



Segmentally Boosted HMM

Computational Perception Lab

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Goal

Discriminative feature selection for hidden Markov Models in **sequence classification**.

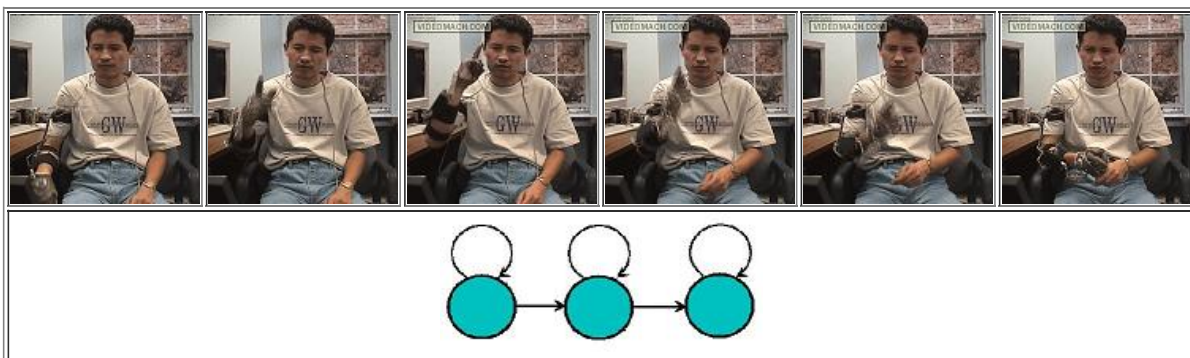
Introduction

Speech recognition, gesture recognition, DNA analysis and many other pattern recognition tasks for time-series data are sequence classification problems, which predict of a **single** label for an entire sequence. The most successful technique for sequence classification is the hidden Markov Model (HMM). The recognition accuracy and efficiency of HMMs can be improved with discriminative features. Traditional feature selection methods require the data being independently and identically distributed (i.i.d.), but time sequences usually contain strong temporal correlation between the adjacent observation frames. Furthermore, features in time sequences may be "sometimes informative", that is, discriminative only in some segments of a sequence. In this research, we propose Segmentally-Boosted HMMs (SBHMMs), which is able to address the problems of both temporal correlation and segmentally-informative features by assuming "piecewise i.i.d." Experiments show that the SBHMMs consistently improve traditional HMM recognition in American Sign Language recognition, human gait identification, lip reading and speech recognition. The reduction of error ranges from 17% to 70%.

Conditional Random Fields [Lafferty, et.al., 2001] or Tandem models [Hermansky, et.al., 2000] require a state-level labeling by human or forced alignment. In practice, such labeling may not be possible. For example, to recognize the sign brother, how can the human labeler precisely supervise the training for the first state when he does not even know the states meaning or how many states comprise the sign? Without such labeling, the discriminative feature selection will be limited to the level of the entire signs or phonemes. In contrast, SBHMMs are truly designed for sequence classification, in which the sub-sequence components are **unknown**, and our experiments show that the feature selection at the sub-sequence (state) level achieves superior performance than that at the sequence level.

