EE E6820: Speech & Audio Processing & Recognition

### Lecture 11: Signal Separation

Sound Mixture Organization

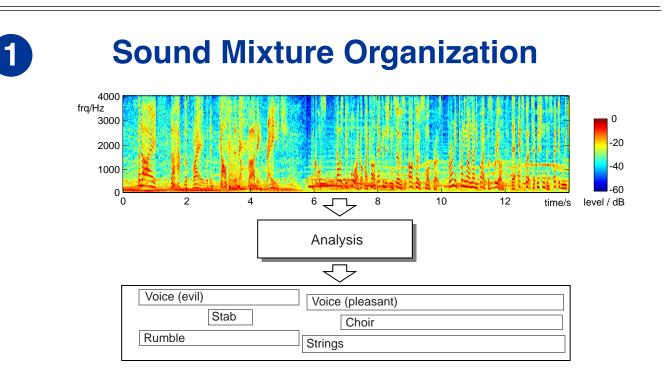
- **2** Computational Auditory Scene Analysis
- **3** Independent Component Analysis
  - Model-Based Separation

Dan Ellis <dpwe@ee.columbia.edu> http://www.ee.columbia.edu/~dpwe/e6820/

Columbia University Dept. of Electrical Engineering Spring 2006



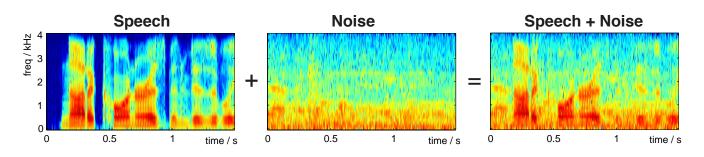
L11 - Signal Separation



- Auditory Scene Analysis: describing a complex sound in terms of high-level sources / events
  - ... like listeners do
- Hearing is ecologically grounded
  - reflects 'natural scene' properties
  - subjective, not absolute



## Sound, mixtures, and learning



- Sound
  - carries useful information about the world
  - complements vision

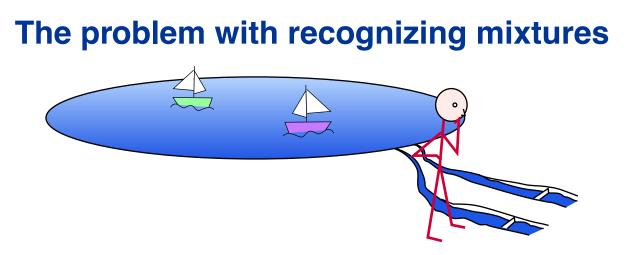
#### Mixtures

- .. are the rule, not the exception
- medium is 'transparent', sources are many
- must be handled!

#### • Learning

- the 'speech recognition' lesson: let the data do the work
- like listeners





"Imagine two narrow channels dug up from the edge of a lake, with handkerchiefs stretched across each one. Looking only at the motion of the handkerchiefs, you are to answer questions such as: How many boats are there on the lake and where are they?" (after Bregman'90)

#### Received waveform is a mixture

- two sensors, N signals ... underconstrained
- Disentangling mixtures as the primary goal?
  - perfect solution is not possible
  - need experience-based constraints

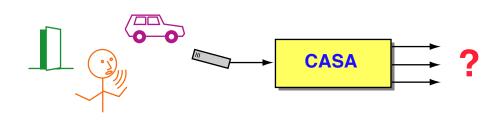


### **Approaches to sound mixture recognition**

- Separate signals, then recognize
  - e.g. Computational Auditory Scene Analysis (CASA), Independent Component Analysis (ICA)
  - nice, if you can do it
- Recognize combined signal
  - 'multicondition training'
  - combinatorics..
- Recognize with parallel models
  - full joint-state space?
  - divide signal into fragments, then use missing-data recognition



### What is the goal of CASA?



- Separate signals?
  - output is unmixed waveforms
  - underconstrained, very hard ...
  - too hard? not required?
- Source classification?
  - output is set of event-names
  - listeners do more than this...
- Something in-between?
   Identify independent sources + characteristics
  - standard task, results?



## **Segregation vs. Inference**

#### Source separation requires attribute separation

- sources are characterized by attributes (pitch, loudness, timbre + finer details)
- need to identify & gather different attributes for different sources ...
- Need representation that segregates attributes
  - spectral decomposition
  - periodicity decomposition

#### Sometimes values can't be separated

- e.g. unvoiced speech
- maybe infer factors from probabilistic model?

