

Multi-Microphone Speaker Localization and Tracking on Manifolds

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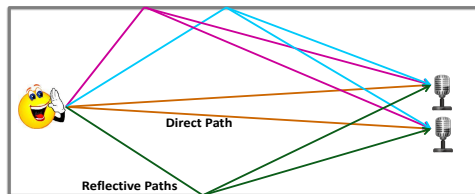
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Acoustic Source Localization

Goal

Locate a sound source given measurements of the sound field



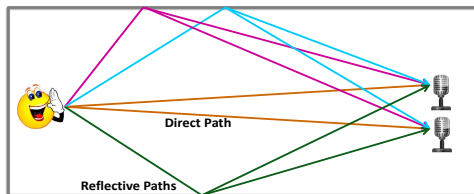
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Applications

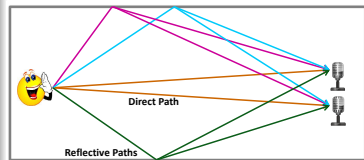
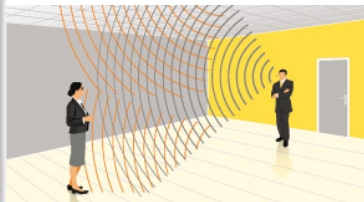
- 1 An essential component in speech enhancement algorithms
- 2 Camera steering
- 3 Teleconferencing
- 4 Robot audition
- 5 Surveillance
- 6 Smart home/clinic/car



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]



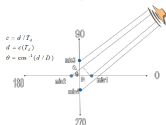
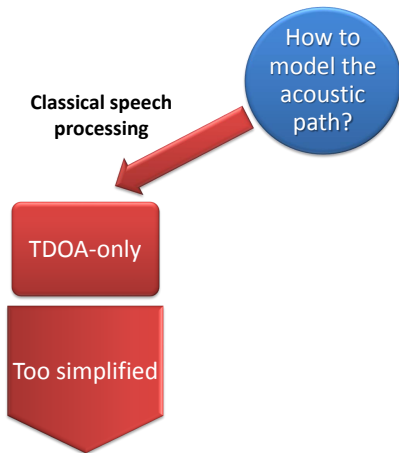
Speech Processing in Acoustic Environments

- 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

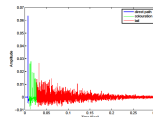
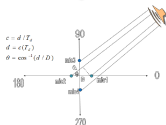
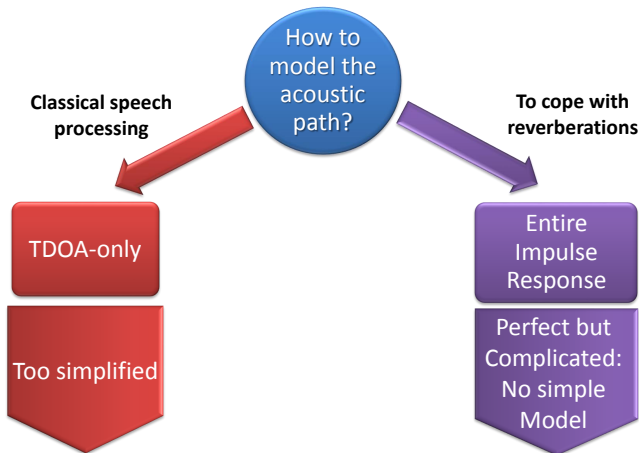
Data-driven approach

