Multi-Microphone Speaker Localization and Tracking on Manifolds

joint work with Bracha Laufer-Goldshtein and Ronen Talmon

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Motivation

Acoustic Source Localization & Tracking

Goal

Locate/track a sound source given measurements of the sound field



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Applications

- An essential component in speech enhancement algorithms
- 2 Camera steering
- 3 Teleconferencing
- Obot audition
- Surveillance
- Smart home/clinic/car





Room Acoustics Essentials

Acoustic propagation models

- When sound propagates in an enclosure it undergoes reflections from the room surfaces
- Reflections can be modeled as images beyond room walls and hence impinging the microphones from many directions [Allen and Berkley, 1979, Peterson, 1986]
- Statistical models for late reflections [Polack, 1993, Schroeder, 1996, Jot et al., 1997]
- Late reflections tend to be diffused, hence do not exhibit directionality [Dal-Degan and Prati, 1988,

Habets and Gannot, 2007





Speech Processing in Acoustic Environments

- Classical multi-microphone speech processing algorithms use time difference of arrival (TDOA)-only model
- Viable speech processing solutions can only be accomplished by an accurate source propagation description, captured by the acoustic impulse response (AIR)
- Describing the wave propagation of any audio source in an arbitrary acoustic environment is, however, a cumbersome task, since:
 - No simple mathematical models exist
 - The estimation of the vast number of parameters used to describe the wave propagation suffers from large errors

Data-driven approach

To alleviate these limitations and to infer a mathematical model that is accurate, simple to describe and simple to implement, we propose a data-driven approach

Background Room Acoustics and Speech Processing

How to Model the Acoustic Environment?



How to Model the Acoustic Environment?



Speaker Localization on Manifolds

How to Model the Acoustic Environment?



Speaker Localization on Manifolds

Main Claims

- The acoustic response can serve as a fingerprint for source localization
- The intrinsic degrees of freedom in acoustic responses have a limited number
- The variability of the acoustic response in specific enclosures depends only on a small number of parameters
 - \Rightarrow manifold learning approaches may improve localization ability



Outline

- Data model and Acoustic Features
- 2 The Acoustic Manifold
- 3 Data-Driven Source Localization: Microphone Pair
- 4 Data-Driven Source Localization: Ad Hoc Array
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Data Model: The Two Microphone Case

Microphone signals:

The measured signals in the two microphones:

$$y_1(n) = a_1(n) * s(n) + u_1(n)$$

 $y_2(n) = a_2(n) * s(n) + u_2(n)$

- s(n) the source signal
- $a_i(n)$, $i = \{1, 2\}$ the acoustic impulse responses relating the source and each of the microphones
- $u_i(n), i = \{1,2\}$ noise signals, independent of the source

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Find a feature vector representing the characteristics of the acoustic path (a fingerprint) and independent of the source signal!

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Data model and Acoustic Features The relative transfer function (RTF)

Relative Transfer Function (RTF) [Gannot et al., 2001]



RTF:

• Defined as the ratio between the transfer functions relating the source and the two mics:

$$H_{12}(k) = rac{A_2(k)}{A_1(k)}$$

• In the time domain: the relative impulse response (RIR) satisfies:

$$a_2(n) = h_{12}(n) * a_1(n)$$

Data model and Acoustic Features The RTF

Relative Transfer Function (RTF) [Gannot et al., 2001]



RTF:

- Represents the acoustic paths and is independent of the source signal
- Generalizes the TDOA
- Depends on a small set of parameters related to the physical characteristics of the environment

Data model and Acoustic Features The RTF

Relative Transfer Function (RTF) [Gannot et al., 2001]



RTF:

- Represents the acoustic paths and is independent of the source signal
- Generalizes the TDOA
- Depends on a small set of parameters related to the physical characteristics of the environment
- In a static environment the source position is the only varying degree of freedom

Relative Transfer Function (RTF)

RTF-based feature vector:

• Estimated based on PSD and cross-PSD

(alternatively [Markovich-Golan and Gannot, 2015,

Koldovsky et al., 2014]):

$$\hat{H}_{12}(k) = rac{\hat{S}_{y_2y_1}(k)}{\hat{S}_{y_1y_1}(k)} \simeq rac{A_2(k)}{A_1(k)}$$

