

A Probability Model for Interaural Phase Difference

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Outline

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Probability model
Observations

Experiments
Results

Summary

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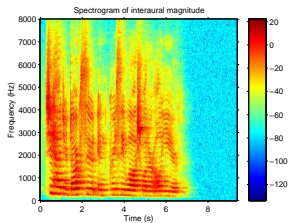
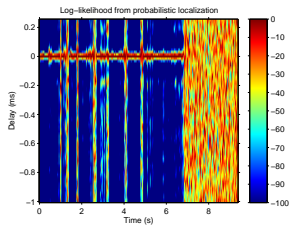
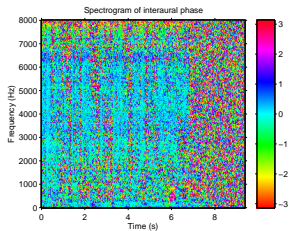
Vision

- ▶ Computer programs that can hear as well as people can
- ▶ In a room with two mics and multiple people talking
 - ▶ Locate the sound sources in space
 - ▶ Characterize the room
 - ▶ Determine what is said

Overview

- ▶ Problem: localization, time delay of arrival estimation
 - ▶ Stereo recording
 - ▶ Single source
 - ▶ Reverberation or additive noise
- ▶ Solution: probability model for interaural phase difference in reverberation
 - ▶ Considers time-frequency points individually, IID
 - ▶ No assumptions about source characteristics
- ▶ Performance: estimates delay more accurately than generalized cross-correlation in simulations

Probabilistic localization



Previous work

- ▶ Knapp and Carter (1976)
 - ▶ Generalized cross correlation and PHase Transform
 - ▶ Maximum likelihood estimator for stationary sources in uncorrelated noise
- ▶ Stern and Trahiotis (1995)
 - ▶ Based more on human perception
 - ▶ Cross-correlations in many frequency bands
 - ▶ “Straightness” measure for combining bands
- ▶ Rennie (2005)
 - ▶ EM algorithm for localizing multiple sources based on whole STFT frames

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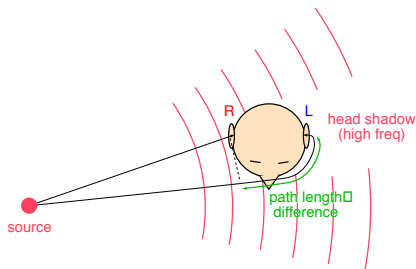
Experiments
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Data

- ▶ Speech from TIMIT database
 - ▶ Phonetically balanced sentences from 630 native English speakers
 - ▶ “She had your dark suit in greasy wash water...”
- ▶ Binaural room impulse responses from Shinn-Cunningham et al. (2005)
 - ▶ Recorded with KEMAR manikin in classroom
 - ▶ 4 locations, 7 directions, 3 distances, 3 repetitions
 - ▶ “She had your dark suit in greasy wash water...”
- ▶ Binaural anechoic impulse responses from Algazi et al. (2001)
 - ▶ Recorded with KEMAR manikin in anechoic chamber
 - ▶ 25 azimuths, 50 elevations
 - ▶ Speech-shaped noise added at 10 dB, direct to reverb ratio
 - ▶ “She had your dark suit in greasy wash water...”

Listening model



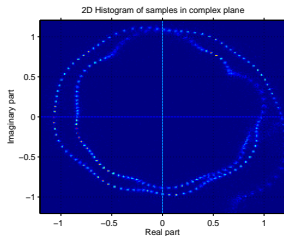
$$l(t) = a_l s(t - \tau_l) * n_l(t)$$

$$r(t) = a_r s(t - \tau_r) * n_r(t)$$

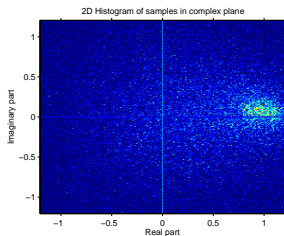
$$N(j\omega) = e^{-a + j\omega\tau} \frac{L(j\omega)}{R(j\omega)}$$