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

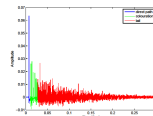
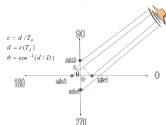
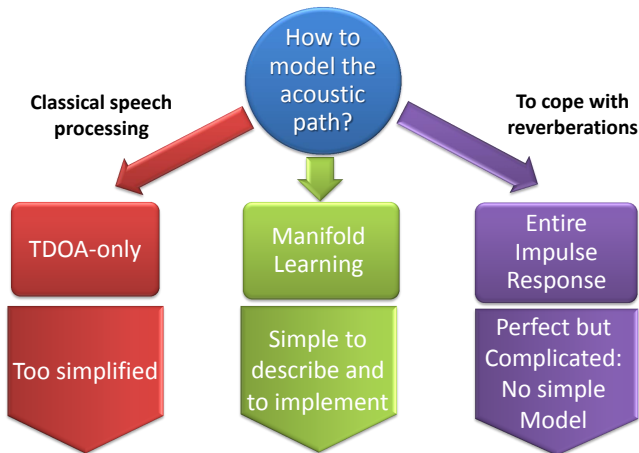
How to Model the Acoustic Environment?



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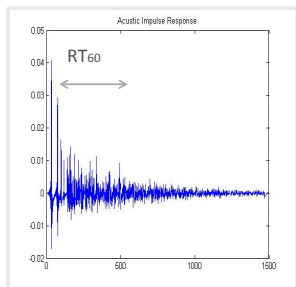


How to Model the Acoustic Environment?



Main Claims

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



Controlling Parameters

- room dimensions
- reverberation time
- microphone position
- source position
- ...



Outline

- 1 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

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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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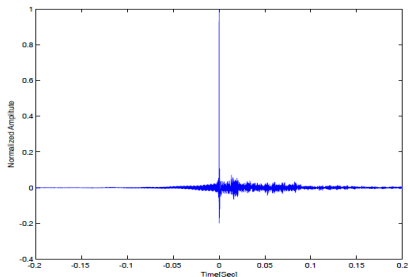
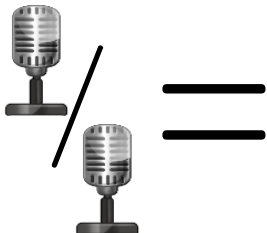
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- $a_i(n)$, $i = \{1, 2\}$ - the **acoustic impulse responses** relating the source and each of the microphones
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Find a **feature vector** representing the characteristics of the acoustic path (a **fingerprint**) and independent of the source signal!

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



RTF:

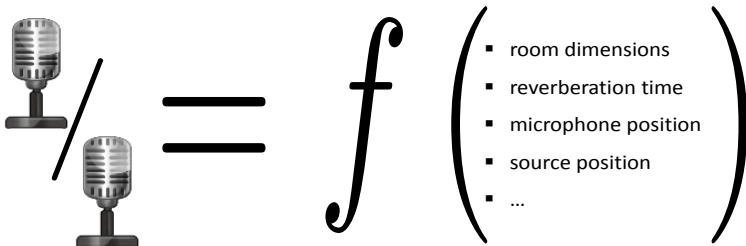
- Defined as the ratio between the **transfer functions** relating the source and the two mics:

$$H_{12}(k) = \frac{A_2(k)}{A_1(k)}$$

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

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

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



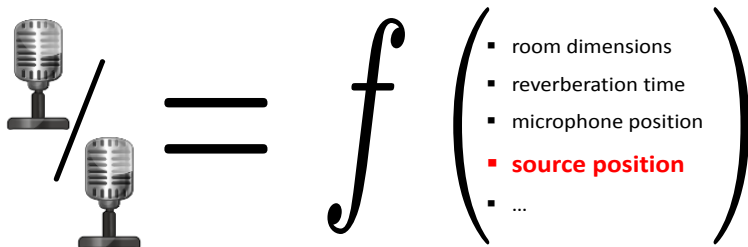
The diagram illustrates the Relative Transfer Function (RTF) as the ratio of two microphone signals, represented by two vintage microphones with a diagonal line between them. This is set equal to a function f that depends on a list of parameters: room dimensions, reverberation time, microphone position, source position, and an ellipsis.

$$\frac{\text{Microphone 1}}{\text{Microphone 2}} = f \left(\begin{array}{l} \text{room dimensions} \\ \text{reverberation time} \\ \text{microphone position} \\ \text{source position} \\ \dots \end{array} \right)$$

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

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



The diagram illustrates the Relative Transfer Function (RTF) as the ratio of two microphone signals, represented by two vintage microphones with a slash between them. This is set equal to a function f of several parameters listed in a large right parenthesis. The parameters are: room dimensions, reverberation time, microphone position, **source position** (highlighted in red), and an ellipsis.

