\tilde{\mathbf{w}}} p(\mathbf{O}|\tilde{\mathbf{w}}; \boldsymbol{\lambda})P(\tilde{\mathbf{w}})}$$

- **Discriminative model** - directly model posterior [1] e.g. **Log-Linear Model**

$$P(\mathbf{w}|\mathbf{O}; \boldsymbol{\alpha}) = \frac{1}{Z} \exp(\boldsymbol{\alpha}^T \phi(\mathbf{O}, \mathbf{w}))$$

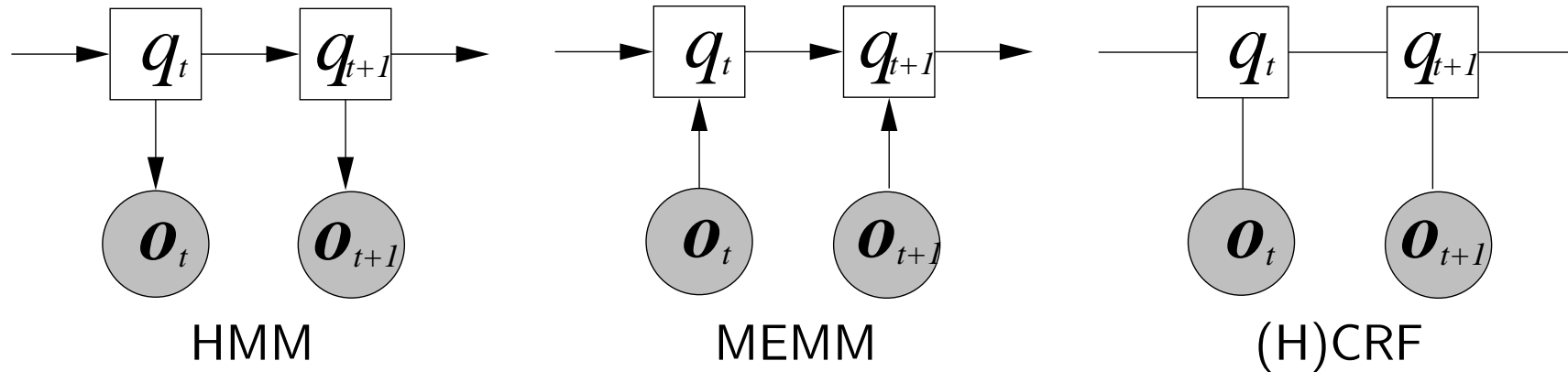
- normalisation term Z (simpler to compute than generative model)

$$Z = \sum_{\tilde{\mathbf{w}}} \exp(\boldsymbol{\alpha}^T \phi(\mathbf{O}, \tilde{\mathbf{w}}))$$

- **BUT** still need to decide form of features $\phi(\mathbf{O}, \mathbf{w})$



Example Standard Sequence Models



- The segmentation, \mathbf{a} , determines the state-sequence \mathbf{q}
 - maximum entropy Markov model [4]

$$P(\mathbf{q}|\mathbf{O}) = \prod_{t=1}^T \frac{1}{Z_t} \exp(\boldsymbol{\alpha}^\top \phi(q_t, q_{t-1}, \mathbf{o}_t))$$

- hidden conditional random field (simplified linear form only) [5]

$$P(\mathbf{q}|\mathbf{O}) = \frac{1}{Z} \prod_{t=1}^T \exp(\boldsymbol{\alpha}^\top \phi(q_t, q_{t-1}, \mathbf{o}_t))$$



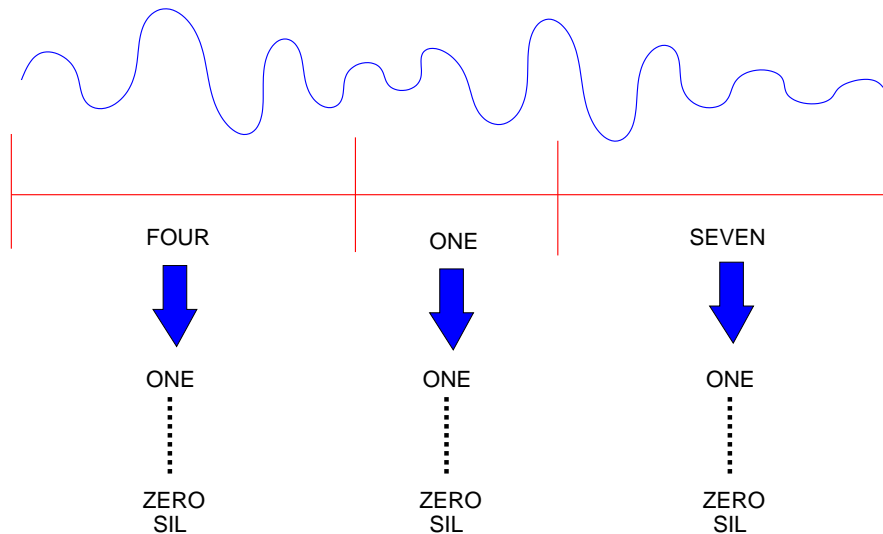
Sequence Discriminative Models

- “Standard” models represent state sequences $P(\mathbf{q}|\mathbf{O})$
 - actually want word posteriors $P(\mathbf{w}|\mathbf{O})$
- Applying discriminative models directly to **speech recognition**:
 1. **Number of possible classes is vast**
 - motivates the use of **structured** discriminative models
 2. **Length of observation \mathbf{O} varies from utterance to utterance**
 - motivates the use of **sequence kernels** to obtain features
 3. **Number of labels (words) and observations (frames) differ**
 - addressed by combining solutions to (1) and (2)



Code-Breaking Style

- Rather than handle complete sequence - split into segments
 - perform simpler classification for each segment
 - complexity determined by segment (simplest word)



1. Using HMM-based hypothesis

- word start/end

2. Foreach segment of a :

- binary SVMs voting

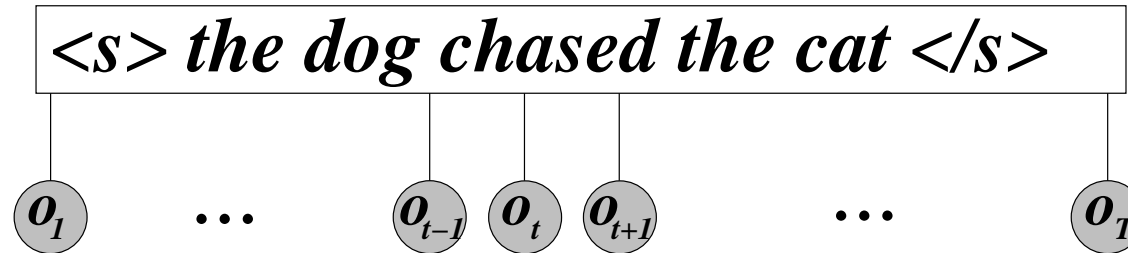
$$- \arg \max_{\omega \in \{\text{ONE}, \dots, \text{SIL}\}} \alpha^{(\omega)T} \phi(\mathbf{O}_{\{a_i\}}, \omega)$$

• Limitations of code-breaking approach [3]

- each segment is treated independently
- restrict to one segmentation, generated by HMMs



Flat Direct Models

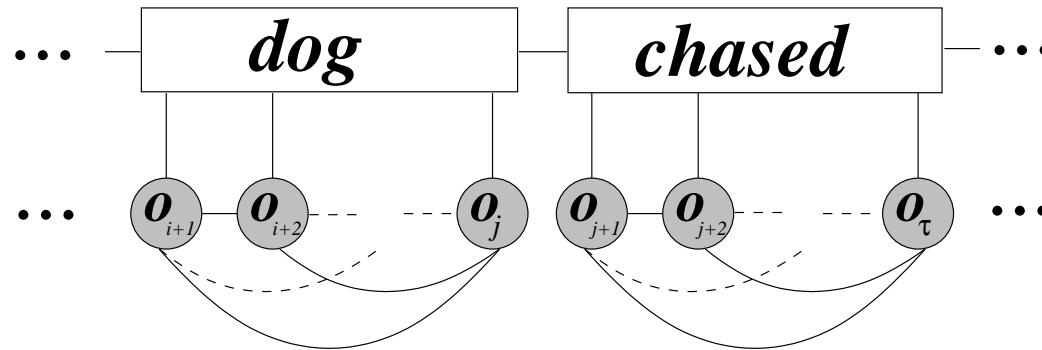


- Log-linear model for complete sentence [7]

$$P(\mathbf{w}|\mathbf{O}) = \frac{1}{Z} \exp(\boldsymbol{\alpha}^T \phi(\mathbf{O}, \mathbf{w}))$$

- Simple model, but **lack of structure** may cause problems
 - extracted feature-space becomes vast (number of possible sentences)
 - associated parameter vector is vast
 - (possibly) large number of unseen examples

Structured Discriminative Models



- Introduce structure into observation sequence [8] - **segmentation a**
 - comprises: segmentation identity a^i , set of observations $\mathbf{O}_{\{a\}}$

$$P(\mathbf{w}|\mathbf{O}) = \frac{1}{Z} \sum_{\mathbf{a}} \exp \left(\alpha^T \left[\sum_{\tau=1}^{|\mathbf{a}|} \phi(\mathbf{O}_{\{a_{\tau}\}}, a_{\tau}^i) \right] \right)$$

- segmentation may be at word, (context-dependent) phone, etc etc
- What form should $\phi(\mathbf{O}_{\{a_{\tau}\}}, a_{\tau}^i)$ have?
 - **must be able to handle variable length $\mathbf{O}_{\{a_{\tau}\}}$**

Features

- Discriminative models performance highly dependent on the features
 - basic features - second-order statistics (almost) a discriminative HMM
 - simplest approach extend frame features (for each unit $w^{(k)}$) [6]

$$\phi(\mathbf{O}_{\{a_\tau\}}, a_\tau^i) = \begin{bmatrix} \vdots \\ \sum_{t \in \{a_\tau\}} \delta(a_\tau^i, w^{(k)}) \mathbf{o}_t \\ \sum_{t \in \{a_\tau\}} \delta(a_\tau^i, w^{(k)}) \mathbf{o}_t \otimes \mathbf{o}_t \\ \sum_{t \in \{a_\tau\}} \delta(a_\tau^i, w^{(k)}) \mathbf{o}_t \otimes \mathbf{o}_t \otimes \mathbf{o}_t \\ \vdots \end{bmatrix}$$

- features have **same** conditional independence assumption as HMM

How to extend range of features?