 $p(O, x, y) \rightarrow p(x, y | O)$ 

 or: just skip those values, infer from higher-level context



# Outline





#### **2** Computational Auditory Scene Analysis

- Human Auditory Scene Analysis
- Bottom-up and Top-down models
- Evaluation

(3) **Independent Component Analysis** 

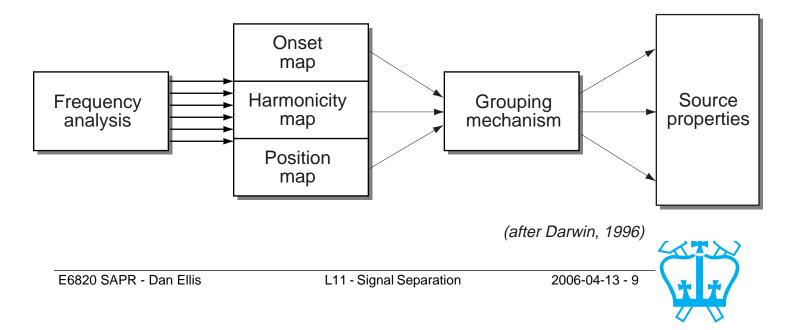
**Model-Based Separation** 



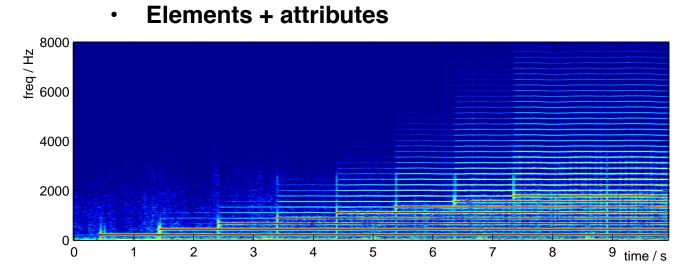


### Auditory Scene Analysis (Bregman 1990)

- How do people analyze sound mixtures?
  - break mixture into small elements (in time-freq)
  - elements are grouped in to sources using cues
  - sources have aggregate attributes
- Grouping 'rules' (Darwin, Carlyon, ...):
  - cues: common onset/offset/modulation, harmonicity, spatial location, ...



## **Cues to simultaneous grouping**



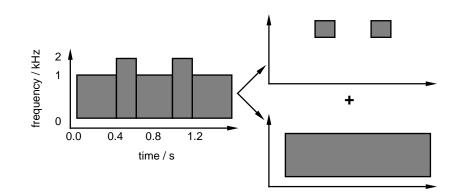
#### Common onset

- simultaneous energy has common source
- Periodicity
  - energy in different bands with same cycle
- Other cues
  - spatial (ITD/IID), familiarity, ...



## The effect of context

- Context can create an 'expectation': i.e. a bias towards a particular interpretation
  - e.g. Bregman's "old-plus-new" principle: A change in a signal will be interpreted as an added source whenever possible



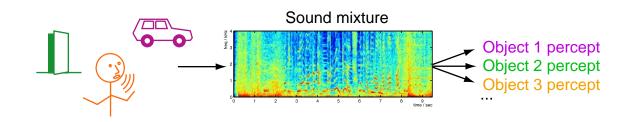
- a different division of the same energy depending on what preceded it



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### Computational Auditory Scene Analysis (CASA)



- Goal: Automatic sound organization; Systems to 'pick out' sounds in a mixture
  - ... like people do

#### • E.g. voice against a noisy background

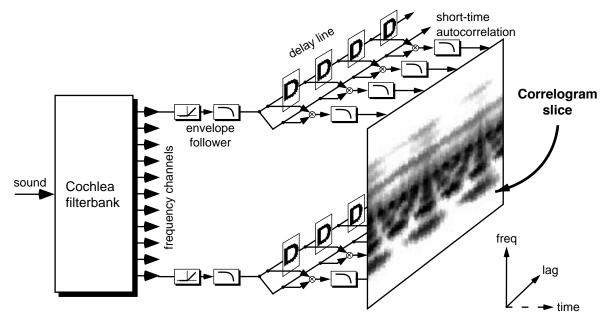
- to improve speech recognition
- Approach:
  - psychoacoustics describes grouping 'rules'
  - ... just implement them?