Approach

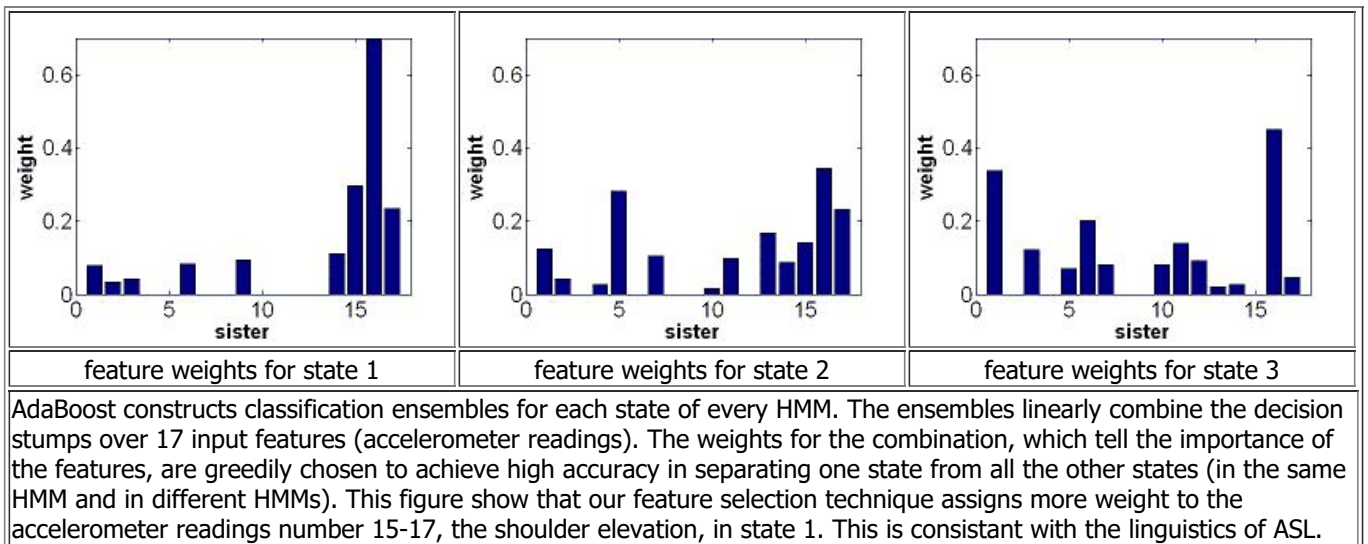
1. Train HMMs for input sequences (the example below is the sign "sister" in ASL).



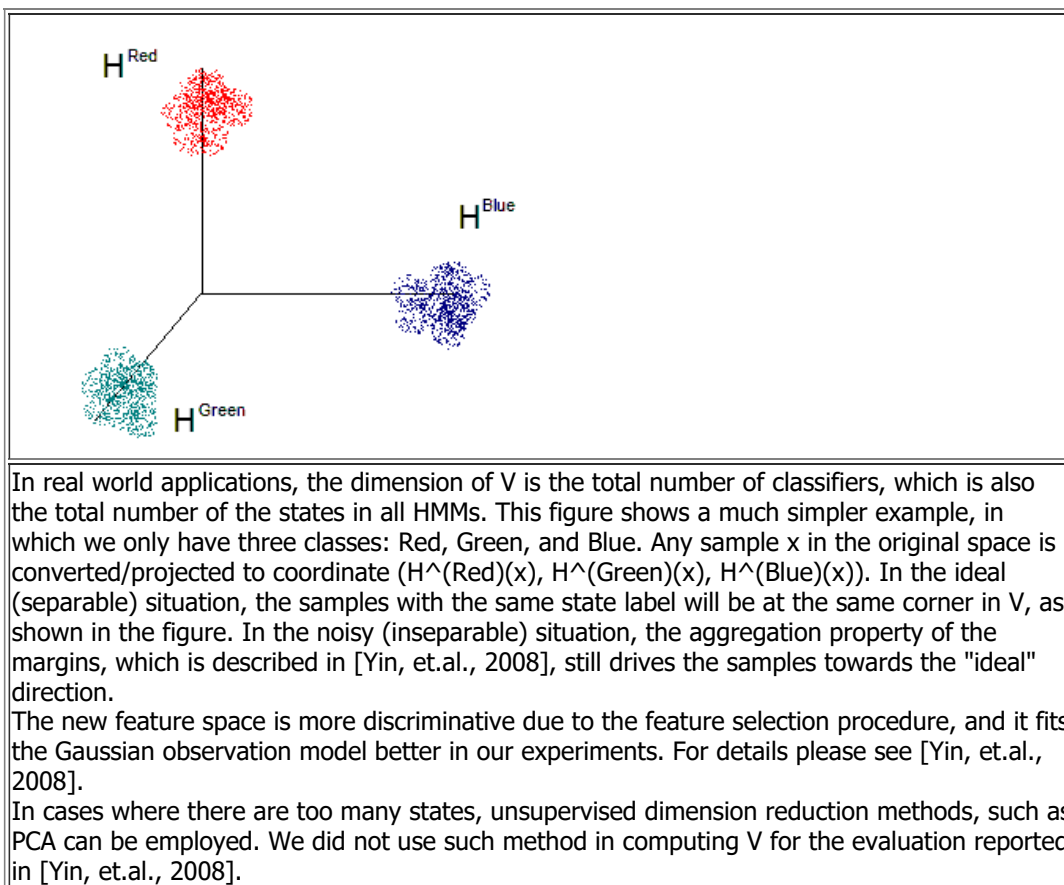
2. Label each frame automatically with its most likely state computed from the Viterbi decoding.