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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]):

$$\hat{H}_{12}(k) = \frac{\hat{S}_{y_2 y_1}(k)}{\hat{S}_{y_1 y_1}(k)} \simeq \frac{A_2(k)}{A_1(k)}$$

- Define the feature vector:

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

- $D \propto$ length of the **relative impulse response** (time domain)



**High dimensional
representation -
due to reverberation**

**Controlled by one
dominant factor -
source position**

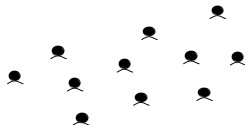
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How to Measure the **Affinity** between RTFs?

[Laufer-Goldshtein et al., 2015, Laufer-Goldshtein et al., 2016b]

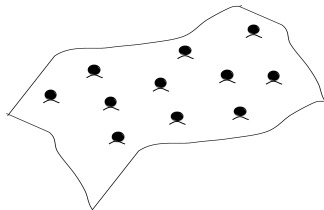
- The RTFs are represented as points in a **high dimensional space**
- Small Euclidean distance of high dimensional vectors implies proximity
- Large Euclidean distance of high dimensional vectors is meaningless



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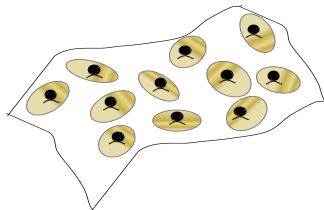
Acoustic manifold

- They lie on a **low dimensional nonlinear manifold \mathcal{M}**

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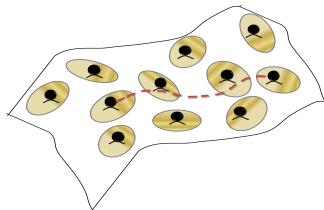
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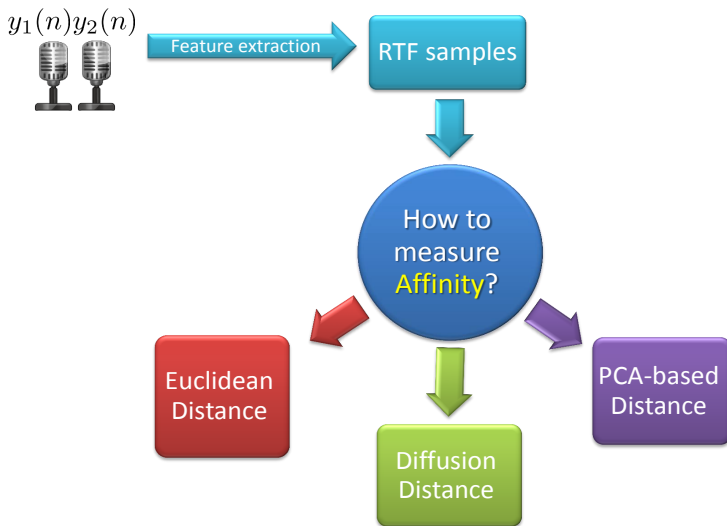
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Acoustic manifold

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



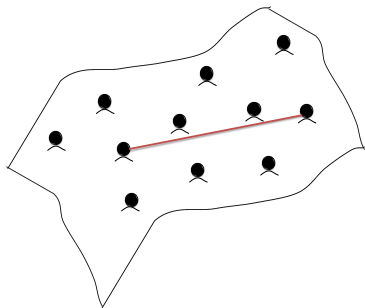
Each distance measure relies on a different **hidden assumption** about the **underlying structure** of the RTF samples