• Define the feature vector:

$$\mathbf{h} = \left[\hat{H}_{12}(k_1), \dots, \hat{H}_{12}(k_D)\right]^{T}$$

 D ∝ length of the relative impulse response (time domain)



High dimensional representation due to reverberation

> Controlled by one dominant factor source position

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How to Measure the Affinity between RTFs? [Laufer-Goldshtein et al., 2015, Laufer-Goldshtein et al., 2016b]

- The RTFs are represented as points in a high dimensional space
- Small Euclidean distance of high dimensional vectors implies proximity
- Large Euclidean distance of high dimensional vectors is meaningless



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Acoustic manifold

 \bullet They lie on a low dimensional nonlinear manifold ${\cal M}$

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- Linearity is preserved in small neighbourhoods

Speaker Localization on Manifolds

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Acoustic manifold

- \bullet They lie on a low dimensional nonlinear manifold ${\cal M}$
- Linearity is preserved in small neighbourhoods
- Distances between RTFs should be measured along the manifold

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Each distance measure relies on a different hidden assumption about the underlying structure of the RTF samples

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Euclidean Distance

The Euclidean distance between RTFs

$$D_{\mathrm{Euc}}(\mathbf{h}_i,\mathbf{h}_j) = \|\mathbf{h}_i - \mathbf{h}_j\|$$

- Compares two RTFs in their original space
- Does not assume an existence of a manifold
- Respects flat manifolds



A good affinity measure only when the RTFs are uniformly scattered all over the space, or when they lie on a flat manifold

PCA-Based Distance

PCA algorithm

- The principal components the *d* dominant eigenvectors $\{\mathbf{v}_i\}_{i=1}^d$ of the covariance matrix of the data
- The RTFs are linearly projected onto the principal components:

$$oldsymbol{
u}\left(\mathbf{h}_{i}
ight)=\left[\mathbf{v}_{1},\ldots\mathbf{v}_{d}
ight]^{T}\left(\mathbf{h}_{i}-oldsymbol{\mu}
ight)$$

PCA-based distance between RTFs

$$D_{ ext{PCA}}(\mathbf{h}_i,\mathbf{h}_j) = \|oldsymbol{
u}(\mathbf{h}_i) - oldsymbol{
u}(\mathbf{h}_j)\|$$

- A global approach extracts principal directions of the entire set
- Linear projections the manifold is assumed to be linear/flat



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Discretization of the manifold

• The manifold can be empirically represented by a graph:

- The RTF samples are the graph nodes
- The weights of the edges are defined using a kernel function:

$$\mathcal{K}_{ij} = k(\mathbf{h}_i, \mathbf{h}_j) = \exp\left\{-rac{\|\mathbf{h}_i - \mathbf{h}_j\|^2}{arepsilon}
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• Define a Markov process on the graph by the transition matrix: $p(\mathbf{h}_i, \mathbf{h}_j) = P_{ij} = K_{ij} / \sum_{r=1}^N K_{ir}$

which is a discretization of a diffusion process on the manifold



Diffusion mapping [Colfman and Lafon, 2006]

- Apply eigenvalue decomposition (EVD) to the matrix P and obtain the eigenvalues {λ_j} and right eigenvectors {φ_j}.
- A nonlinear mapping into a new low-dimensional Euclidean space:

$$\mathbf{\Phi}_d: \mathbf{h}_i \mapsto \left[\lambda_1 \varphi_1^{(i)}, \dots, \lambda_d \varphi_d^{(i)}\right]^T$$



The mapping provides a parametrization of the manifold and represents the latent variables - here, the position of the source

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Diffusion Distance

Diffusion distance between RTFs

The distance along the manifold is approximated by the diffusion distance:

$$D_{\mathrm{Diff}}^{2}(\mathbf{h}_{i},\mathbf{h}_{j})=\sum_{r=1}^{N}\left(p\left(\mathbf{h}_{i},\mathbf{h}_{r}
ight) -p\left(\mathbf{h}_{j},\mathbf{h}_{r}
ight)
ight) ^{2}/\phi_{0}^{\left(r
ight)}$$

• Two points are close if they are highly connected in the graph

• The diffusion distance can be well approximated by the Euclidian distance in the embedded domain:

$$D_{ ext{Diff}}(\mathbf{h}_i,\mathbf{h}_j)\cong \|\mathbf{\Phi}_d(\mathbf{h}_i)-\mathbf{\Phi}_d(\mathbf{h}_j)\|$$



Diffusion Distance

Illustration



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Which of the distance measures is proper? What is the true underlying structure of the RTFs?