## **CASA front-end processing**

• Correlogram:

Loosely based on known/possible physiology

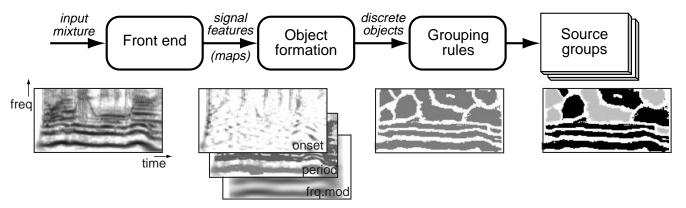


- linear filterbank cochlear approximation
- static nonlinearity
- zero-delay slice is like spectrogram
- periodicity from delay-and-multiply detectors

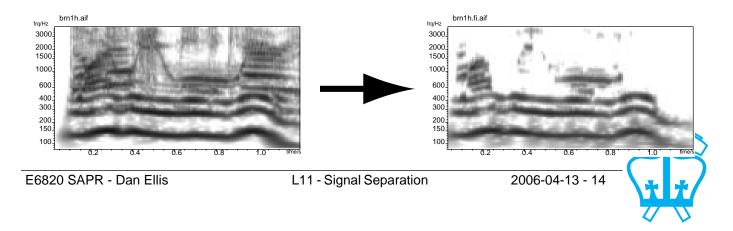


### The Representational Approach (Brown & Cooke 1993)

Implement psychoacoustic theory

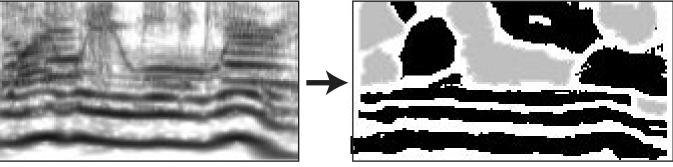


- 'bottom-up' processing
- uses common onset & periodicity cues
- Able to extract voiced speech:



### **Problems with 'bottom-up' CASA**





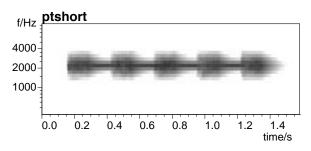
time -

- Circumscribing time-frequency elements
  - need to have 'regions', but hard to find
- Periodicity is the primary cue
  - how to handle aperiodic energy?
- Resynthesis via masked filtering
  - cannot separate within a single t-f element
- Bottom-up leaves no ambiguity or context
  - how to model illusions?

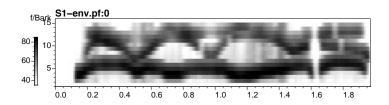


## **Restoration in sound perception**

- Auditory 'illusions' = hearing what's not there
- The continuity illusion



• Sinewave Speech (SWS)



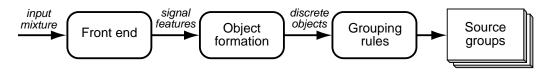
- duplex perception
- What kind of model accounts for this?
  - is it an important part of hearing?



### Adding top-down constraints: Prediction-Driven CASA (PDCASA)

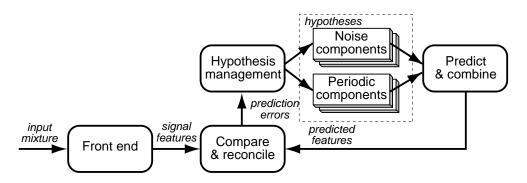
Perception is not direct but a search for plausible hypotheses

Data-driven (bottom-up)...



- objects irresistibly appear

### vs. Prediction-driven (top-down)

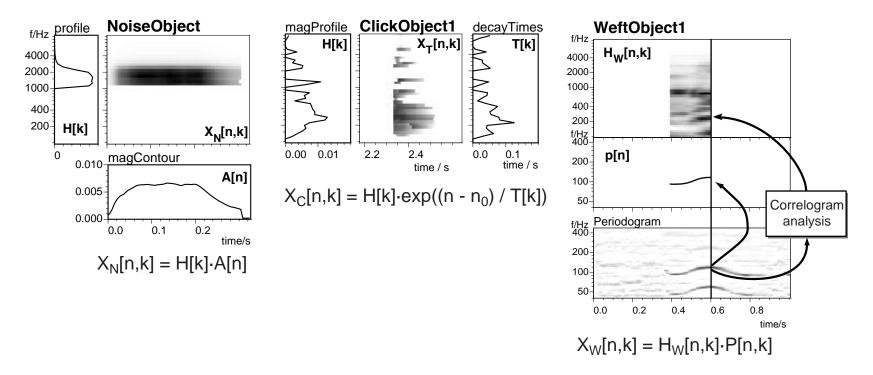


- match observations with a 'world-model'
- need world-model constraints...