3. Train AdaBoost for this labeling.



4. The new feature space V is the output space of the AdaBoost ensembles.



5. Train new HMMs in V. Those HMMs have higher accuracy due to the discriminative features.

Data

- **Georgia Tech Speech Reading Data**

Description: Continuous audio-visual speech recognition data, audio captured by one microphone at 16kHz and visual markers captured by Motion Capture devices at 120Hz.

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Total Length	30m45s	MoCap Rate	120Hz
Training Data	24m42s	Testing Data	06m03s
Total Sentences	275	Total Phones	8468
Total Phonemes	39	Total Samples	> 200,000

Download: Compressed file (84MB): [gtsr.rar](#)

- **Georgia Tech Gait Recognition Data**

Description: Gait identification data captured by Motion Capture (MoCap) devices at 120Hz.

It contains 22 tracker readings for 15 subjects. We used three leg-related readings of the first five subjects following the convention in [Kim and Pavlovic, 2006]

Download: [link](#) (via FTP)

- **MIT American Sign Language Recognition Data**

Description: Continuous ASL data captured by video cameras mounted on the hat.

It contains 500 five-sign sentences composed of 40 different signs by one subject.

Download: [sterner97.zip](#) from [Contextual Computing Group](#) (CCG) at Georgia Tech

- **Georgia Tech American Sign Language Recognition Data**

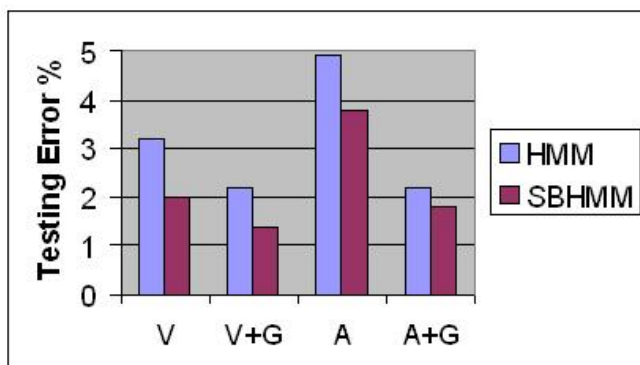
Description: Continuous ASL data captured by accelerometers on the gloves

It contains 665 four-sign sentences composed of 141 different signs by one subject.

Download: [acceleglove.zip](#) from [Contextual Computing Group](#) (CCG) at Georgia Tech.