- Consider extracting features for a **complete segment** of speech
 - number of frames will vary from segment to segment
 - need to map to a fixed dimensionality independent of number of frames



Sequence Kernels



Sequence Kernel

- Sequence kernels are a class of kernel that handles sequence data
 - also applied in a range of biological applications, text processing, speech
 - these kernels may be partitioned into three broad classes
- Discrete-observation kernels
 - appropriate for text data
 - string kernels simplest form
- Distributional kernels (not discussed in this talk)
 - distances between distributions trained on sequences
- Generative kernels:
 - parametric form: use the parameters of the generative model
 - derivative form: use the derivatives with respect to the model parameters



String Kernel

- For speech and text processing input space has variable dimension:
 - use a kernel to map from variable to a fixed length;
 - string kernels are an example for text [9].
- Consider the words cat, cart, bar and a **character** string kernel

	c-a	c-t	c-r	a-r	r-t	b-a	b-r
$\phi(\text{cat})$	1	λ	0	0	0	0	0
$\phi(\text{cart})$	1	λ^2	λ	1	1	0	0
$\phi(\text{bar})$	0	0	0	1	0	1	λ

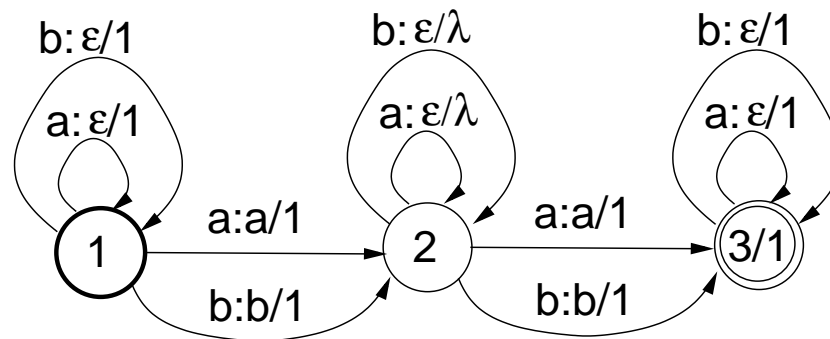
$$K(\text{cat}, \text{cart}) = 1 + \lambda^3, \quad K(\text{cat}, \text{bar}) = 0, \quad K(\text{cart}, \text{bar}) = 1$$

- Successfully applied to various text classification tasks:
 - **how to make process efficient (and more general)?**



Rational Kernels

- Rational kernels [10] encompass various standard feature-spaces and kernels:
 - bag-of-words and N-gram counts, gappy N-grams (string Kernel),
- A **transducer**, T , for the string kernel (gappy bigram) (vocab {a, b})



The **kernel** is: $K(\mathbf{O}_i, \mathbf{O}_j) = w [\mathbf{O}_i \circ (T \circ T^{-1}) \circ \mathbf{O}_j]$

- This form can also handle uncertainty in decoding:
 - **lattices** can be used rather than the 1-best output (\mathbf{O}_i).
- Can also be applied for continuous data kernels [11].



Generative Score-Spaces

- Generative kernels use scores of the following form [12]

$$\phi(\mathbf{O}; \boldsymbol{\lambda}) = [\log(p(\mathbf{O}; \boldsymbol{\lambda}))]$$

– simplest form maps sequence to 1-dimensional score-space

- **Parametric** score-space increase the score-space size

$$\phi(\mathbf{O}; \boldsymbol{\lambda}) = \begin{bmatrix} \hat{\boldsymbol{\lambda}}^{(1)} \\ \vdots \\ \hat{\boldsymbol{\lambda}}^{(K)} \end{bmatrix}$$

– parameters estimated on \mathbf{O} : related to the **mean-supervector** kernel

- **Derivative** score-space take the following form

$$\phi(\mathbf{O}; \boldsymbol{\lambda}) = [\nabla_{\boldsymbol{\lambda}} \log(p(\mathbf{O}; \boldsymbol{\lambda}))]$$

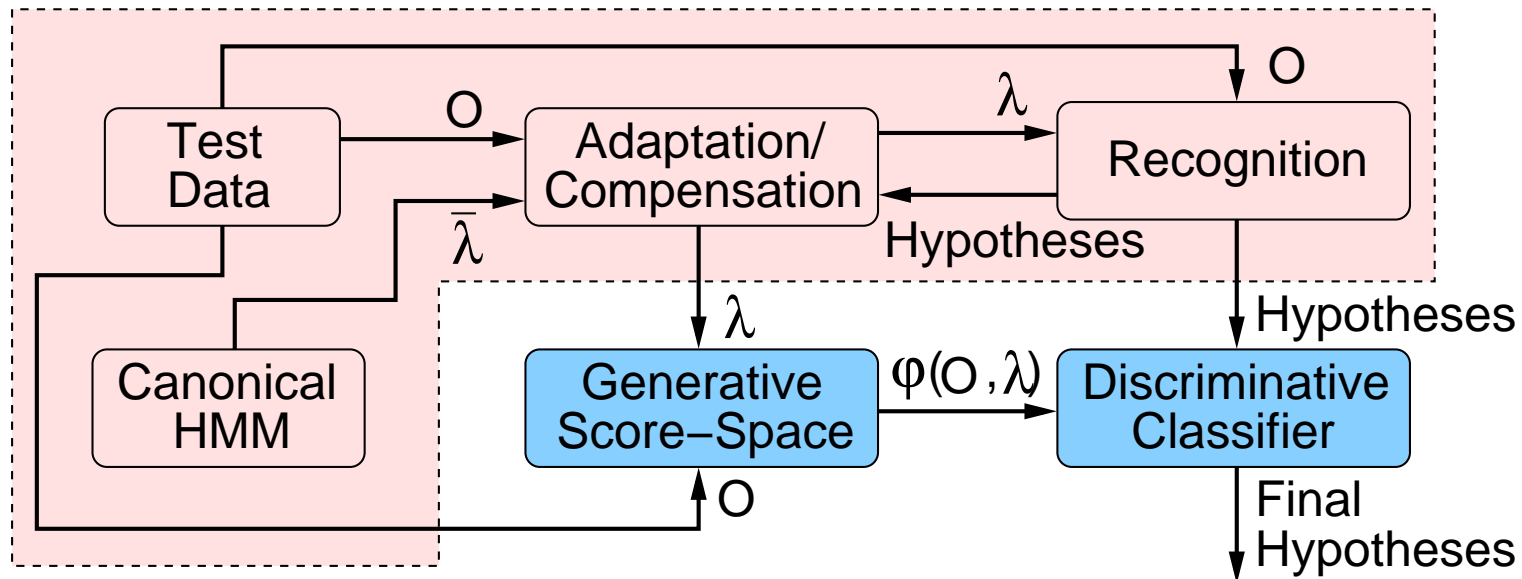
– using the appropriate metric this is the **Fisher** kernel [13]



Combining Generative & Discriminative Models



Combining Discriminative and Generative Models



- Use generative model to extract features [13, 12] (we do like HMMs!)
 - adapt generative model - speaker/noise independent discriminative model
- Use favourite form of discriminative classifier for example
 - log-linear model/logistic regression
 - binary/multi-class support vector machines

Derivative Score-Spaces

- Need a systematic approach to extracting sufficient statistics
 - what about using the sequence-kernel score-spaces?

$$\phi(\mathbf{O}) = \phi(\mathbf{O}; \boldsymbol{\lambda})$$

- does this help with the dependencies?
- For an HMM the mean derivative elements become