### **Generic sound elements for PDCASA**

- Goal is a representational space that
  - covers real-world perceptual sounds
  - minimal parameterization (sparseness)
  - separate attributes in separate parameters

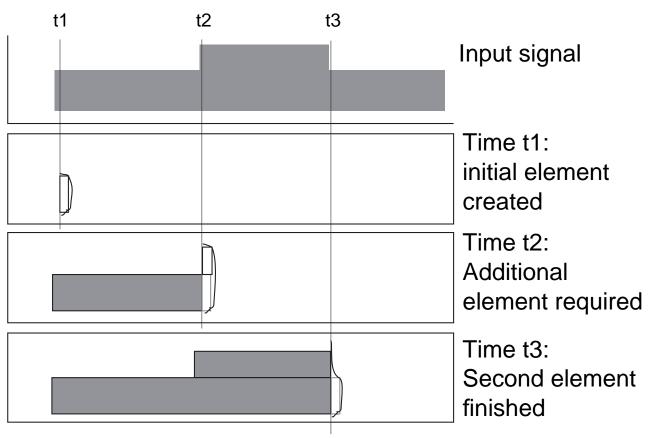


Object hierarchies built on top...



### **PDCASA** for old-plus-new

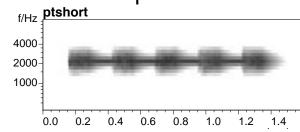




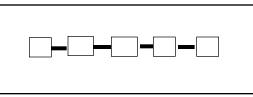


### **PDCASA** for the continuity illusion

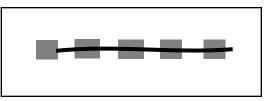
- Subjects hear the tone as continuous
  - ... if the noise is a plausible masker



• Data-driven analysis gives just visible portions:



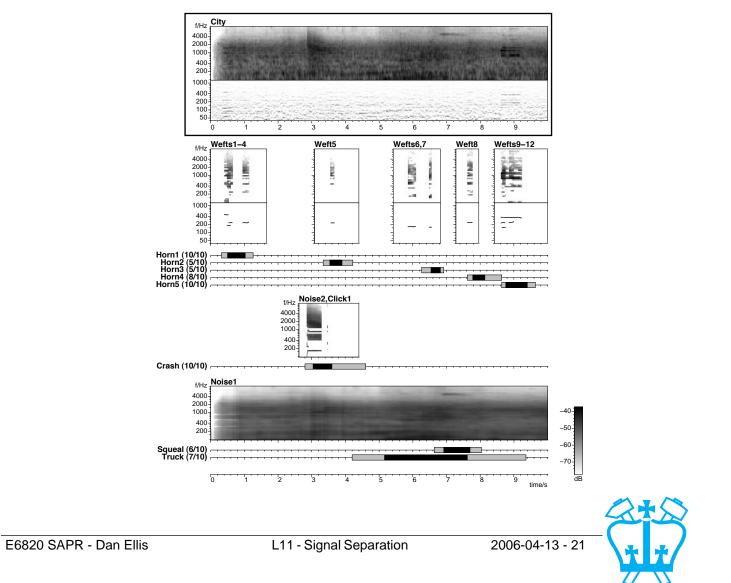
• Prediction-driven can infer masking:





### **Prediction-Driven CASA**

Explain a complex sound with basic elements



### **Aside: Ground Truth**

- What do people hear in sound mixtures?
  - do interpretations match?
- → Listening tests to collect 'perceived events':

Subject dpwe / Example city / Part A		
Names	Marks	
horn1		
crash		
squeal		
hom2		
「		
Play Stop		Go on



## **Aside: Evaluation**

- Evaluation is a big problem for CASA
  - what is the goal, really?
  - what is a good test domain?
  - how do you measure performance?

#### • SNR improvement

- tricky to derive from before/after signals: correspondence problem
- can do with fixed filtering mask; but rewards removing signal as well as noise

#### • Speech Recognition (ASR) improvement

- recognizers typically very sensitive to artefacts
- 'Real' task?
  - mixture corpus with specific sound events...



# Outline

- **1** Sound Mixture Organization
- **2** Computational Auditory Scene Analysis

### **3** Independent Component Analysis

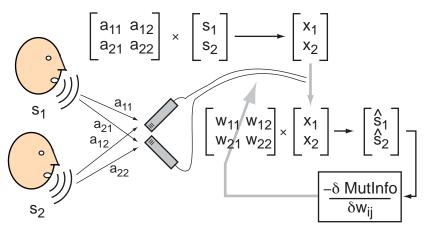
- Blind source separation
- Independence and kurtosis
- Limits of the approach
- 4 Model-Based Separation



# **3 Independent Component Analysis (ICA)**

(Bell & Sejnowski 1995 etc.)