Euclidean Distance

The Euclidean distance between RTFs

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

- 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

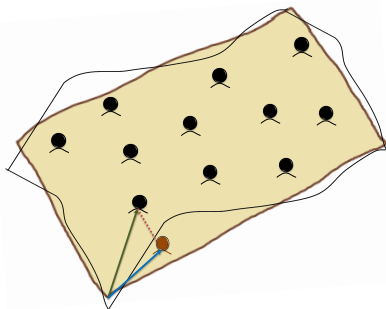
- The **principal components** - the d dominant eigenvectors $\{\mathbf{v}_i\}_{i=1}^d$ of the covariance matrix of the data
- The RTFs are **linearly projected** onto the principal components:

$$\nu(\mathbf{h}_i) = [\mathbf{v}_1, \dots, \mathbf{v}_d]^T (\mathbf{h}_i - \mu)$$

PCA-based distance between RTFs

$$D_{\text{PCA}}(\mathbf{h}_i, \mathbf{h}_j) = \|\nu(\mathbf{h}_i) - \nu(\mathbf{h}_j)\|$$

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



PCA-Based Distance

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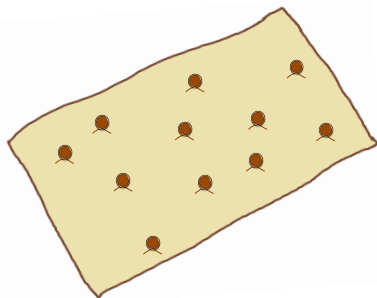
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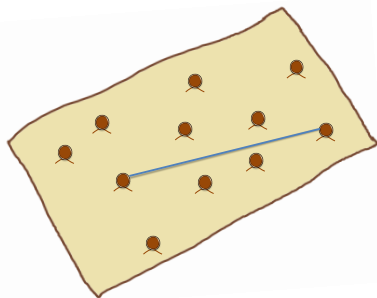
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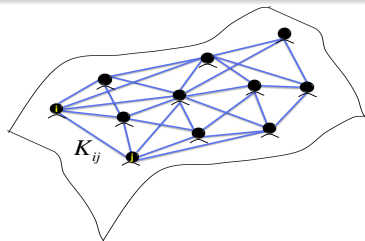


Diffusion Maps

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:

$$K_{ij} = k(\mathbf{h}_i, \mathbf{h}_j) = \exp \left\{ -\frac{\|\mathbf{h}_i - \mathbf{h}_j\|^2}{\varepsilon} \right\}$$

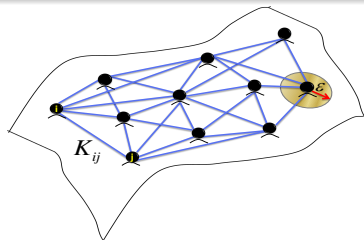


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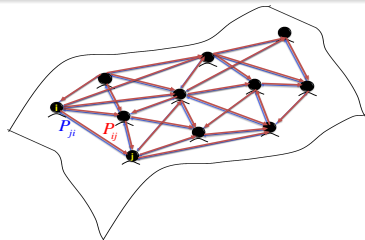
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- Define a **Markov process** on the graph by the **transition matrix**:

$$p(\mathbf{h}_i, \mathbf{h}_j) = P_{ij} = K_{ij} / \sum_{r=1}^N K_{ir}$$

which is a discretization of a **diffusion** process on the manifold

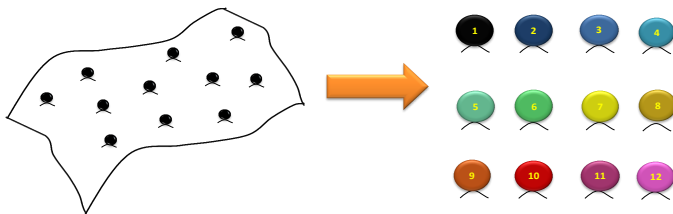


Diffusion Maps

Diffusion mapping [Coifman and Lafon, 2006]