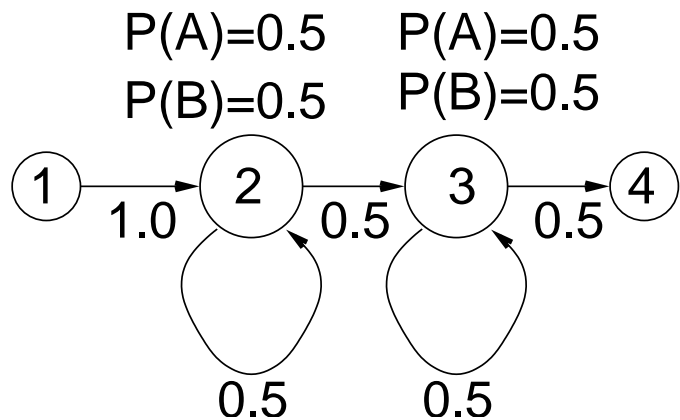
$$\nabla_{\boldsymbol{\mu}^{(jm)}} \log(p(\mathbf{O}; \boldsymbol{\lambda})) = \sum_{t=1}^T P(\mathbf{q}_t = \{\theta_j, m\} | \mathbf{O}; \boldsymbol{\lambda}) \boldsymbol{\Sigma}^{(jm)-1} (\mathbf{o}_t - \boldsymbol{\mu}^{(jm)})$$

- state/component posterior a function of complete sequence \mathbf{O}
- introduces longer term dependencies
- different conditional-independence assumptions than generative model



Score-Space Dependencies

- Consider a simple 2-class, 2-symbol $\{A, B\}$ problem:
 - Class ω_1 : AAAA, BBBB
 - Class ω_2 : AABB, BBAA



Feature	Class ω_1		Class ω_2	
	AAAA	BBBB	AABB	BBAA
Log-Lik	-1.11	-1.11	-1.11	-1.11
∇_{2A}	0.50	-0.50	0.33	-0.33
$\nabla_{2A} \nabla_{2A}^T$	-3.83	0.17	-3.28	-0.61
$\nabla_{2A} \nabla_{3A}^T$	-0.17	-0.17	-0.06	-0.06

- ML-trained HMMs are the same for both classes
- First derivative classes separable, but not linearly separable
 - also true of second derivative within a state
- Second derivative across state linearly separable



Score-Spaces for ASR

- Forms of score-space used in the experiments:

$$\phi_0^a(\mathbf{O}; \boldsymbol{\lambda}) = \begin{bmatrix} \log(p(\mathbf{O}; \boldsymbol{\lambda}^{(1)})) \\ \vdots \\ \log(p(\mathbf{O}; \boldsymbol{\lambda}^{(K)})) \end{bmatrix}; \quad \phi_{1\mu}^b(\mathbf{O}; \boldsymbol{\lambda}) = \begin{bmatrix} \log(p(\mathbf{O}; \boldsymbol{\lambda}^{(i)})) \\ \nabla_{\boldsymbol{\mu}^{(i)}} \log(p(\mathbf{O}; \boldsymbol{\lambda}^{(i)})) \end{bmatrix}$$

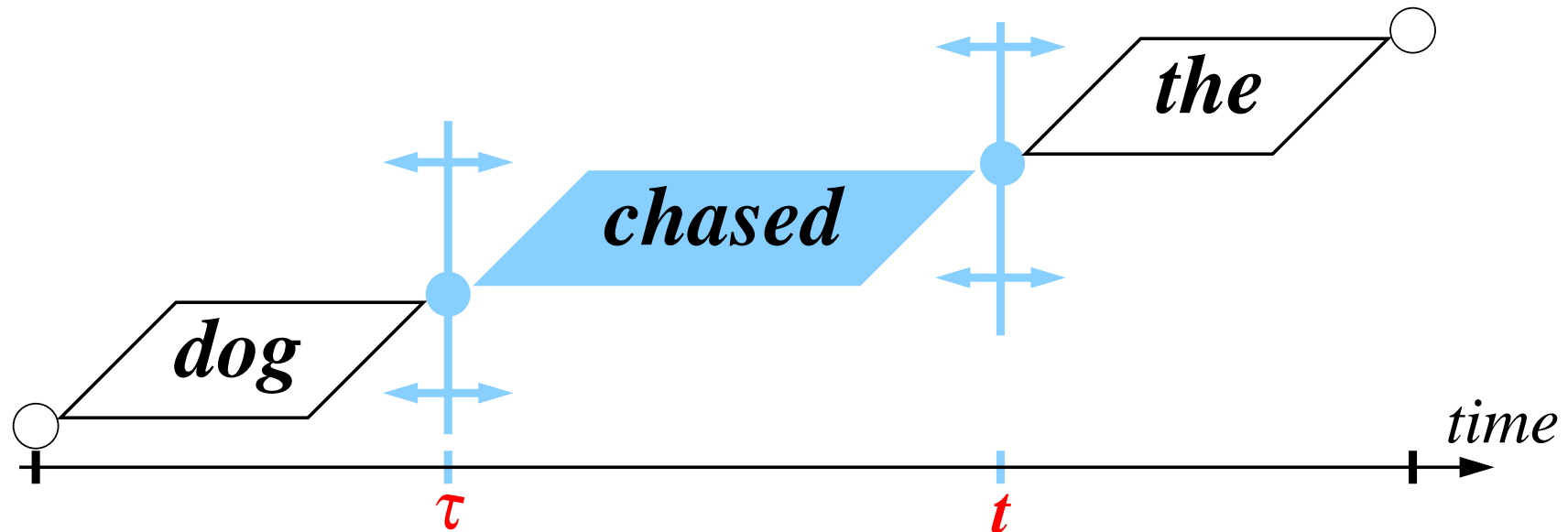
- **appended log-likelihood**: $\phi_0^a(\mathbf{O}; \boldsymbol{\lambda})$
- **derivative** (means only for class ω_i): $\phi_{1\mu}^b(\mathbf{O}; \boldsymbol{\lambda})$
- **log-likelihood** (for class ω_i): $\phi_0^b(\mathbf{O}; \boldsymbol{\lambda}) = [\log(p(\mathbf{O}; \boldsymbol{\lambda}^{(i)}))]$
- In common with most discriminative models **Joint Feature Spaces**,

$$\phi(\mathbf{O}, \mathbf{a}; \boldsymbol{\lambda}) = \begin{bmatrix} \sum_{\tau=1}^{|\mathbf{a}|} \delta(a_{\tau}^i, w^{(1)}) \phi(\mathbf{O}_{\{a_{\tau}\}}; \boldsymbol{\lambda}) \\ \vdots \\ \sum_{\tau=1}^{|\mathbf{a}|} \delta(a_{\tau}^i, w^{(P)}) \phi(\mathbf{O}_{\{a_{\tau}\}}; \boldsymbol{\lambda}) \end{bmatrix}$$

for α -tied yielding “units” $\{w^{(1)}, \dots, w^{(P)}\}$, underlying score-space $\phi(\mathbf{O}; \boldsymbol{\lambda})$.



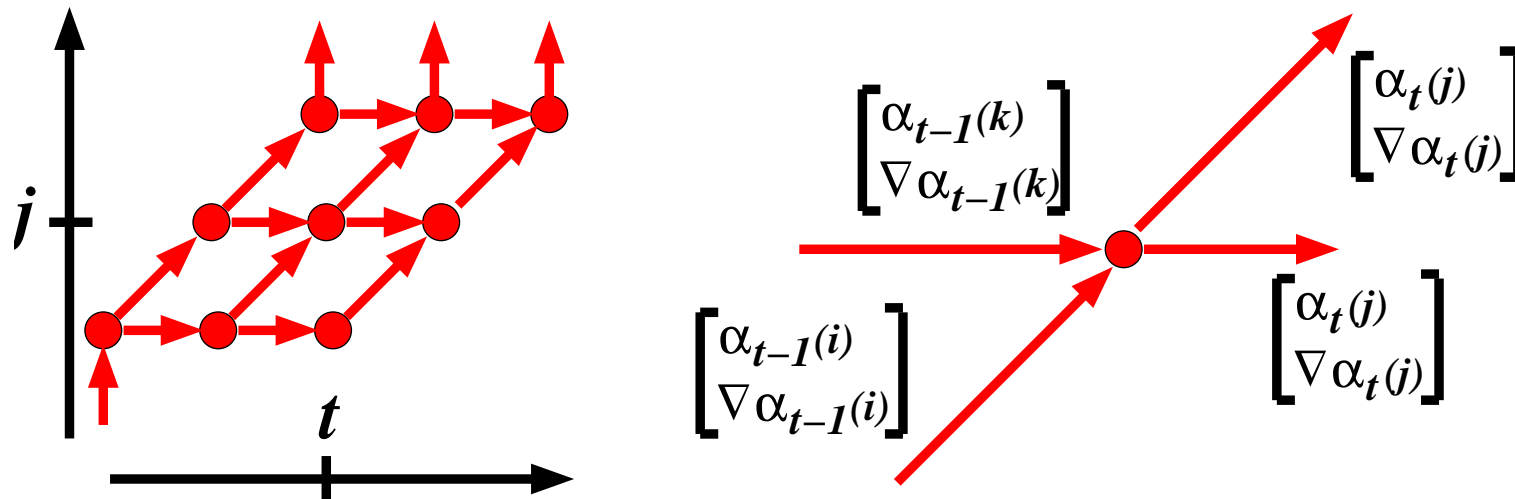
General Feature Extraction



- General features depend on all elements of the observation sequence
 - Consider $\phi(\mathbf{O}_{\tau:t}, w_l)$ for all possible start/end times – T^2 feature evaluations
 - general complexity $\mathcal{O}(T^3)$ – assuming each evaluation $\mathcal{O}(T)$

Computationally expensive!