• If mixing is like matrix multiplication, then separation is searching for the inverse matrix



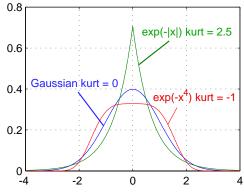
- i.e.  $W \approx A^{-1}$
- with *N* different versions of the mixed signals (microphones), we can find *N* different input contributions (sources)
- how to rate quality of outputs?
   i.e. when do outputs look separate?



### **Gaussianity, Kurtosis & Independence**

- A signal can be characterized by its PDF p(x)
  - i.e. as if successive time values are drawn from a random variable (RV)
  - Gaussian PDF is 'least interesting'
  - Sums of independent RVs (PDFs convolved) tend to Gaussian PDF (Weak law of large nums)
- Measures of deviations from Gaussianity: 4th moment is Kurtosis ("bulging")

$$kurt(y) = E\left[\left(\frac{y-\mu}{\sigma}\right)^4\right] - 3$$



-kurtosis of Gaussian is zero (this def.)

-'heavy tails'  $\rightarrow kurt > 0$ 

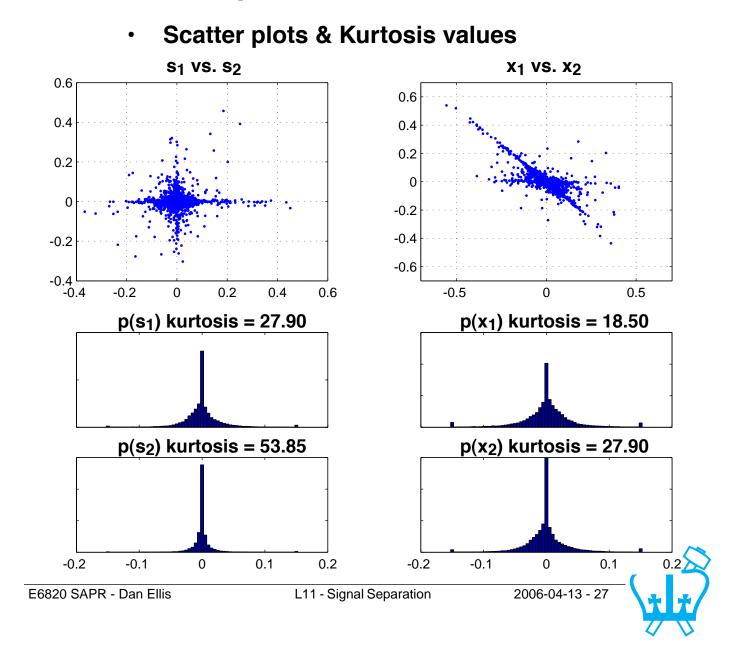
-closer to uniform dist.  $\rightarrow kurt < 0$ 

#### •Directly related to KL divergence from Gaussian PDF

E6820 SAPR - Dan Ellis



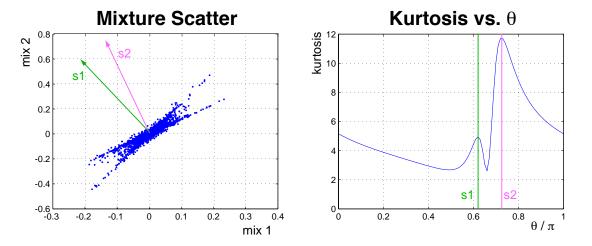
### **Independence in Mixtures**



### **Finding Independent Components**

Sums of independent RVs are more Gaussian
 → minimize Gaussianity to undo sums

- i.e. search over w<sub>ij</sub> terms in inverse matrix



• Solve by Gradient descent or Newton-Raphson:

$$\mathbf{w}^{+} = E\{\mathbf{x}g(\mathbf{w}^{T}\mathbf{x})\} - E\{g'(\mathbf{w}^{T}\mathbf{x})\}\mathbf{w}$$
  
$$\mathbf{w} = \mathbf{w}^{+}/||\mathbf{w}^{+}||$$

"Fast ICA", http://www.cis.hut.fi/projects/ica/fastica/

L11 - Signal Separation

# Limitations of ICA

- Assumes instantaneous mixing
  - real world mixtures have delays & reflections
  - STFT domain?