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

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



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

Diffusion Distance

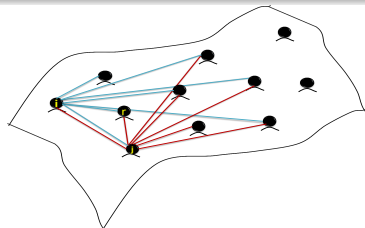
Diffusion distance between RTFs

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

$$D_{\text{Diff}}^2(\mathbf{h}_i, \mathbf{h}_j) = \sum_{r=1}^N (p(\mathbf{h}_i, \mathbf{h}_r) - p(\mathbf{h}_j, \mathbf{h}_r))^2 / \phi_0^{(r)}$$

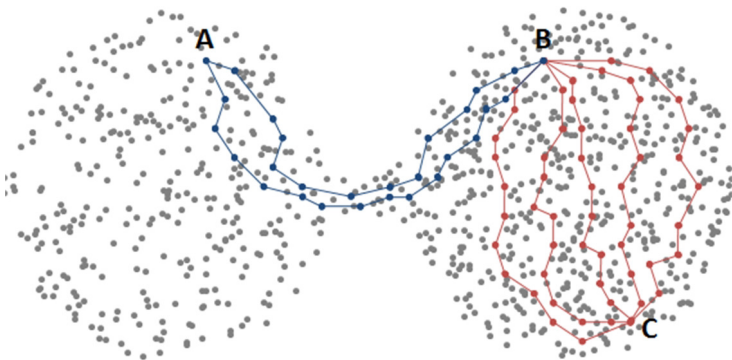
- Two points are close if they are highly connected in the graph
- The diffusion distance can be well approximated by the Euclidian distance in the embedded domain:

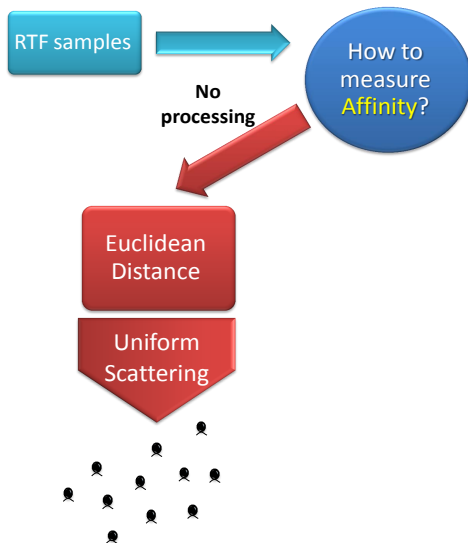
$$D_{\text{Diff}}(\mathbf{h}_i, \mathbf{h}_j) \cong \|\Phi_d(\mathbf{h}_i) - \Phi_d(\mathbf{h}_j)\|$$