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Simulation Results

Room setup

Simulate a reverberant room using the image method [Allen and Berkley, 1979]:

- $\bullet~$ Room dimension $6\times 6.2\times 3m$
- \bullet Microphones at: [3,3,1] and [3.2,3,1]
- The source is positioned at 2m from the mics, the azimuth angle in 10° ÷ 60°.
- $T_{60} = 150/300/500 \text{ ms}$
- SNR= 20 dB



Test

Measure the distance between each of the RTFs and the RTF corresponding to $10^\circ\colon$

- If monotonic with respect to the angle proper distance
- If not monotonic with respect to the angle improper distance

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Euclidean Distance & PCA-based Distance [Laufer-Goldshtein et al., 2015]



For both distance measures:

- Monotonic with respect to the angle only in a limited region
- This region becomes smaller as the reverberation time increases
- They are inappropriate for measuring angles' proximity

Diffusion Maps



The diffusion distance:

- Monotonic with respect to the angle for almost the entire range
- It is an appropriate distance measure in terms of the source DOA
- Mapping corresponds well with angles recovers the latent parameter

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Mixed of supervised (attached with known locations as anchors) and unsupervised (unknown locations) learning

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Why using unlabeled data?

1 Localization - training should fit the specific environment of interest

- Cannot generate a general database for all possible acoustic scenarios
- Generating a large amount of labelled data is cumbersome/impractical
- Unlabelled data is freely available whenever someone is speaking



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2 Unlabelled data can be utilized to recover the manifold structure



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- Unlabelled data is freely available whenever someone is speaking
- Output the second se
- Semi-supervised learning is the natural setting for human learning



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Data-Driven Source Localization: Microphone Pair

Semi-Supervised Learning

Unlabelled Samples

Labelled Samples

Recover the Manifold Structure

Anchor Points – Translate RTFs to Positions







Data:

- $H_L = {\mathbf{h}_i}_{i=1}^{n_L} n_L$ labelled samples
- $P_L = \{\bar{p}_i\}_{i=1}^{n_L}$ labels/positions
- $H_U = {\{\mathbf{h}_i\}}_{i=n_L+1}^{n_D}$ n_U unlabelled samples
- $H_D = H_L \cup H_U$ entire training set
- $H_T = {\mathbf{h}_i}_{i=n_D+1}^n n_T$ test samples



Test Set



Data:

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- $H_T = {\mathbf{h}_i}_{i=n_D+1}^n n_T$ test samples



Test Set

Goal: Recover a (component-wise) function $p = f(\mathbf{h})$ which transforms an RTF to position

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Optimization in a reproducing kernel Hilbert space (RKHS) [Belkin et al., 2006]:

$$f^* = \operatorname*{argmin}_{f \in \mathcal{H}_k} \frac{1}{n_L} \sum_{i=1}^{n_L} (\bar{p}_i - f(\mathbf{h}_i))^2 + \gamma_k \|f\|_{\mathcal{H}_k}^2 + \gamma_M \|f\|_{\mathcal{M}}^2$$

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Speaker Localization on Manifolds

Manifold Regularization

Discretization of the manifold

• The manifold is empirically represented by a graph G with weights:

$$W_{ij} = \left\{ egin{array}{c} \exp\left\{-rac{\|\mathbf{h}_i - \mathbf{h}_j\|^2}{arepsilon_w}
ight\} & ext{if } \mathbf{h}_j \in \mathcal{N}_i ext{ or } \mathbf{h}_i \in \mathcal{N}_i \ \mathbf{0} & ext{otherwise} \end{array}
ight.$$

where \mathcal{N}_j is a set consisting of the *d* nearest-neighbours of \mathbf{h}_j

• The graph Laplacian of G: $\mathbf{M} = \mathbf{S} - \mathbf{W}$, with $S_{ii} = \sum_{j=1}^{n_D} \mathbf{W}_{ij}$

• Regularization: $||f||_{\mathcal{M}}^2 = \mathbf{f}_D^T \mathbf{M} \mathbf{f}_D = \frac{1}{2} \sum_{i,j=1}^{n_D} W_{ij} (f(\mathbf{h}_i) - f(\mathbf{h}_j))^2$ with $\mathbf{f}_D^T = [f_1, f_2, \dots, f_{n_D}]$ • Proof