Efficient Extraction using Expectation Semiring



- Efficient calculate derivative features using expectation semirings [20, 14]
 - extend statistics propagated/combined in forward pass
 - scalar summation extended to vector summation
- Expectation semirings allows to accumulate statistics in one pass
 - derivative features can be computed for **any** node in the trellis - $\mathcal{O}(T^2)$

Handling Speaker/Noise Differences

- A standard problem with discriminative approaches is adaptation/robustness
 - not a problem with generative kernels/score-spaces
 - adapt generative models using **model-based adaptation**
- Standard approaches for speaker/environment adaptation
 - **(Constrained) Maximum Likelihood Linear Regression [15]**

$$\mathbf{x}_t = \mathbf{A}\mathbf{o}_t + \mathbf{b}; \quad \boldsymbol{\mu}^{(m)} = \mathbf{A}\boldsymbol{\mu}_x^{(m)} + \mathbf{b}$$

- **Vector Taylor Series Compensation [16]** (used in this work)

$$\boldsymbol{\mu}^{(m)} = \mathbf{C} \log \left(\exp(\mathbf{C}^{-1}(\boldsymbol{\mu}_x^{(m)} + \boldsymbol{\mu}_h^{(m)})) + \exp(\mathbf{C}^{-1}\boldsymbol{\mu}_n^{(m)}) \right)$$

- Discriminative model parameters speaker/noise independent.

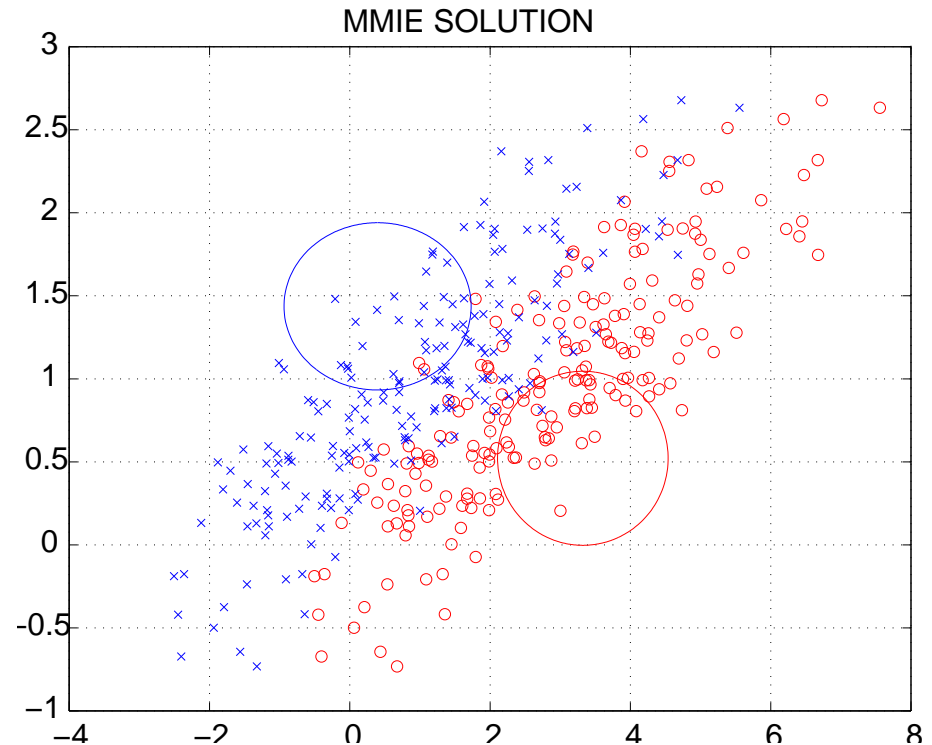
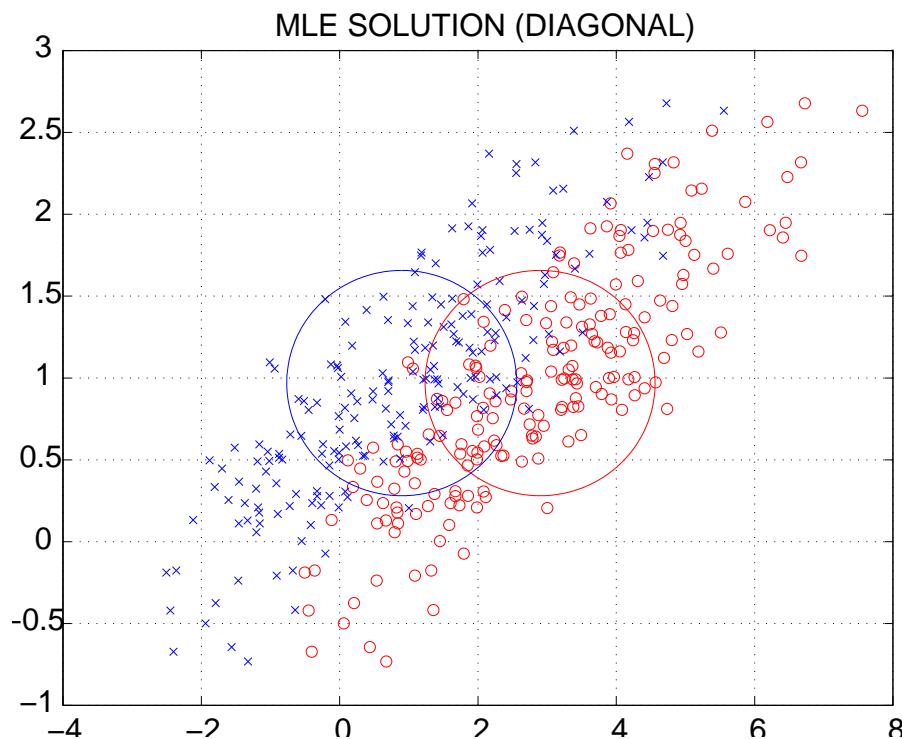


Training Criteria



Simple MMIE Example

- HMMs are not the correct model - discriminative criteria a possibility



- Discriminative criteria a function of posteriors $P(\mathbf{w}|\mathbf{O}; \lambda)$
 - use to train the discriminative model parameters α



Discriminative Training Criteria

- Apply discriminative criteria to train discriminative model parameters α
 - **Conditional Maximum Likelihood (CML)** [21, 22]: maximise

$$\mathcal{F}_{\text{cml}}(\alpha) = \frac{1}{R} \sum_{r=1}^R \log(P(\mathbf{w}_{\text{ref}}^{(r)} | \mathbf{O}^{(r)}; \alpha))$$

- **Minimum Classification Error (MCE)** [23]: minimise

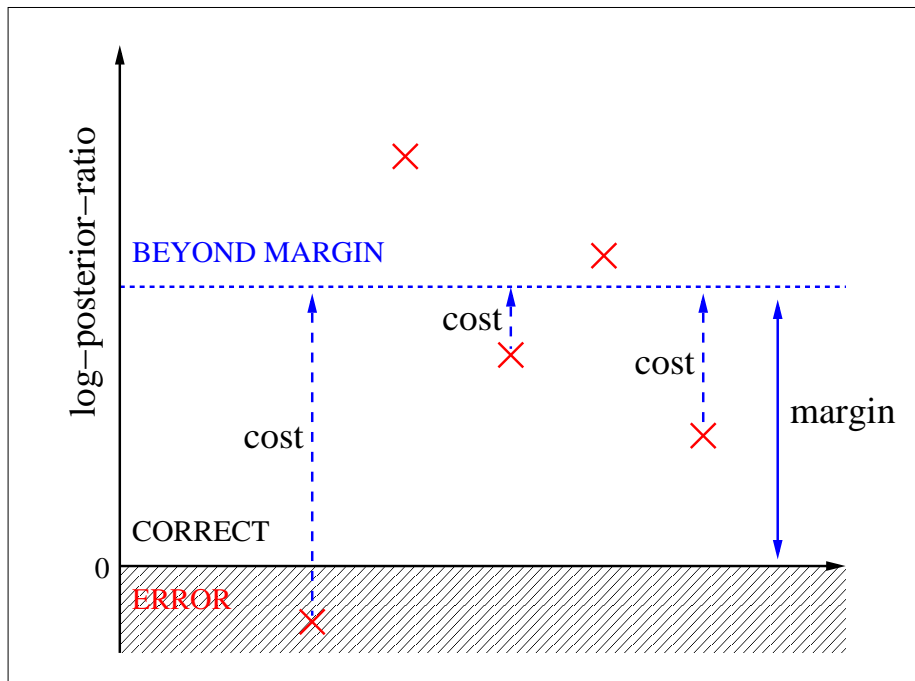
$$\mathcal{F}_{\text{mce}}(\alpha) = \frac{1}{R} \sum_{r=1}^R \left(1 + \left[\frac{P(\mathbf{w}_{\text{ref}}^{(r)} | \mathbf{O}^{(r)}; \alpha)}{\sum_{\mathbf{w} \neq \mathbf{w}_{\text{ref}}^{(r)}} P(\mathbf{w} | \mathbf{O}^{(r)}; \alpha)} \right]^{\rho} \right)^{-1}$$

- **Minimum Bayes' Risk (MBR)** [24, 25]: minimise

$$\mathcal{F}_{\text{mbr}}(\alpha) = \frac{1}{R} \sum_{r=1}^R \sum_{\mathbf{w}} P(\mathbf{w} | \mathbf{O}^{(r)}; \alpha) \mathcal{L}(\mathbf{w}, \mathbf{w}_{\text{ref}}^{(r)})$$