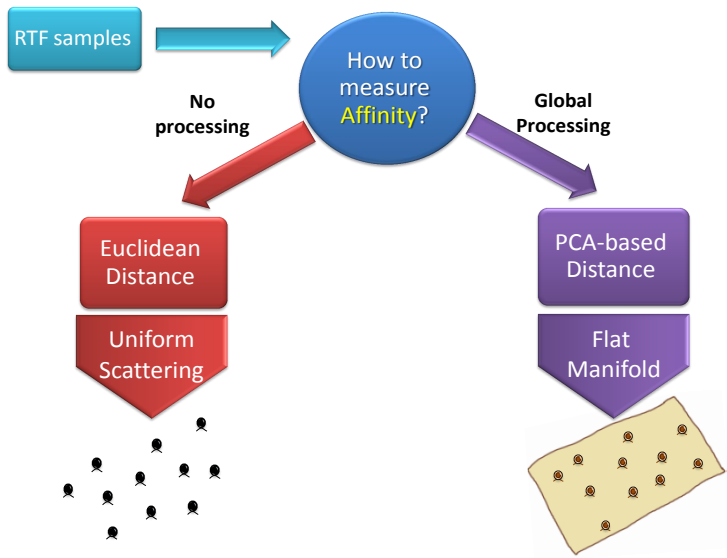


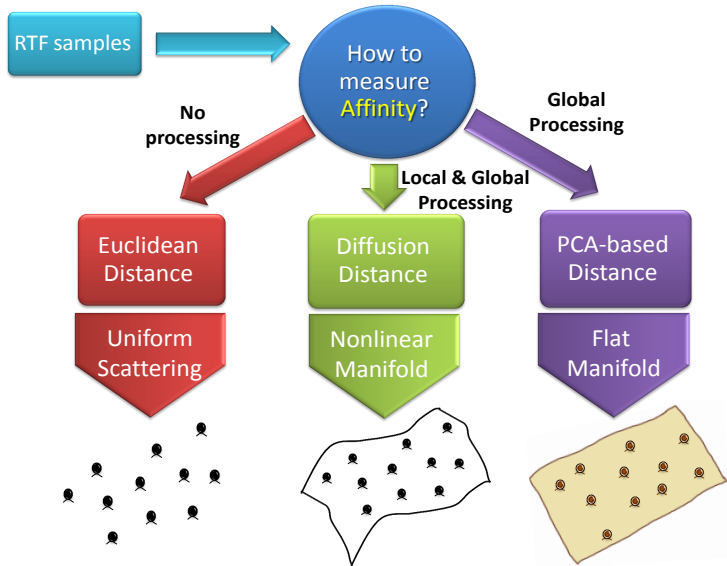
Diffusion Distance

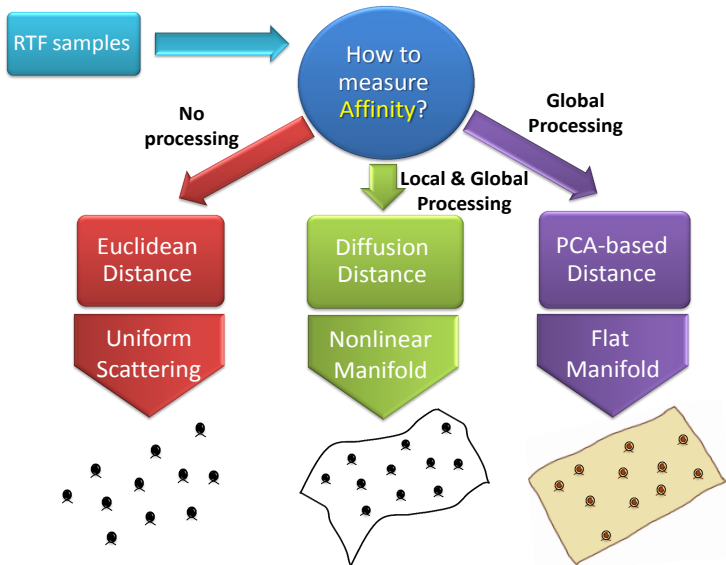
Illustration











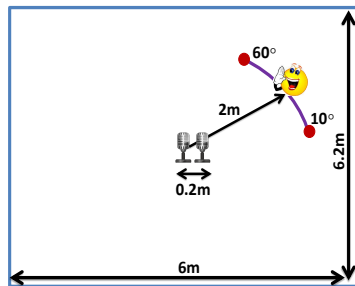
Which of the **distance measures** is proper?
What is the true **underlying structure** of the RTFs?