The optimization problem can be recast as:

$$f^* = \operatorname*{argmin}_{f \in \mathcal{H}_k} \frac{1}{n_L} \sum_{i=1}^{n_L} (\bar{p}_i - f(\mathbf{h}_i))^2 + \gamma_k \|f\|_{\mathcal{H}_k}^2 + \gamma_M \mathbf{f}_D^T \mathbf{M} \mathbf{f}_D$$

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The representer theorem:

The minimizer over \mathcal{H}_k of the regularized optimization is represented by:

$$f(\mathbf{h}) = \sum_{i=1}^{n_D} a_i k(\mathbf{h}_i, \mathbf{h})$$

where $k : \mathcal{M} \times \mathcal{M} \to \mathbb{R}$ is the reproducing kernel of \mathcal{H}_k with $K_{ij} = k(\mathbf{h}_i, \mathbf{h}_j) = \exp\left\{-\frac{\|\mathbf{h}_i - \mathbf{h}_j\|^2}{\varepsilon}\right\}$

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$$f(\mathbf{h}) = \sum_{i=1}^{n_D} a_i k(\mathbf{h}_i, \mathbf{h}) \quad \Rightarrow \quad \text{closed-form solution for } \mathbf{a}^*$$

where $k:\mathcal{M}\times\mathcal{M}
ightarrow\mathbb{R}$ is the reproducing kernel of \mathcal{H}_k



Manifold Regularization for Localization (MRL)

[Laufer-Goldshtein et al., 2016c]



$$f^* = \operatorname{argmin}_{f \in \mathcal{H}_k} \frac{1}{n_L} \sum_{i=1}^{n_L} (\bar{p}_i - f(\mathbf{h}_i))^2 + \gamma_k \|f\|_{\mathcal{H}_k}^2 + \gamma_M \mathbf{f}_D^T \mathbf{M} \mathbf{f}_D$$

Search in RKHS defined by the kernel k Cost Function \mathcal{H}_k norm Manifold Regularization

$$f^{*} = \operatorname{argmin}_{f \in \mathcal{H}_{k}} \frac{1}{n_{L}} \sum_{i=1}^{n_{L}} (\bar{p}_{i} - f(\mathbf{h}_{i}))^{2} + \gamma_{k} \|f\|_{\mathcal{H}_{k}}^{2} + \gamma_{M} \mathbf{f}_{D}^{T} \mathbf{M} \mathbf{f}_{D}$$
Search in RKHS defined by the kernel k

$$f^{*} = \operatorname{argmin}_{f \in \mathcal{H}_{\tilde{k}}} \frac{1}{n_{L}} \sum_{i=1}^{n_{L}} (\bar{p}_{i} - f(\mathbf{h}_{i}))^{2} + \gamma_{k} \|f\|_{\mathcal{H}_{\tilde{k}}}^{2}$$
Search in RKHS defined by the kernel \tilde{k}

$$\operatorname{Cost Function} \qquad \mathcal{H}_{\tilde{k}} \operatorname{norm}$$



Localization

MAP/MMSE estimator of $f(\mathbf{h}_t)$, s.t. $\mathbf{h}_t \in \mathcal{M}$:

- $\mathbf{\bar{p}}_L = [\bar{p}_1, \dots, \bar{p}_{n_L}]^T$ measured positions of the labelled set
- $\bar{p}_i = p_i + \eta_i$ noisy versions of the actual position p_i
- η_i independent Gaussian noise with variance σ^2
- $\bar{\mathbf{p}}_L$ and $f(\mathbf{h}_t)$ are jointly Gaussian $(\tilde{\mathbf{\Sigma}}_{HH} \Leftrightarrow \tilde{k}(\mathbf{h}_i, \mathbf{h}_j))$:

$$\begin{bmatrix} \bar{\mathbf{p}}_L \\ f(\mathbf{h}_t) \end{bmatrix} \left| H_L, H_U \sim \mathcal{N} \left(\mathbf{0}_{n_L+1}, \begin{bmatrix} \tilde{\mathbf{\Sigma}}_{LL} + \sigma^2 \mathbf{I}_{n_L} & \tilde{\mathbf{\Sigma}}_{Lt} \\ \tilde{\mathbf{\Sigma}}_{Lt}^T & \tilde{\mathbf{\Sigma}}_{tt} \end{bmatrix} \right)$$