- ▶ Delay between channels because of path difference
- ▶ Gain between channels because of head shadowing
- ▶ Convolutional noise to model reverberation
- ▶ Noise assumed to be compact

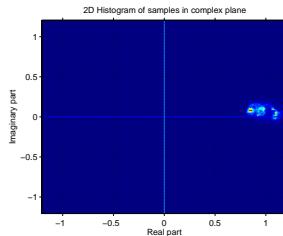
Noise distributions in the complex plane



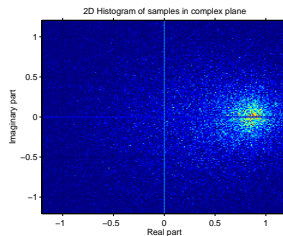
Incorrect τ



Additive

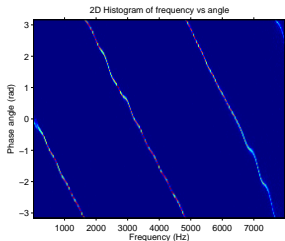


Anechoic

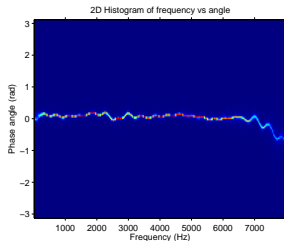


Reverberant

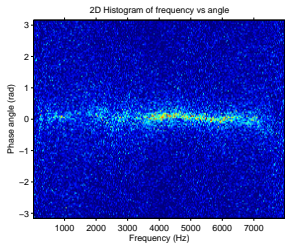
Phase of noise vs frequency



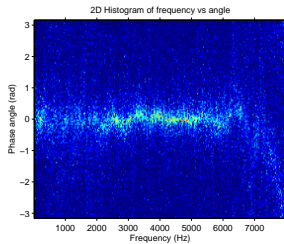
Incorrect τ



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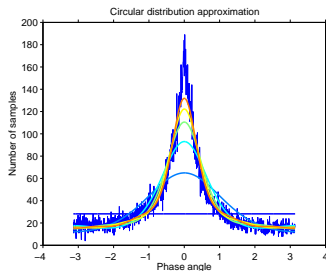


Additive



Reverberant

Probability model



$$p(\Delta\phi | \omega T) = \exp \left(\sum_k a_k \cos^k(\Delta\phi - \omega T) \right)$$

- ▶ Circular maximum entropy probability distribution
- ▶ Fourier series gives desired moments

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Experiments

- ▶ Compare PHAT to probabilistic method
- ▶ Compare sensitivity of probabilistic method to learned noise distribution
- ▶ Test in simulated reverberation and additive noise, at varying azimuths
- ▶ Pool probabilistic estimates over frequency to compare to PHAT
- ▶ Measure average error in delay estimate as a function of pooling over time

PHase Transform (PHAT)

Classical Time-Delay of Arrival (TDOA) estimator
Cross-correlation:

$$\hat{\tau} = \arg \max_{\tau} \ell(t) * r(-t)$$

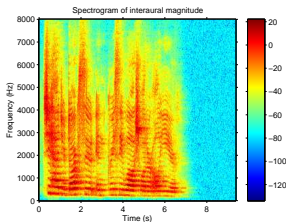
Generalized cross-correlation:

$$\hat{\tau} = \arg \max_{\tau} \sum_{\omega} \psi(j\omega) L(j\omega) R^*(j\omega) e^{j\omega\tau}$$

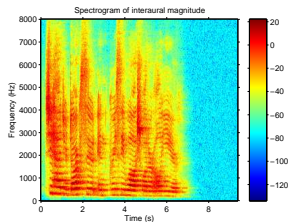
e.g. PHase Transform:

$$\hat{\tau} = \arg \max_{\tau} \sum_{\omega} \frac{1}{|L(j\omega)| |R(j\omega)|} L(j\omega) R^*(j\omega) e^{j\omega\tau}$$