Large Margin Based Criteria



- Standard criterion for SVMs
 - improves generalisation
- Require log-posterior-ratio

$$\min_{\mathbf{w} \neq \mathbf{w}_{\text{ref}}} \left\{ \log \left(\frac{P(\mathbf{w}_{\text{ref}} | \mathbf{O}; \boldsymbol{\alpha})}{P(\mathbf{w} | \mathbf{O}; \boldsymbol{\alpha})} \right) \right\}$$

to be beyond margin

- As sequences being used can make margin function of the “loss” - **minimise**

$$\mathcal{F}_{\text{lm}}(\boldsymbol{\alpha}) = \frac{1}{R} \sum_{r=1}^R \left[\max_{\mathbf{w} \neq \mathbf{w}_{\text{ref}}^{(r)}} \left\{ \mathcal{L}(\mathbf{w}, \mathbf{w}_{\text{ref}}^{(r)}) - \log \left(\frac{P(\mathbf{w}_{\text{ref}}^{(r)} | \mathbf{O}^{(r)}; \boldsymbol{\alpha})}{P(\mathbf{w} | \mathbf{O}^{(r)}; \boldsymbol{\alpha})} \right) \right\} \right]_+$$

use **hinge-loss** $[f(x)]_+$. Many variants possible [26, 27, 28, 29]



Relationship to (Structured) SVM

- Commonly add a Gaussian prior for regularisation

$$\mathcal{F}(\boldsymbol{\alpha}) = \log(\mathcal{N}(\boldsymbol{\alpha}; \boldsymbol{\mu}_\alpha; \boldsymbol{\Sigma}_\alpha)) + \mathcal{F}_{\text{lm}}(\boldsymbol{\alpha})$$

- Make the posteriors a log-linear model ($\boldsymbol{\alpha}$) with generative score-space ($\boldsymbol{\lambda}$) [30]
 - restrict parameters of the prior: $\mathcal{N}(\boldsymbol{\alpha}; \boldsymbol{\mu}_\alpha; \boldsymbol{\Sigma}_\alpha) = \mathcal{N}(\boldsymbol{\alpha}; \mathbf{0}, C\mathbf{I})$

$$\mathcal{F}(\boldsymbol{\alpha}) = \frac{1}{2} \|\boldsymbol{\alpha}\|^2 + \frac{C}{R} \sum_{r=1}^R \left[\max_{\mathbf{w} \neq \mathbf{w}_{\text{ref}}^{(r)}} \left\{ \mathcal{L}(\mathbf{w}, \mathbf{w}_{\text{ref}}^{(r)}) - \log \left(\frac{\boldsymbol{\alpha}^\top \boldsymbol{\phi}(\mathbf{O}^{(r)}, \mathbf{w}_{\text{ref}}^{(r)}; \boldsymbol{\lambda})}{\boldsymbol{\alpha}^\top \boldsymbol{\phi}(\mathbf{O}^{(r)}, \mathbf{w}; \boldsymbol{\lambda})} \right) \right\} \right]_+$$

- Standard result - it's a **structured SVM** [31, 30]



Structured SVM Training

- Training α , so that $\alpha^T \phi(\mathbf{O}, \mathbf{w})$ is max for correct reference \mathbf{w}_{ref} :

$$\begin{array}{l}
 \min_{\alpha} \quad \frac{1}{2} \|\alpha\|^2 \\
 \text{s.t.} \quad \alpha^T \phi(\text{audio}, \text{"123"}) \geq \alpha^T \phi(\text{audio}, \text{"000"}) + 1, \\
 \qquad \qquad \qquad \vdots \\
 \alpha^T \phi(\text{audio}, \text{"123"}) \geq \alpha^T \phi(\text{audio}, \text{"999"}) + 1, \\
 \qquad \qquad \qquad \vdots \\
 \qquad \qquad \qquad \vdots \\
 \alpha^T \phi(\text{audio}, \text{"456"}) \geq \alpha^T \phi(\text{audio}, \text{"000"}) + 1, \\
 \qquad \qquad \qquad \vdots \\
 \alpha^T \phi(\text{audio}, \text{"456"}) \geq \alpha^T \phi(\text{audio}, \text{"999"}) + 1, \\
 \qquad \qquad \qquad \vdots
 \end{array}
 \begin{array}{l}
 \left. \begin{array}{l} \\ \\ \\ \\ \end{array} \right\} \text{Training Sample 1} \\
 \begin{array}{l} \text{audio} \\ \text{"123"} \end{array} \mathbf{O}^{(1)}, \\
 \mathbf{w}_{\text{ref}}^{(1)}, \\
 \\
 \left. \begin{array}{l} \\ \\ \\ \\ \end{array} \right\} \text{Training Sample n} \\
 \begin{array}{l} \text{audio} \\ \text{"456"} \end{array} \mathbf{O}^{(n)}, \\
 \mathbf{w}_{\text{ref}}^{(n)},
 \end{array}$$

- General unconstrained form: use [cutting plane algorithm](#) to solve [32, 33]

$$\frac{1}{2} \|\alpha\|^2 + \frac{C}{R} \sum_{r=1}^n \left[- \overbrace{\alpha^T \phi(\mathbf{O}^{(r)}, \mathbf{w}_{\text{ref}}^{(r)})}^{\text{linear}} + \overbrace{\max_{\mathbf{w} \neq \mathbf{w}_{\text{ref}}^{(r)}} \left\{ \mathcal{L}(\mathbf{w}, \mathbf{w}_{\text{ref}}^{(r)}) + \alpha^T \phi(\mathbf{O}^{(r)}, \mathbf{w}) \right\}}^{\text{convex}} \right]_+$$



Handling Latent Variables

- Ignored the issue of alignment so far
 - for SSVM necessary to use the “best” segmentation
- Simplest solution is to use the single segmentation from the original HMM

$$\hat{\mathbf{a}}_{\text{hmm}} = \underset{\mathbf{a}}{\operatorname{argmax}} \{ \log (P(\mathbf{a}|\mathbf{O}, \mathbf{w}; \boldsymbol{\lambda})) \} = \underset{\mathbf{a}}{\operatorname{argmax}} \{ \log (P(\mathbf{O}|\mathbf{a}, \mathbf{w}; \boldsymbol{\lambda})P(\mathbf{a}|\mathbf{w}; \boldsymbol{\lambda})) \}$$

- equivalent of phone/word-marking lattices
- **BUT** underlying model changes: would like

$$\hat{\mathbf{a}} = \underset{\mathbf{a}}{\operatorname{argmax}} \{ \log (P(\mathbf{O}|\mathbf{a}, \mathbf{w}; \boldsymbol{\lambda}, \boldsymbol{\alpha})) + \log (P(\mathbf{a}|\mathbf{w}; \boldsymbol{\lambda}, \boldsymbol{\alpha})) \}$$

Maps into a **Concave-Convex Procedure** (CCCP) [34]

$$\left[\overbrace{-\max_{\mathbf{a}} \boldsymbol{\alpha}^T \phi(\mathbf{O}^{(i)}, \mathbf{w}_{\text{ref}}^{(i)}, \mathbf{a})}^{\text{concave}} + \overbrace{\max_{\mathbf{w} \neq \mathbf{w}_{\text{ref}}, \mathbf{a}} \left\{ \mathcal{L}(\mathbf{w}, \mathbf{w}_{\text{ref}}^{(i)}) + \boldsymbol{\alpha}^T \phi(\mathbf{O}^{(i)}, \mathbf{w}, \mathbf{a}) \right\}}^{\text{convex}} \right]_+$$



Evaluation Tasks



Preliminary Evaluation Tasks

- **AURORA-2** small vocabulary digit string recognition task
 - whole-word models, 16 emitting-states with 3 components per state
 - clean training data for HMM training - HTK parametrisation SNR
 - Set B and Set C unseen noise conditions even for multi-style data
 - **Noise estimated in a ML-fashion** for each utterance
- **AURORA-4** medium vocabulary speech recognition
 - training data from WSJ0 SI84 to train clean acoustic models
 - state-clustered states, cross-word triphones ($\approx 3K$ states $\approx 50k$ components)
 - 5-15dB SNR range of noises added
 - **Noise estimated in a ML-fashion** for each utterance
- **WARNING:** optimisation techniques improved over time
 - don't compare results cross-tables!