• The posterior $p(f(\mathbf{h}_t)|P_L, H_L, H_U)$ is a multivariate Gaussian with: $\mu_{\text{cond}} = \tilde{\boldsymbol{\Sigma}}_{Lt}^T \left(\tilde{\boldsymbol{\Sigma}}_{LL} + \sigma^2 \mathbf{I}_{n_L} \right)^{-1} \bar{\mathbf{p}}_L \Rightarrow \text{MAP: } \hat{f}(\mathbf{h}_t)$ $\sigma_{\text{cond}}^2 = \tilde{\boldsymbol{\Sigma}}_{tt} - \tilde{\boldsymbol{\Sigma}}_{Lt}^T \left(\tilde{\boldsymbol{\Sigma}}_{LL} + \sigma^2 \mathbf{I}_{n_L} \right)^{-1} \tilde{\boldsymbol{\Sigma}}_{Lt} \Rightarrow \text{respective reliability}$

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Source Localization with Ad Hoc Array [Laufer-Goldshtein et al., 2016d]



Each node:

- Represents a different view points of the same acoustic event
- Induces relations between RTFs w.r.t. the associated manifold

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Source Localization with Ad Hoc Array [Laufer-Goldshtein et al., 2016d]



How to fuse the different views in a unified mapping $f : \bigcup_{m=1}^{M} \mathcal{M}_m \mapsto \mathbb{R}$?

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Speaker Localization on Manifolds

Define the average process $f = \frac{1}{M}(f^1 + f^2 + \ldots + f^M) \sim \mathcal{GP}(0, \tilde{k})$



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The covariance between $p_r = f(\mathbf{h}_r)$ and $p_l = f(\mathbf{h}_l)$

 $\operatorname{cov}(f(\mathbf{h}_r), f(\mathbf{h}_l)) \equiv \tilde{k}(\mathbf{h}_r, \mathbf{h}_l) = \frac{1}{M^2} \sum_{q,w=1}^{M} \sum_{i=1}^{n_D} k_q(\mathbf{h}_r^q, \mathbf{h}_i^q) k_w(\mathbf{h}_l^w, \mathbf{h}_i^w)$

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 $\operatorname{cov}(f(\mathbf{h}_r), f(\mathbf{h}_l)) \equiv \tilde{k}(\mathbf{h}_r, \mathbf{h}_l) = \frac{1}{M^2} \sum_{q,w=1}^{M} \sum_{i=1}^{n_D} k_q(\mathbf{h}_r^q, \mathbf{h}_i^q) k_w(\mathbf{h}_l^w, \mathbf{h}_i^w)$

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Speaker Localization on Manifolds

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Speaker Localization on Manifolds
Multiple Manifold Gaussian Process (MMGP)

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Localization

MAP/MMSE estimator:

- Goal: estimate the function value at some test sample $\mathbf{h}_t \in \mathcal{M}$
- The training positions $\bar{\mathbf{p}}_L = \operatorname{vec}\{P_L\}$ and $f(\mathbf{h}_t)$ are jointly Gaussian $(\tilde{\mathbf{\Sigma}}_{HH} \Leftrightarrow \tilde{k}(\mathbf{h}_i, \mathbf{h}_j))$:

$$\begin{bmatrix} \bar{\mathbf{p}}_L \\ f(\mathbf{h}_t) \end{bmatrix} \left| H_L, H_U \sim \mathcal{N} \left(\mathbf{0}_{n_L+1}, \begin{bmatrix} \tilde{\mathbf{\Sigma}}_{LL} + \sigma^2 \mathbf{I}_{n_L} & \tilde{\mathbf{\Sigma}}_{Lt} \\ \tilde{\mathbf{\Sigma}}_{Lt}^T & \tilde{\mathbf{\Sigma}}_{tt} \end{bmatrix} \right)$$