 $\begin{aligned} x_1(t) &= a_{11}(t) \otimes s_1(t) + a_{12}(t) \otimes s_2(t) \\ \Rightarrow X_1(\omega) &= A_{11}(\omega) S_1(\omega) + A_{12}(\omega) S_2(\omega) \end{aligned}$ 

Solve  $\boldsymbol{\omega}$  subbands separately, match up answers

- Searching for best possible inverse matrix
  - cannot find more than N outputs from N inputs but: "projection pursuit" ideas

     time-frequency masking...
- Cancellation inherently fragile
  - $\hat{s}_1 = w_{11} \cdot x_1 + w_{12} \cdot x_2$  to cancel out  $s_2$
  - sensitive to noise in *x* channels
  - time-varying mixtures are a problem



# Outline

- **1** Sound Mixture Organization
- **2** Computational Auditory Scene Analysis
- **3** Independent Component Analysis

### 4 Model-Based Separation

- Fitting models to mixtures
- Missing-data recognition
- Speech Fragment Decoding

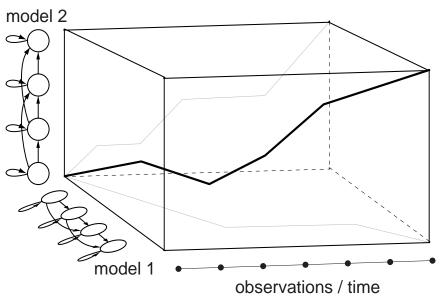




## Model-Based Separation: HMM decomposition

(e.g. Varga & Moore 1991, Gales & Young 1996)

Independent state sequences
 for 2+ component source models



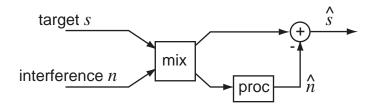
• New combined state space  $q' = \{q_1 q_2\}$ 

- need pdfs for combinations  $p(X|q_1, q_2)$ 

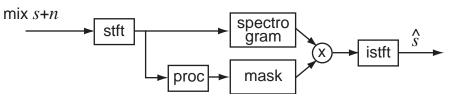


### **One-channel Separation: Masked Filtering**

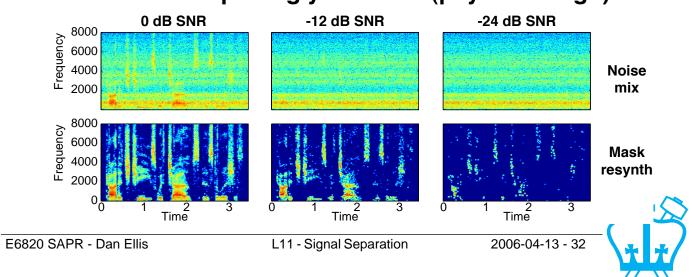
• Multichannel → ICA: Inverse filter & cancel



• One channel: find a time-frequency mask



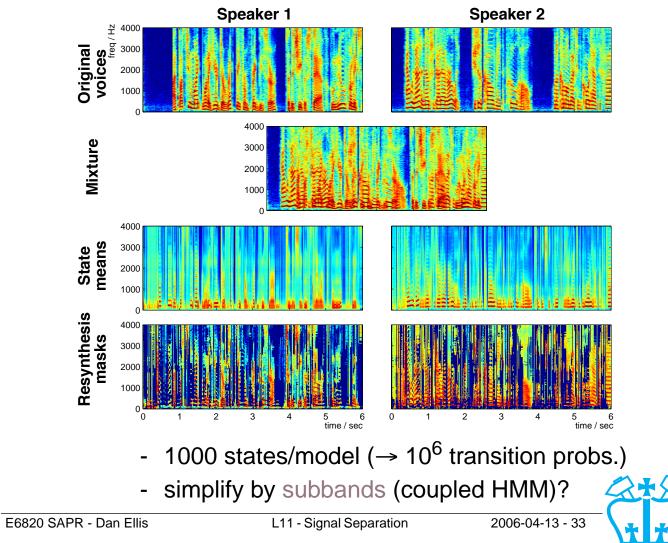
• Cannot remove overlapping noise in TF cells, but surprisingly effective (psy. masking?):



## "One microphone source separation"

(Roweis 2000, Manuel Reyes)

State sequences  $\rightarrow$  t-f estimates  $\rightarrow$  mask ٠



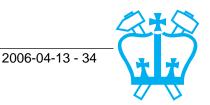
## **Speech Fragment Recognition**

(Jon Barker & Martin Cooke, Sheffield)