AURORA-2 - Training Criterion

Model	Criterion	Test set			Avg
		A	B	C	
HMM	—	9.8	9.1	9.5	9.5
LLM (ϕ_0^a)	CML	8.1	7.7	8.3	8.1
	MWE	7.9	7.4	8.2	7.9
	LM	7.8	7.3	8.0	7.6

- All approaches yield gains over the baseline VTS system
 - very few additional parameters added ($12 \times 12 = 144$) for log-linear models (though these parameters are discriminatively trained)
- Large-margin log-linear model will be referred to as Structured SVM



AURORA-2 - Support Vector Machines

Model	Features	Test set			Avg
		A	B	C	
HMM	—	9.8	9.1	9.5	9.5
SVM	ϕ_0^a	9.1	8.7	9.2	9.0
MSVM		8.3	8.1	8.6	8.3
SSVM		7.8	7.3	8.0	7.6

- Possible to compare SSVM with more standard SVMs
 - segmentation for SVMs and multi-class SVMs (MSVMs) obtained from HMM
 - majority voting (HMM decision for ties on standard SVM)
- The difference between the MSVM and SSVM is the fixed HMM segmentation
 - does have an important on the performance



AURORA-2 - Derivative Score-Spaces - MWE Criterion

HMM	SDM	\hat{a}	Test set			Avg
			A	B	C	
VTS	—	—	9.8	9.1	9.5	9.5
	$\phi_{1\mu}^b$	\hat{a}_{hmm}	7.0	6.6	7.6	7.0
		\hat{a}	6.8	6.4	7.3	6.7
VAT	—	—	8.9	8.3	8.8	8.6
	$\phi_{1\mu}^b$	\hat{a}_{hmm}	6.6	6.5	7.0	6.6
		\hat{a}	6.2	6.1	6.8	6.3
DVAT	—	—	6.7	6.6	7.0	6.7
	$\phi_{1\mu}^b$	\hat{a}_{hmm}	6.1	6.2	6.7	6.3
		\hat{a}	6.1	6.1	6.6	6.2

- Derivative score-spaces ($\phi_{1\mu}^b$) consistent gains over all baseline HMM systems
 - derivative score-space larger (1873 dimensions for each base score-space)
 - adds approximately 50% more parameters to the system



AURORA-4 - Derivative Score-Space - MPE Criterion

System	Test set				Avg
	A	B	C	D	
VTS	7.1	15.3	12.1	23.1	17.9
VAT	8.6	13.8	12.0	20.1	16.0
DVAT	7.2	12.8	11.5	19.7	15.3
VAT + ϕ_0^b	7.7	13.1	11.0	19.5	15.3
VAT + $\phi_{1\mu}^b$	7.4	12.6	10.7	19.0	14.8

- Contrast of DVAT system with log-linear system (4020 classes)
 - single dimension space (ϕ_0^b) with VAT system yields DVAT performance
- Gains from derivative score-space disappointing (limited training data)
 - need to look at DVAT + $\phi_{1\mu}^b$ (need to try on more data)



Conclusions

- Combination of generative and discriminative models
 - use generative models to derive features for discriminative model
 - robustness and adaptation achieved by adapting underlying acoustic model
- Derivative features of generative models
 - different conditional independence assumptions to underlying model
 - systematic way to incorporate different dependencies into model
- Large margin training criterion
 - yields structured SVM (use standard optimisation code)
 - still an issue scaling to large tasks/score-spaces

Interesting classifier options - without throwing away HMMs



Acknowledgements

- This work has been funded from the following sources:

TOSHIBA
Leading Innovation >>>

EPSRC
Pioneering research
and skills

- Cambridge Research Lab, Toshiba Research Europe Ltd
- EPSRC - Generative Kernels and Score-Spaces for Classification of Speech



References

- [1] C. M. Bishop, *Pattern Recognition and Machine Learning*, Springer Verlag, 2006.
- [2] G. Zweig et al, "Speech recognition with segmental conditional random fields: A summary of the JHU CLSP Summer workshop," in *Proceedings of ICASSP*, 2011.
- [3] V. Venkataramani, S. Chakrabartty, and W. Byrne, "Support vector machines for segmental minimum Bayes risk decoding of continuous speech," in *ASRU 2003*, 2003.
- [4] H-K. Kuo and Y. Gao, "Maximum entropy direct models for speech recognition," *IEEE Transactions Audio Speech and Language Processing*, 2006.
- [5] A. Gunawardana, M. Mahajan, A. Acero, and J.C. Platt, "Hidden conditional random fields for phone classification," in *Interspeech*, 2005.
- [6] S. Wiesler, M. Nußbaum-Thom, G. Heigold, R. Schlüter, and H. Ney, "Investigations on features for log-linear acoustic models in continuous speech recognition," in *Automatic Speech Recognition & Understanding, 2009. ASRU 2009. IEEE Workshop on*, 2009, pp. 52–57.
- [7] P. Nguyen, G. Heigold, and G. Zweig, "Speech recognition with flat direct models," *IEEE Journal of Selected Topics in Signal Processing*, vol. 4, pp. 994–1006, 2010.
- [8] M. J. F. Gales, S. Watanabe, and E. Fosler-Lussier, "Structure discriminative models for speech recognition," *IEEE Signal Processing Magazine*, 2012.
- [9] H. Lodhi, C. Saunders, J. Shawe-Taylor, N. Cristianini, and C. Watkins, "Text classification using string kernels," *Journal of Machine Learning Research*, vol. 2, pp. 419–444, 2002.
- [10] C. Cortes, P. Haffner, and M. Mohri, "Weighted automata kernels - general framework and algorithms," in *Proc. Eurospeech*, 2003.
- [11] Layton MI and MJF Gales, "Acoustic modelling using continuous rational kernels," *Journal of VLSI Signal Processing Systems for Signal, Image, and Video Technology*, August 2007.
- [12] N.D. Smith and M.J.F. Gales, "Speech recognition using SVMs," in *Advances in Neural Information Processing Systems*, 2001.
- [13] T. Jaakkola and D. Haussler, "Exploiting generative models in discriminative classifiers," in *Advances in Neural Information Processing Systems 11*, S.A. Solla and D.A. Cohn, Eds. 1999, pp. 487–493, MIT Press.
- [14] R. C. van Dalen, A. Ragni, and M. J. F. Gales, "Efficient decoding with continuous rational kernels using the expectation semiring," Tech. Rep. CUED/F-INFENG/TR.674, 2012.



- [15] M J F Gales, "Maximum likelihood linear transformations for HMM-based speech recognition," *Computer Speech and Language*, vol. 12, pp. 75–98, 1998.
- [16] A Acero, L Deng, T Kristjansson, and J Zhang, "HMM Adaptation using Vector Taylor Series for Noisy Speech Recognition," in *Proc. ICSLP*, Beijing, China, 2000.
- [17] M. Layton, *Augmented statistical models for classifying sequence data*, Ph.D. thesis, Cambridge University, 2006.
- [18] G. Zweig and P. Nguyen, "A segmental CRF approach to large vocabulary continuous speech recognition," in *ASRU*, 2009, pp. 152–157.
- [19] A. Ragni and M. J. F. Gales, "Structured discriminative models for noise robust continuous speech recognition," in *ICASSP*, 2011, pp. 4788–4791.
- [20] J. Eisner, "Parameter estimation for probabilistic finite-state transducers," in *Proc. ACL*, 2002.
- [21] P.S. Gopalakrishnan, D. Kanevsky, A. Nádas, and D. Nahamoo, "An inequality for rational functions with applications to some statistical estimation problems," *IEEE Trans. Information Theory*, 1991.
- [22] P. C. Woodland and D. Povey, "Large scale discriminative training of hidden Markov models for speech recognition," *Computer Speech & Language*, vol. 16, pp. 25–47, 2002.
- [23] B.-H. Juang and S. Katagiri, "Discriminative learning for minimum error classification," *IEEE Transactions on Signal Processing*, 1992.
- [24] J. Kaiser, B. Horvat, and Z. Kacic, "A novel loss function for the overall risk criterion based discriminative training of HMM models," in *Proc. ICSLP*, 2000.
- [25] W. Byrne, "Minimum Bayes risk estimation and decoding in large vocabulary continuous speech recognition," *IEICE Special Issue on Statistical Modelling for Speech Recognition*, 2006.
- [26] F. Sha and L.K. Saul, "Large margin gaussian mixture modelling for phonetic classification and recognition," in *ICASSP*, 2007.
- [27] J. Li, M. Siniscalchi, and C-H. Lee, "Approximate test risk minimization through soft margin training," in *ICASSP*, 2007.
- [28] G Heigold, T Deselaers, R Schluter, and H Ney, "Modified MMI/MPE: A direct evaluation of the margin in speech recognition," in *Proc. ICML*, 2008.
- [29] G Saon and D Povey, "Penalty function maximization for large margin HMM training," in *Proc. Interspeech*, 2008.
- [30] S.-X. Zhang, Anton Ragni, and M. J. F. Gales, "Structured log linear models for noise robust speech recognition," *Signal Processing Letters, IEEE*, vol. 17, pp. 945–948, 2010.



- [31] Ioannis Tsochantaridis, Thorsten Joachims, Thomas Hofmann, and Yasemin Altun, “Large margin methods for structured and interdependent output variables,” *J. Mach. Learn. Res.*, vol. 6, pp. 1453–1484, 2005.
- [32] Thorsten Joachims, Thomas Finley, and Chun-Nam John Yu, “Cutting-plane training of structural SVMs,” *Mach. Learn.*, vol. 77, no. 1, pp. 27–59, 2009.
- [33] S.-X. Zhang and M. J. F. Gales, “Extending noise robust structured support vector machines to larger vocabulary tasks,” in *Proc. ASRU*, 2011.
- [34] Chun-Nam Yu and Thorsten Joachims, “Learning structural SVMs with latent variables,” in *Proceedings of ICML*, 2009.
- [35] W.M. Campbell, D. Sturim, D.A. Reynolds, and A. Solomonoff, “SVM based speaker verification using a GMM supervector kernel and NAP variability compensation,” in *Proc. ICASSP*, 2006.