Simulation Results

Room setup

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

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

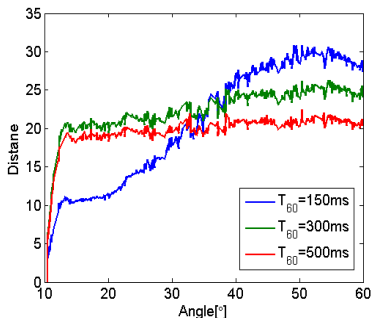


Test

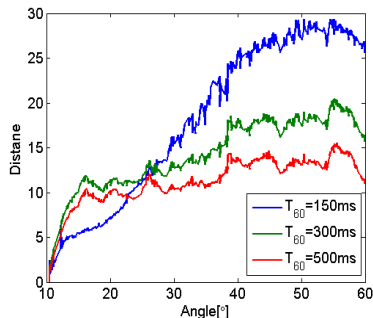
Measure the distance between each of the RTFs and the RTF corresponding to 10° :

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

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



(a) Euclidean Distance

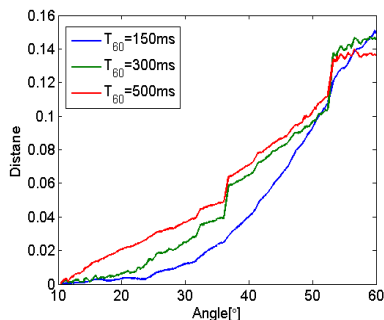


(b) PCA-based Distance

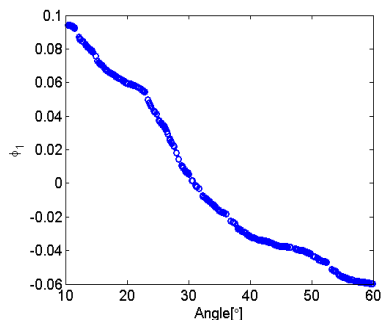
For both distance measures:

- 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



(c) Diffusion Distance



(d) Diffusion Mapping

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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Semi-Supervised Learning

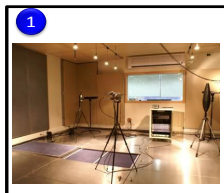
Mixed of **supervised** (attached with known locations as anchors) and **unsupervised** (unknown locations) learning

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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

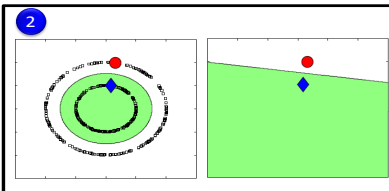
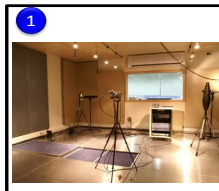


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 - Unlabelled data is **freely available** - whenever someone is speaking
- ② Unlabelled data can be utilized to recover the **manifold structure**

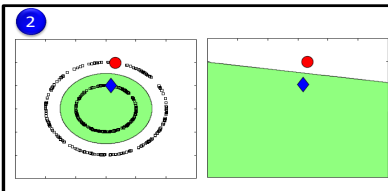
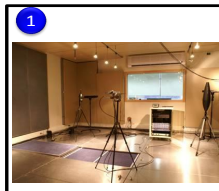


Semi-Supervised Learning

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

Why using unlabeled data?

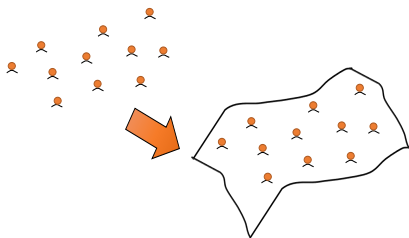
- ① **Localization** - training should fit the specific environment of interest
 - Cannot generate a general database for all possible acoustic scenarios
 - Generating a large amount of **labelled data** is cumbersome/impractical
 - Unlabelled data is **freely available** - whenever someone is speaking
- ② Unlabelled data can be utilized to recover the **manifold structure**
- ③ Semi-supervised learning is the natural setting for human learning