- Signal separation is too hard! Instead:
  - segregate features into partially-observed sources
  - then classify
- Made possible by missing data recognition
  - integrate over uncertainty in observations for true posterior distribution
- Goal:

Relate clean speech models *P*(*X*|*M*) to speech-plus-noise mixture observations

- .. and make it tractable



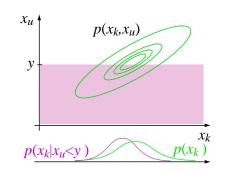
### **Missing Data Recognition**

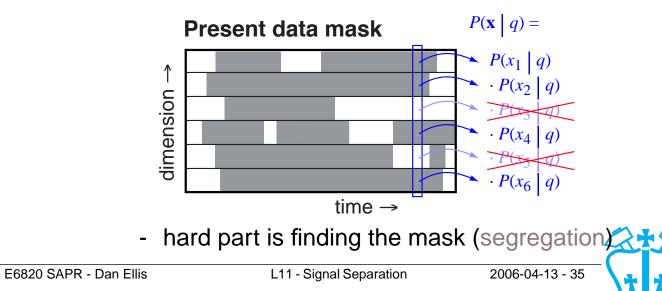
- Speech models  $p(\mathbf{x}|m)$  are multidimensional...
  - i.e. means, variances for every freq. channel
  - need values for all dimensions to get  $p(\bullet)$
- But: can evaluate over a subset of dimensions  $x_k$

$$p(\mathbf{x}_k|m) = \int p(\mathbf{x}_k, \mathbf{x}_u|m) d\mathbf{x}_u$$

•

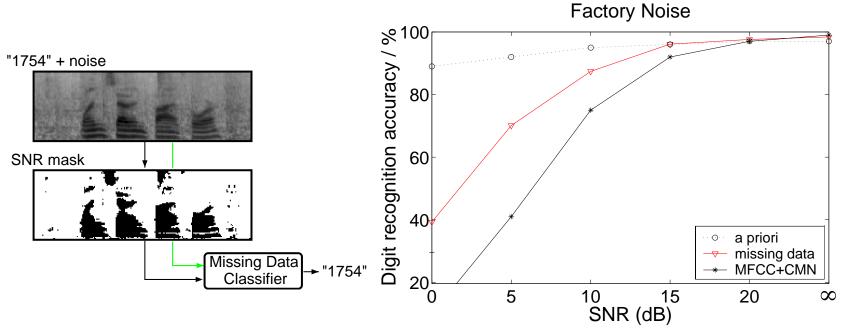
Hence, missing data recognition:





## **Missing Data Results**

- Estimate static background noise level *N*(*f*)
- Cells with energy close to background are considered "missing"



- must use spectral features!
- But: nonstationary noise  $\rightarrow$  spurious mask bits
  - can we try removing parts of mask?

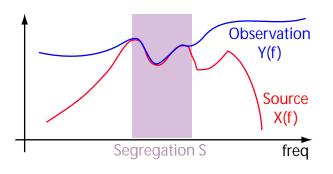


### **Comparing different segregations**

• Standard classification chooses between models *M* to match source features *X* 

$$M^* = \underset{M}{\operatorname{argmax}} P(M|X) = \underset{M}{\operatorname{argmax}} P(X|M) \cdot \frac{P(M)}{P(X)}$$

• Mixtures: observed features *Y*, segregation *S*, all related by P(X|Y, S)



Joint classification of model and segregation:

$$P(M, S|Y) = P(M) \int P(X|M) \cdot \frac{P(X|Y, S)}{P(X)} dX \cdot P(S|Y)$$

- P(X) no longer constant



### **Calculating fragment matches**

 $P(M, S|Y) = P(M) \int P(X|M) \cdot \frac{P(X|Y, S)}{P(X)} dX \cdot \frac{P(S|Y)}{P(X)}$ 

- P(X|M) the clean-signal feature model
- P(X|Y,S)/P(X) is X 'visible' given segregation?
- Integration collapses some bands...
- P(S|Y) segregation inferred from observation
  - just assume uniform, find S for most likely M
  - or: use extra information in *Y* to distinguish *S*'s...
- Result:
  - probabilistically-correct relation between clean-source models *P*(*X*|*M*) and inferred, recognized source + segregation *P*(*M*,*S*|*Y*)