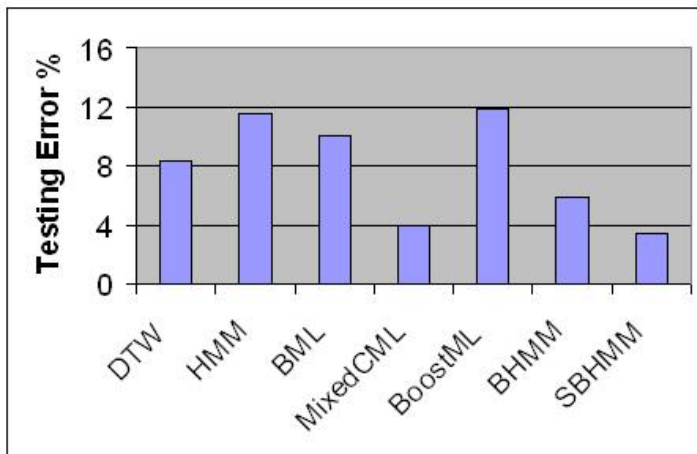
Experimental Results

- **American Sign Language Recognition on MIT data and Georgia Tech data**



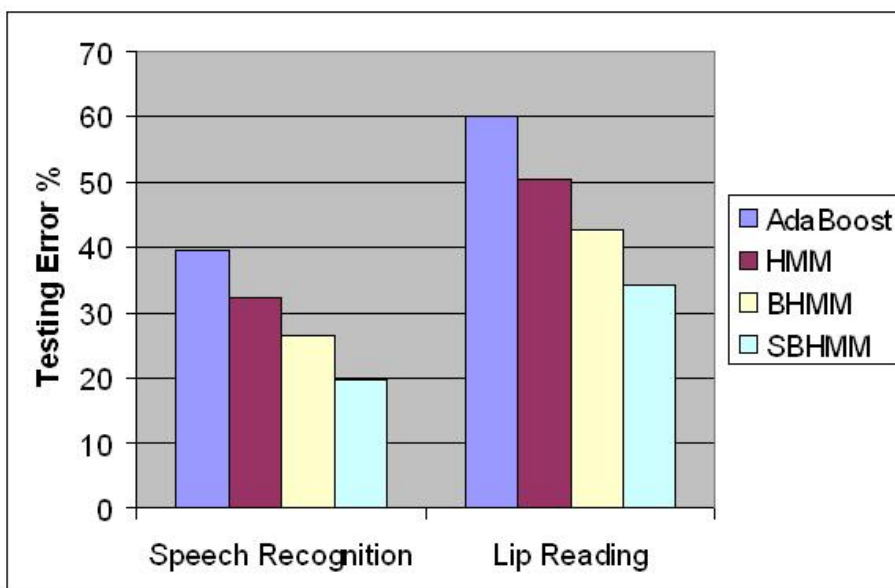
V: vision-based MIT data, A: accelerometer-based Georgia Tech data, and G: using grammar for postprocessing.

- **Georgia Tech Gait Recognition data**



The performance for DTW, HMM, BML, MixedCML, BoostedML are directly from [Kim and Pavlovic, 2006]. BHMM is boosted HMM in [Yin, et.al., 2004]. Please refer to these two papers for the details of the data and the experiments conducted.

- **Georgia Tech Audio-Visual Speech Recognition Data**



In this chart, we report the testing error for phoneme recognition. We only use audio information in speech recognition and visual information in lip reading. Note that audio-visual fusion may further reduce the recognition error as in [Yin, et.al., 2003]. For the details of the data and the experiments conducted, please refer to [Yin, et.al., 2004] and [Yin, et.al., 2008].

Publications

Pei Yin, Irfan Essa, Thad Starner, James M. Rehg, "Discriminative Feature Selection for Hidden Markov Models Using Segmental Boosting", in Proc. of IEEE International Conference on Acoustics, Speech, and Signal Processing (ICASSP 2008), Mar. 2008. ([pdf](#)) ([bibtex](#)).

Pei Yin, Irfan Essa, James M. Rehg, "Segmental Boosting Algorithm for Time-series Analysis," in Snowbird Machine Learning Workshop, Mar. 2007.

Pei Yin, Irfan Essa, James M. Rehg, "Asymmetrically Boosted HMM for Speech Reading," in Proc. of *IEEE Conference on Computer Vision and Pattern Recognition (CVPR 2004)*, pp II755-761, June 2004 ([pdf](#)) ([bibtex](#))

Pei Yin, Irfan Essa, James M. Rehg, "Boosted Audio-Visual HMM for Speech Reading," in Proc. of *IEEE International Workshop on Analysis and Modeling of Faces and Gestures (AMFG)*, pp 68-73, Oct. 2003/held in conjunction with *ICCV-2003*. A version of this paper also appears in Proc. of *Asilomar Conference on Signals, Systems, and Computers*, pp 2013-2018, Nov. 2003 as an invited paper. ([pdf](#)) ([bibtex](#))

Acknowledgements

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Code Download

Integrated with [HTK](#)

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