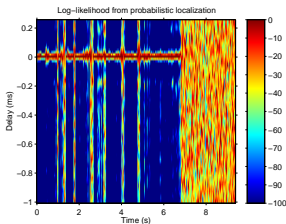
Example localizations



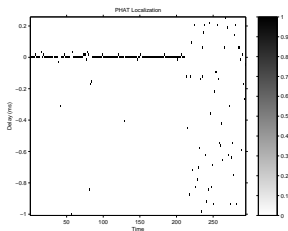
Speech signal



Speech signal

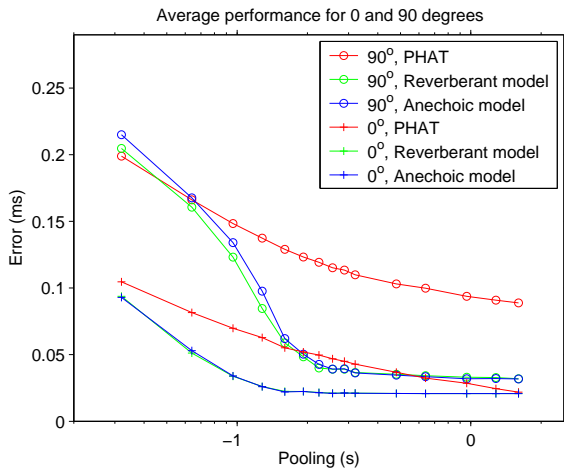


Probabilistic localization

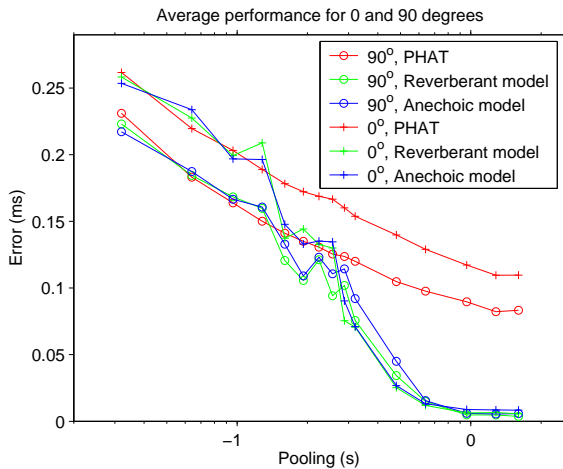


PHAT localization

Results in reverberation



Results in additive noise



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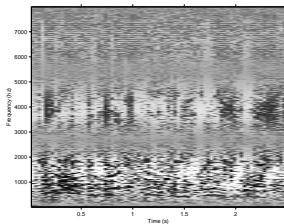
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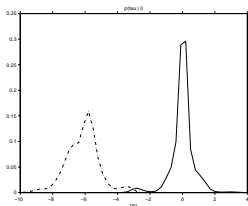
Summary

- ▶ Probabilistic model of phase at individual interaural spectrogram points
- ▶ Estimates true delays better than PHAT in reverberation
- ▶ Estimates true delays as well as, if not better than, PHAT in additive noise
- ▶ Makes no assumptions about the sources involved
- ▶ Allows powerful analyses, e.g. likelihoods of delays over arbitrary regions of spectrogram
- ▶ Lends itself to localizing multiple sources

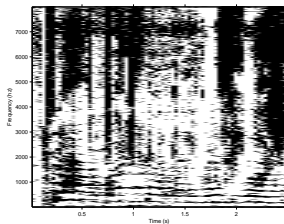
EM Algorithm



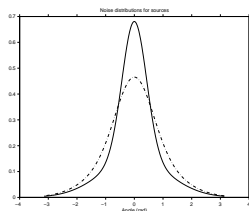
EM separation



$$p(\tau | i)$$



Ground truth



$$p(\Delta\phi - \omega\tau | i)$$

Thank you

Any questions?