• The posterior $p(f(\mathbf{h}_t)|P_L, H_L, H_U)$ is a multivariate Gaussian with: $\mu_{\text{cond}} = \tilde{\mathbf{\Sigma}}_{Lt}^T \left(\tilde{\mathbf{\Sigma}}_{LL} + \sigma^2 \mathbf{I}_{n_L}\right)^{-1} \bar{\mathbf{p}}_L$ $\sigma_{\text{cond}}^2 = \tilde{\mathbf{\Sigma}}_{tt} - \tilde{\mathbf{\Sigma}}_{Lt}^T \left(\tilde{\mathbf{\Sigma}}_{LL} + \sigma^2 \mathbf{I}_{n_L}\right)^{-1} \tilde{\mathbf{\Sigma}}_{Lt}$

The MAP/MMSE estimator of $f(\mathbf{h}_t)$ is given by:

$$\hat{f}(\mathbf{h}_t) = \mu_{\text{cond}} = \tilde{\mathbf{\Sigma}}_{Lt}^T \left(\tilde{\mathbf{\Sigma}}_{LL} + \sigma^2 \mathbf{I}_{n_L} \right)^{-1} \bar{\mathbf{p}}_L$$

Recordings Setup

Setup:

- Real recordings carried out at Bar-Ilan acoustic lab
- A $6 \times 6 \times 2.4m$ room controllable reverberation time (set to 620ms)
- $\bullet\,$ Region of interest: Source position is confined to a $2.8\times2.1m$ area
- 3 microphone pairs with inter-distance of 0.2m (position unknown)



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Experimental Results [Laufer-Goldshtein et al., 2016d]

Setup:

- Training: 20 labelled samples (0.7m resolution), 50 unlabelled samples
- Test: 25 random samples in the defined region
- Two noise types: air-conditioner noise and babble noise

Compare with:

- Concatenated independent measurements (Kernel-mult)
- Average of single-node estimates (Mean)
- Beamformer scanning (SRP-PHAT [DiBiase et al., 2001])



Outline

- Data model and Acoustic Features
- 2 The Acoustic Manifold
- 3 Data-Driven Source Localization: Microphone Pair
- 4 Data-Driven Source Localization: Ad Hoc Array
- 5 Speaker Tracking on Manifolds

Dynamic Scenario

Received Signals

$$y^{mi}(n) = \sum_{k} a_n^{mi}(k) s(n-k) + u^{mi}(n); \quad m = 1, \dots, M, i = 1, 2$$

- a_n^{mi} a time-varying AIR at node m, microphone i in time n
- $\mathbf{h}^m(t)$ the RTF vector at node m in the STFT frame t
- h(t) = [[h¹(t)]^T,..., [h^M(t)]^T]^T a concatenation of the RTF vectors from all nodes
- p_c(t) = f(h(t)), c ∈ {x, y, z} mapping of the concatenated RTF vector to position (for brevity p_c(t) ≡ p(t))

Reminder: The covariance between $p_r = f(\mathbf{h}_r)$ and $p_l = f(\mathbf{h}_l)$ $\operatorname{cov}(f(\mathbf{h}_r), f(\mathbf{h}_l)) \equiv \tilde{k}(\mathbf{h}_r, \mathbf{h}_l) = \frac{1}{M^2} \sum_{q,w=1}^{M} \sum_{i=1}^{n_D} k_q(\mathbf{h}_r^q, \mathbf{h}_i^q) k_w(\mathbf{h}_l^w, \mathbf{h}_i^w)$

Bayesian Inference for Source Tracking

Standard (Nonlinear) State-Space Model

$$p(t) = b_t(p(t-1)) + \xi_t$$

 $q_t = c_t(p(t)) + \zeta_t$

Bayesian Inference for Source Tracking

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Propagation Model

- Relate current and previous positions arbitrarily using random walk or Langevin
- Independent of measurements
- Noise statistics is unknown



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Observation Model

- Relate current position to measurements
- Examples: TDOA or steered response power readings
- Noise statistics is unknown



Tracking on the Manifold [Laufer-Goldshtein et al., 2017]

Propagation Model - Local

Transform nonlinear regression of high-dimensional RTFs to linear transition of source positions



Tracking on the Manifold [Laufer-Goldshtein et al., 2017]

Propagation Model - Local

Transforms nonlinear regression of high-dimensional RTFs to linear transition of source positions

Observation model - Global

Formed by a regression of training positions according to relations on the manifold





State Space Representation (1)

Probabilistic Motion Model:

• Current and previous positions, $p(t) = f(\mathbf{h}(t))$ and $p(t-1) = f(\mathbf{h}(t-1))$, are jointly GP:

$$\left[egin{array}{c} p(t) \ p(t-1) \end{array}
ight]\sim \mathcal{N}\left(oldsymbol{0}, \left[egin{array}{c} \widetilde{\Sigma}_{t,t} & \widetilde{\Sigma}_{t,t-1} \ \widetilde{\Sigma}_{t,t-1} & \widetilde{\Sigma}_{t-1,t-1} \end{array}
ight]
ight)$$

• Their conditional probability is given by:

$$p(t)|p(t-1) \sim \mathcal{N}\left(\frac{\tilde{\Sigma}_{t,t-1}}{\tilde{\Sigma}_{t-1,t-1}}p(t-1),\tilde{\Sigma}_{t,t}-\frac{\tilde{\Sigma}_{t,t-1}^2}{\tilde{\Sigma}_{t-1,t-1}}\right)$$

where $\tilde{\Sigma}_{t,\tau} \equiv \tilde{k} \left(\mathbf{h}(t), \mathbf{h}(\tau) \right)$

State Space Representation (2)

Propagation Model:

Can be transformed into a linear propagation equation with an additive Gaussian noise ξ_t :

 $p(t) = b_t \cdot p(t-1) + \xi_t$

with

•
$$b_t = \frac{\tilde{\Sigma}_{t,t-1}}{\tilde{\Sigma}_{t-1,t-1}}$$
 - The Wiener filter
• $\xi_t \sim \mathcal{N}\left(0, \sigma_{\xi}^2\right)$ with $\sigma_{\xi}^2 = \tilde{\Sigma}_{t,t} - \frac{\tilde{\Sigma}_{t,t-1}^2}{\tilde{\Sigma}_{t-1,t-1}}$, the corresponding variance



State Space Representation (3)

Probabilistic Observation Model:

- $\mathbf{\bar{p}}_L = [\bar{p}_1, \dots, \bar{p}_{n_L}]^T$ measured positions of the labelled set
- $\bar{p}_i = p_i + \eta_i$ noisy versions of the actual position p_i
- η_i independent Gaussian noise with variance σ^2
- $p(t) = f(\mathbf{h}(t))$ and $\mathbf{\bar{p}}_L$ are jointly GP:

$$\begin{bmatrix} p(t) \\ \bar{\mathbf{p}}_L \end{bmatrix} \sim \mathcal{N} \left(\mathbf{0}, \begin{bmatrix} \tilde{\Sigma}_{t,t} & \tilde{\Sigma}_{Lt} \\ \tilde{\Sigma}_{Lt} & \tilde{\Sigma}_{LL} + \sigma^2 \mathbf{I}_{n_L} \end{bmatrix} \right)$$

• Their conditional probability is given by:

$$\begin{split} & \mathcal{N}\left(\tilde{\boldsymbol{\Sigma}}_{Lt}^{H}\left(\tilde{\boldsymbol{\Sigma}}_{L}+\sigma^{2}\boldsymbol{\mathsf{I}}_{n_{L}}\right)^{-1}\bar{\boldsymbol{\mathsf{p}}}_{L},\tilde{\boldsymbol{\Sigma}}_{t,t}-\tilde{\boldsymbol{\Sigma}}_{Lt}^{H}\left(\tilde{\boldsymbol{\Sigma}}_{L}+\sigma^{2}\boldsymbol{\mathsf{I}}_{n_{L}}\right)^{-1}\tilde{\boldsymbol{\Sigma}}_{Lt}\right) \end{split}$$

State-Space Representation (4)

Observation model:

• Can be transformed into a noisy artificial observation q_t that represents a linear regression on the training set:

$$q_t = \mathbf{Q}_t \bar{\mathbf{p}}_L$$

where
$$\mathbf{Q}_t = \tilde{\mathbf{\Sigma}}_{Lt}^H \left(\tilde{\mathbf{\Sigma}}_{LL} + \sigma^2 \mathbf{I}_{n_L} \right)^{-1}$$



The corresponding observation model:

$$q_t = p(t) + \zeta_t$$

where
$$\zeta_t \sim \mathcal{N}\left(0, \sigma_{\zeta}^2\right)$$
 with $\sigma_{\zeta}^2 = \tilde{\Sigma}_{t,t} - \tilde{\Sigma}_{Lt}^H \left(\tilde{\Sigma}_{LL} + \mathbf{I}_{n_L}\right)^{-1} \tilde{\Sigma}_{Lt}$.