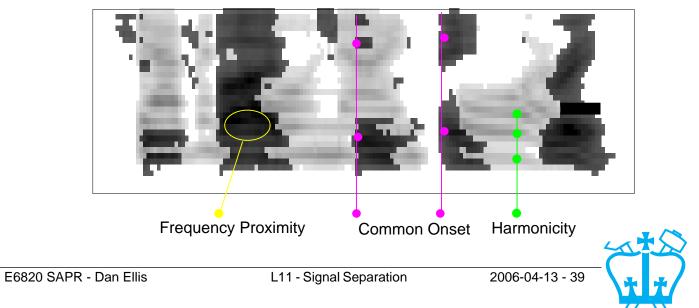
## **Using CASA features**

- P(S|Y) links acoustic information to segregation
  - is this segregation worth considering?
  - how likely is it?
- Opportunity for CASA-style information to contribute
  - periodicity/harmonicity:

these different frequency bands belong together

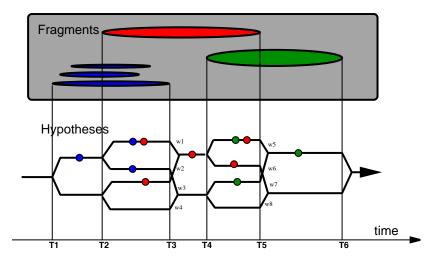
- onset/continuity:

this time-frequency region must be whole



## **Fragment decoding**

• Limiting *S* to whole fragments makes hypothesis search tractable:

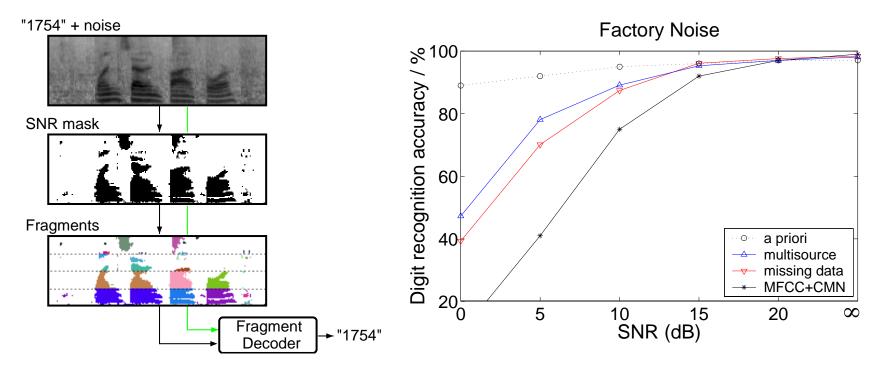


- choice of fragments reflects P(S|Y) P(X|M)
   i.e. best combination of segregation
   and match to speech models
- Merging hypotheses limits space demands
  - .. but erases specific history



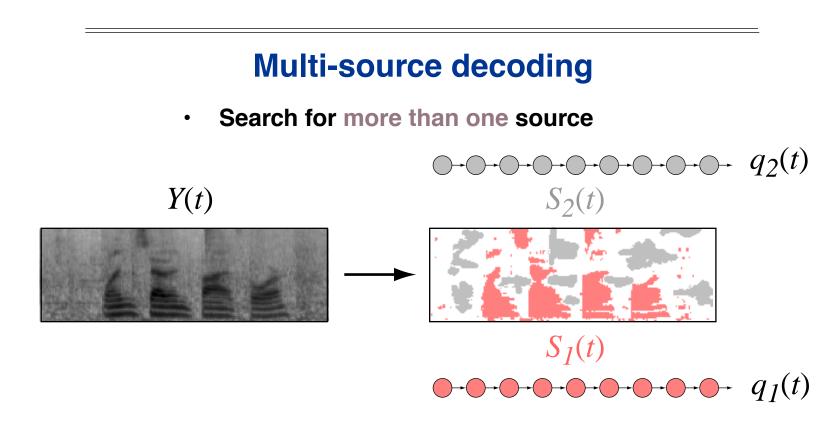
### **Speech fragment decoder results**

- Simple *P*(*S*|*Y*) model forces contiguous regions to stay together
  - big efficiency gain when searching S space



Clean-models-based recognition
 rivals trained-in-noise recognition





- Mutually-dependent data masks
  - disjoint subsets of cells for each source
  - each model match  $P(M_X|S_X,Y)$  is independent
  - masks are mutually dependent:  $P(S_1, S_2|Y)$
- Huge practical advantage over full search



## Summary

- Auditory Scene Analysis: Hearing: partially understood, very successful
- Independent Component Analysis:
   Simple and powerful, some practical limits
- Model-based separation:
   Real-world constraints, implementation tricks

Mixture separation the main obstacle in many applications e.g. soundtrack recognition



### References

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