Distributional Kernels

- General family of kernel that operates on distances between distributions
 - using the available estimate a distribution given the sequence

$$\boldsymbol{\lambda}^{(i)} = \underset{\boldsymbol{\lambda}}{\operatorname{argmax}} \{ \log(p(\mathbf{O}_i; \boldsymbol{\lambda})) \}$$

- Forms of kernel normally based (f_i distribution with parameters $\boldsymbol{\lambda}^{(i)}$)
 - **Kullback-Leibler divergence:**

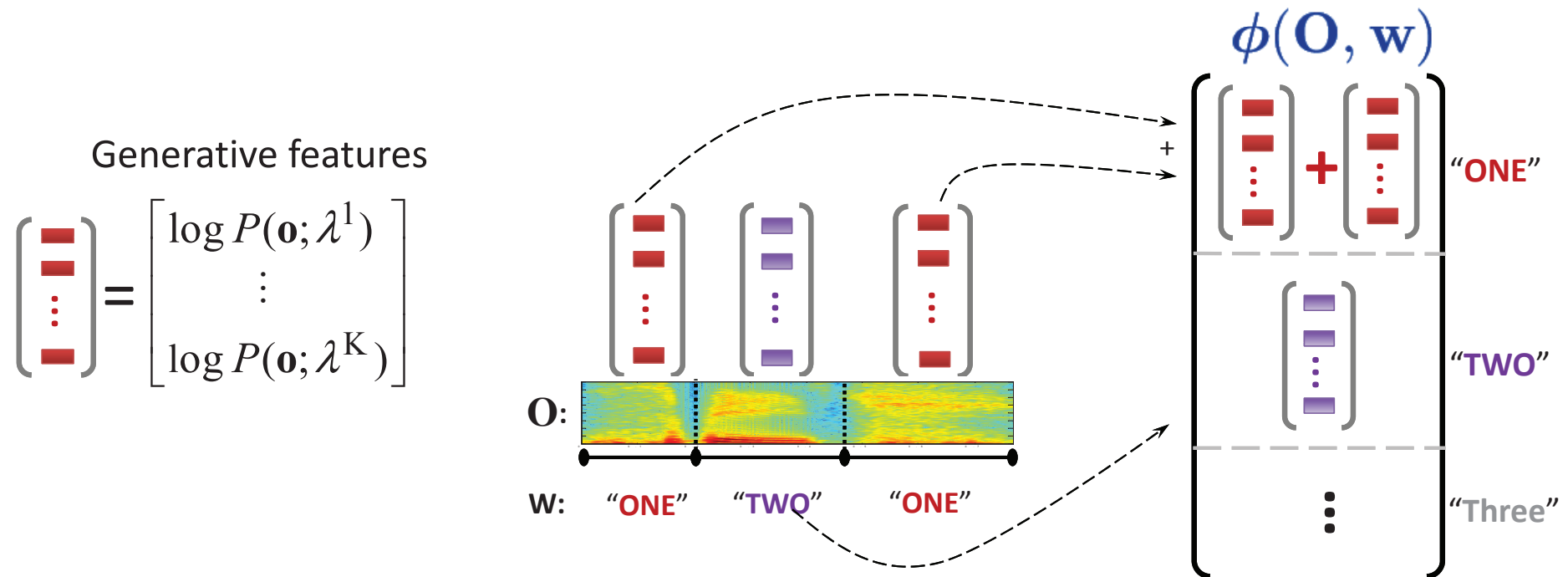
$$\mathcal{KL}(f_i || f_j) = \int f_i(\mathbf{O}) \log \left(\frac{f_i(\mathbf{O})}{f_j(\mathbf{O})} \right) d\mathbf{O}$$

- **Bhattacharyya affinity measure:**

$$\mathcal{B}(f_i || f_j) = \int \sqrt{f_i(\mathbf{O}) f_j(\mathbf{O})} d\mathbf{O}$$



Joint Feature-Space Example



- Size of joint feature-space is the **product** of
 1. **feature-space size** (K)- determined by generative model
 2. **number of α classes** (P) - determined by discriminative model
- Segmentation of the sentence will alter scores

GMM Mean-Supervector Kernel

- GMM-mean supervector derived from a range of approximations [35]
 - use symmetric KL-divergence: $\mathcal{KL}(f_i||f_j) + \mathcal{KL}(f_j||f_i)$
 - use matched pair KL-divergence approximation
 - GMM distributions **only** differ in terms of the means
 - use polarisation identity
- Form of kernel is

$$K(\mathbf{O}_i, \mathbf{O}_j; \boldsymbol{\lambda}) = \sum_{m=1}^M c_m \boldsymbol{\mu}^{(im)\top} \boldsymbol{\Sigma}^{(m)-1} \boldsymbol{\mu}^{(jm)}$$

- $\boldsymbol{\mu}^{(im)}$ is the mean (ML or MAP) for component m using sequence \mathbf{O}_i
- Used in a range of speaker verification applications
 - **BUT** required to explicitly operate in feature-space



AURORA-2 - Optimising Segmentation

Model	Training	Segmentation {trn, tst}	Test set			Avg
			A	B	C	
HMM	—	—	9.8	9.1	9.5	9.5
SSVM	<i>n</i> -slack	$\{\hat{\mathbf{a}}_{\text{hmm}}, \hat{\mathbf{a}}_{\text{hmm}}\}$	7.8	7.3	8.0	7.6
		$\{\hat{\mathbf{a}}_{\text{hmm}}, \hat{\mathbf{a}}\}$	7.6	7.2	8.0	7.5
SSVM	<i>n</i> -slack batch	$\{\hat{\mathbf{a}}_{\text{hmm}}, \hat{\mathbf{a}}_{\text{hmm}}\}$	7.9	7.4	8.2	7.8
		$\{\hat{\mathbf{a}}_{\text{hmm}}, \hat{\mathbf{a}}\}$	7.8	7.2	8.0	7.6
		$\{\hat{\mathbf{a}}, \hat{\mathbf{a}}\}$	7.6	7.1	7.8	7.4
SSVM	1-slack	$\{\hat{\mathbf{a}}_{\text{hmm}}, \hat{\mathbf{a}}\}$	7.6	7.3	7.9	7.5

- Just using the HMM segmentation is suboptimal in terms of WER
 - *n*-slack batch and 1-slack schemes similar to full approach



AURORA-4 - Structured SVM Results

- SSVM training configuration:
 - 1-slack variable training
 - prior distribution matched to score-space ϕ_0^a , mean set to $1/\text{LM}$ – scale
 - α tied at the monophone-level (47-classes)

Model	Segmentation $\{\text{trn}, \text{tst}\}$	Test set				Avg
		A	B	C	D	
HMM	—	7.1	15.3	12.1	23.1	17.9
SSVM	$\{\hat{\mathbf{a}}_{\text{hmm}}, \hat{\mathbf{a}}_{\text{hmm}}\}$	7.5	14.3	11.4	21.9	16.9
	$\{\hat{\mathbf{a}}_{\text{hmm}}, \hat{\mathbf{a}}\}$	7.4	14.2	11.3	21.9	16.8

- SSVM gains over baseline HMM-VTS system
 - disappointing gain from segmentation - though only in test at the moment
 - working on optimal training segmentation as well



AURORA-4 - Derivative Score-Space

Classes	System	Comp tied α	Test set				Avg
			A	B	C	D	
	VTS		7.1	15.3	12.1	23.1	17.9
47	$\phi_{1\mu}^b$	yes	7.5	14.1	11.3	21.6	16.6
		no	7.4	14.3	11.7	21.9	16.9
4020	$\phi_{1\mu}^b$	yes	6.8	13.7	10.6	21.3	16.2
		no	6.7	13.5	10.2	21.1	16.0

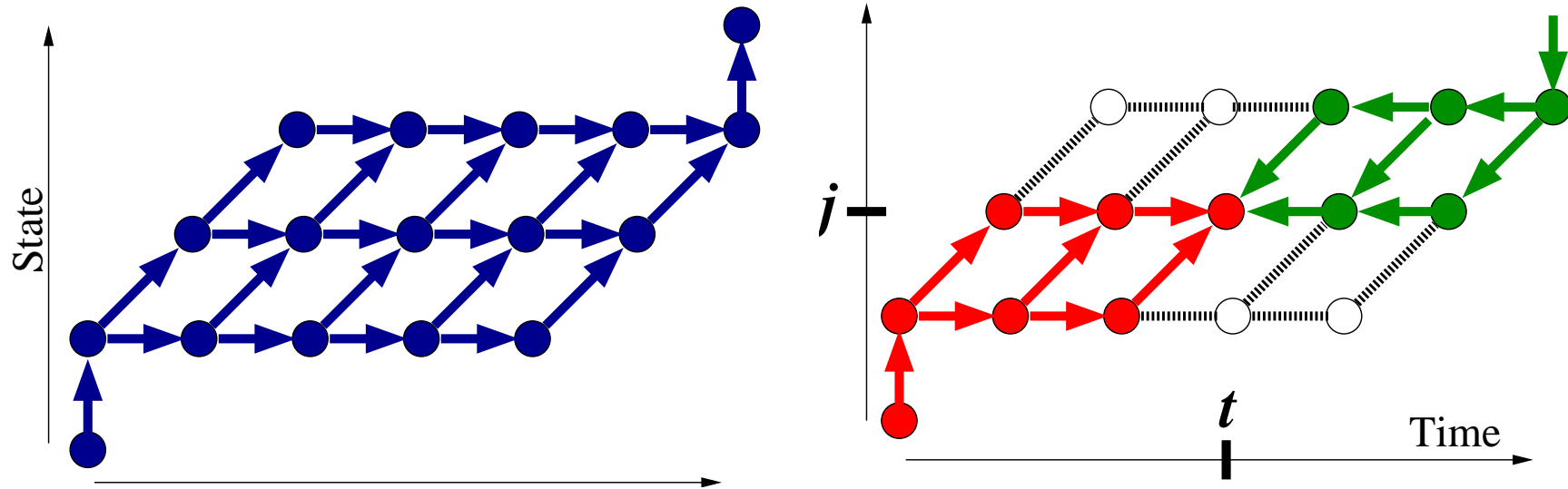
- MPE training for the log-linear model parameters
 - derivative score-spaces give large gains over (ML VTS) baseline
- Component tying important for heavily tied α (47 monophone classes)