Tracking Algorithm

Space-State Representation:

The proposed state-space model is given by:

$$p(t) = b_t \cdot p(t-1) + \xi_t$$
$$q_t = p(t) + \zeta_t$$



Speaker Localization on Manifolds

Experimental Results

Setup:

- A 5.2 \times 6.2 \times 3m room with $T_{60} = 300$ ms
- M = 4 nodes with 0.2m distance between microphones
- Region of interest: a $2 \times 2m$ square region
- Training: 36 samples (0.4m resolution)



Results

Test:

- Trajectories: straight line (for 3s) and sinusoidal movement (for 5s).
- Velocity: approximately 1m/s



RMSE: 13cm for straight line and 17cm for sinusoidal movement.

Challenges and Perspectives

Manifold learning approach for source localization

- Data-driven manifold inference with a few labeled anchor positions and unknown microphone positions.
- Location is shown to be the controlling variable of the RTF manifold
- It's practical!
- Active research field [Deleforge et al., 2015][Yu et al., 2016][Xiao et al., 2015]
- Improved speaker tracking \Rightarrow Hybrid approach $_{[Laufer-Goldshtein \mbox{ et al., 2018}]}$

Challenges

- Robustness to changes in array constellation and acoustic scenario
- Application to multiple concurrent speakers
- Beamforming more complicated as it targets enhanced speech rather than its location
 - A first attempt using projections to the inferred manifold

[Talmon and Gannot, 2013]

Details

Manifold Regularization

Measuring smoothness over \mathcal{M} :

- The gradient $\bigtriangledown_{\mathcal{M}} f(\mathbf{h})$ represents variations around \mathbf{h}
- A natural choice for intrinsic regularization:

$$\|f\|_{\mathcal{M}}^2 = \int_{\mathcal{M}} \|\bigtriangledown_{\mathcal{M}} f(\mathbf{h})\|^2 dp(\mathbf{h})$$

which is a global measure of smoothness for f

• Stokes' theorem links gradient and Laplacian:

$$\int_{\mathcal{M}} \| \bigtriangledown_{\mathcal{M}} f(\mathbf{h}) \|^2 dp(\mathbf{h}) = \int_{\mathcal{M}} f(\mathbf{h}) \bigtriangleup_{\mathcal{M}} f(\mathbf{h}) dp(\mathbf{h}) = \langle f(\mathbf{h}), \bigtriangleup_{\mathcal{M}} f(\mathbf{h}) \rangle$$

where $riangle_{\mathcal{M}}$ is the Laplace-Beltrami operator

How to reconstruct the Laplace-Beltrami operator on \mathcal{M} , given the training samples from the manifold?

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Speaker Localization on Manifolds

Manifold Regularization

Graph Laplacian:

• The manifold is empirically represented by a graph G, with weights:

$$W_{ij} = \begin{cases} \exp\left\{-\frac{\|\mathbf{h}_i - \mathbf{h}_j\|^2}{\varepsilon_w}\right\} \text{ if } \mathbf{h}_j \in \mathcal{N}_i \text{ or } \mathbf{h}_i \in \mathcal{N}_j \\ 0 & \text{otherwise} \end{cases}$$

where \mathcal{N}_j is a set consisting of the *d* nearest-neighbours of \mathbf{h}_j .

• The graph Laplacian of G: $\mathbf{M} = \mathbf{S} - \mathbf{W}$, with $S_{ii} = \sum_{j=1}^{n_D} \mathbf{W}_{ij}$.

• Smoothness functional of G:

$$\langle \mathbf{f}_D, \mathbf{M}\mathbf{f}_D \rangle = \mathbf{f}_D^T \mathbf{M}\mathbf{f}_D$$

where $\mathbf{f}_{D} = [f(\mathbf{h}_{1}), ..., f(\mathbf{h}_{n_{D}})]$

● It can be shown: ▶ Back

$$\mathbf{f}_D^T \mathbf{M} \mathbf{f}_D = \frac{1}{2} \sum_{i,j=1}^{n_D} W_{ij} \left(f(\mathbf{h}_i) - f(\mathbf{h}_j) \right)^2$$

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