Efficient Feature Extraction



Standard HMM Algorithms

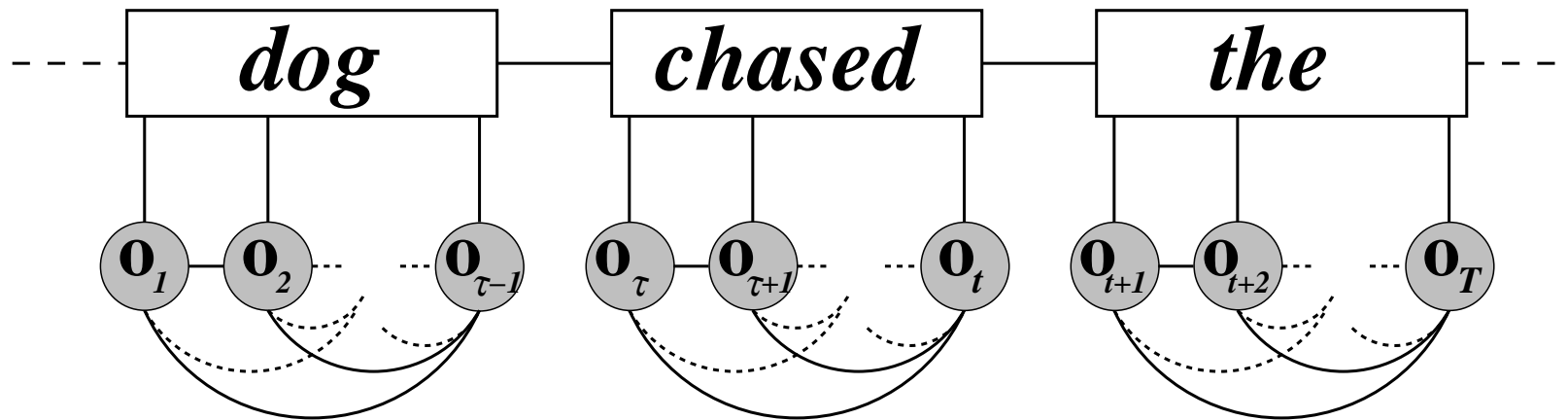


- Efficient training and inference
 - based on forward-backward/Viterbi algorithms

$$\gamma_t^{(j)} = P(q_t^{(j)} | \mathbf{O}_{1:T}; \lambda) = \frac{1}{p(\mathbf{O}_{1:T}; \lambda)} \cdot p(\mathbf{O}_{1:t}, q_t^{(j)}; \lambda) \cdot p(\mathbf{O}_{t+1:T} | q_t^{(j)}; \lambda)$$

- time/memory requirement $\mathcal{O}(T) + \mathcal{O}(T)$

Structured Discriminative Models

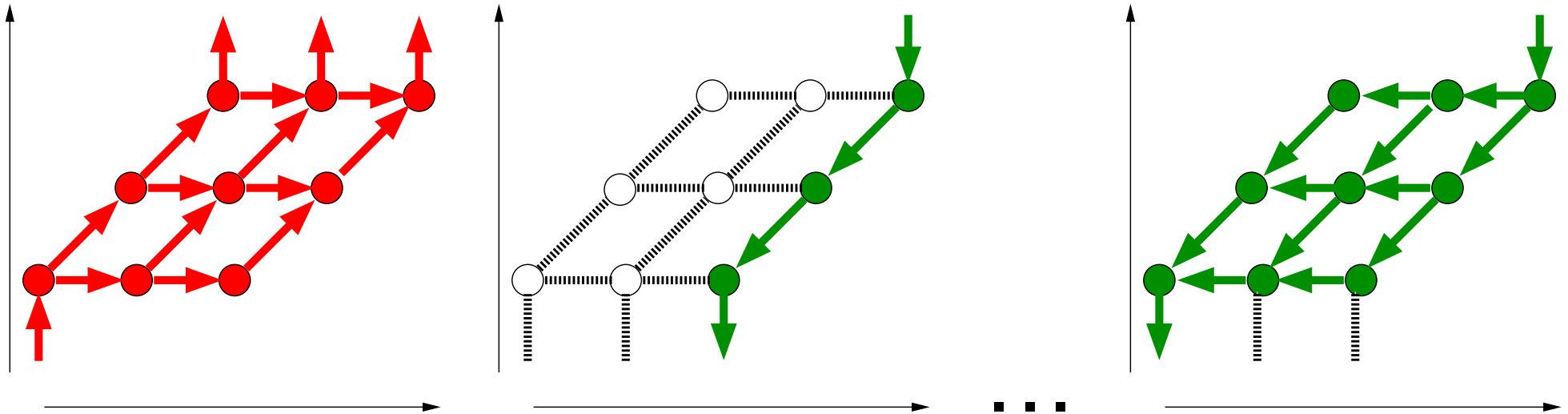


- Relate speech segments to words [17, 18, 19]

$$P(\mathbf{w}_{1:L} | \mathbf{O}_{1:T}; \boldsymbol{\alpha}) = \frac{1}{Z} \sum_{\mathbf{a}} \exp \left(\boldsymbol{\alpha}^T \sum_{\tau=1}^{|\mathbf{a}|} \phi(\mathbf{O}_{\{a_{\tau}\}}, a_{\tau}^i) \right)$$

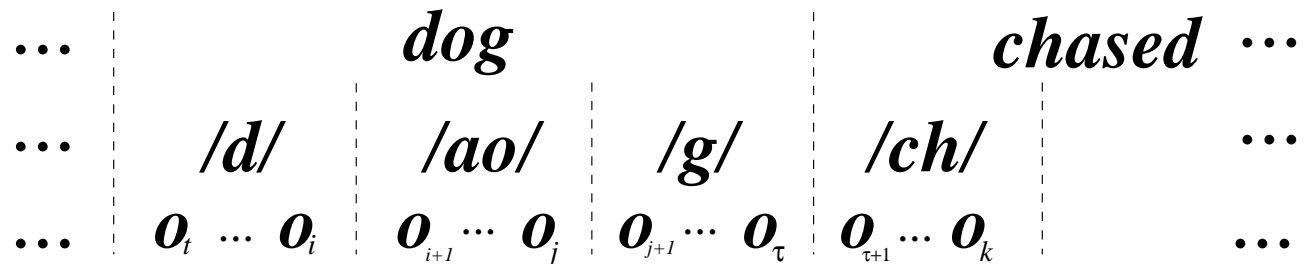
- alignment unknown marginalised over in training (or 1-best taken)
- Features extracted from variable length observation sequence $\mathbf{O}_{\{a_{\tau}\}}$
 - need to use **sequence kernel or score-space**

Forward/Backward Caching



- Cache all state-level forward probabilities – $\mathcal{O}(T)$ forward passes
- For each of the possible $\mathcal{O}(T)$ start-times
 - compute backward probabilities – $\mathcal{O}(T)$ possible backward passes
 - intersect of forward/backward yields required posterior
- **BUT** need to accumulate statistics for each start/end time – total $\mathcal{O}(T^3)$

Segmentation



- Segmentation can be viewed at multiple levels
 - **sentence**: yields flat direct model - standard problems
 - **word**: easy implementation for small vocab, sparsity issues
 - **phone**: may be context-dependent
 - **state**: very flexible, but large number of segments
- Multiple levels of segmentation can be used/combined
 - multiple segmentations can be used to derive features
- Training/inference either marginalise or pick best segmentation



Approximate Training/Inference Schemes

- If HMMs are being used anyway - use for segmentation $\mathcal{O}(T)$
 - simplest approach use Viterbi (1-best) segmentation from HMM, $\hat{\mathbf{a}}_{\text{hmm}}$
 - use fixed segmentation in training and test - highly efficient

$$P(\mathbf{w}|\mathbf{O}) \approx \frac{1}{Z} \prod_{\tau=1}^{|\hat{\mathbf{a}}_{\text{hmm}}|} \exp(\boldsymbol{\alpha}^T \phi(\mathbf{O}_{\{\hat{\mathbf{a}}_{\text{hmm}\tau}\}}, \hat{\mathbf{a}}_{\text{hmm}\tau}^i))$$

$$\hat{\mathbf{a}}_{\text{hmm}} = \underset{\mathbf{a}}{\operatorname{argmax}} \{p(\mathbf{O}|\mathbf{a}, \boldsymbol{\lambda})P(\mathbf{a})\}$$

- Assumption: segmentation not dependent on discriminative model parameters
 - unclear how accurate/appropriate this is for ASR
- Efficient inference feature extraction will be described later [14]