Semi-Supervised Learning

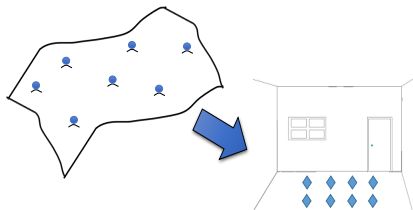
Unlabelled Samples

Recover the Manifold
Structure

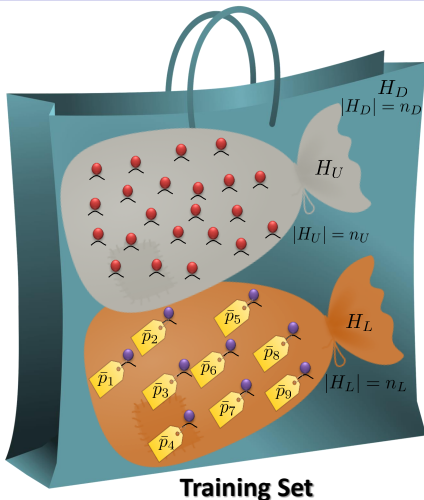


Labelled Samples

Anchor Points – Translate
RTFs to Positions

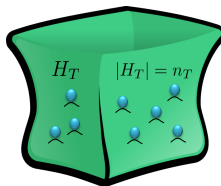


Semi-Supervised Learning



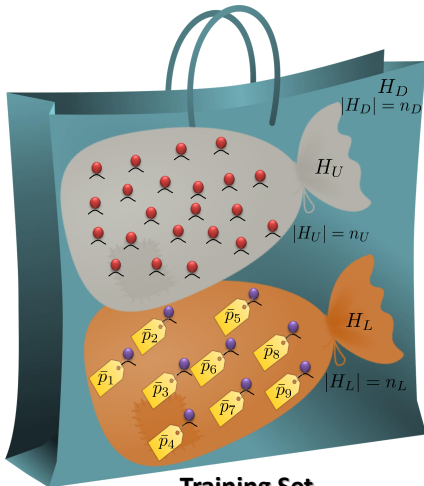
Data:

- $H_L = \{\mathbf{h}_i\}_{i=1}^{n_L}$ - 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

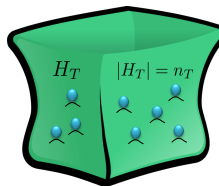
Semi-Supervised Learning



Training Set

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Test Set

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

Optimization and Manifold Regularization

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

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

Optimization and Manifold Regularization

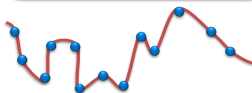
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Cost function

$$\frac{1}{n_L} \sum_{i=1}^{n_L} (\bar{p}_i - f(\mathbf{h}_i))^2$$

**correspondence
between
function values
and labels**



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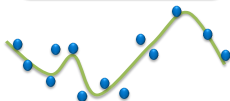
**Tikhonov
Regularization**

$$\|f\|_{\mathcal{H}_k}^2$$

**correspondence
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**smoothness
condition in
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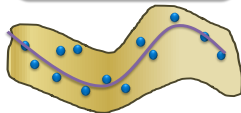
**smoothness
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**Manifold
Regularization**

$$\|f\|_{\mathcal{M}}^2$$

**smoothness
penalty with
respect to the
manifold**



Manifold Regularization

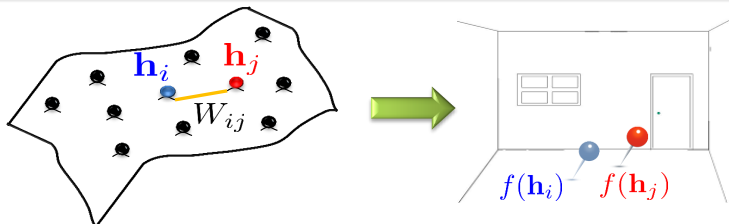
Discretization of the manifold

- 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 $S_{ii} = \sum_{j=1}^{n_D} \mathbf{W}_{ij}$
- 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](#)



Optimization and Manifold Regularization

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

with $K_{ij} = k(\mathbf{h}_i, \mathbf{h}_j) = \exp \left\{ -\frac{\|\mathbf{h}_i - \mathbf{h}_j\|^2}{\epsilon} \right\}$

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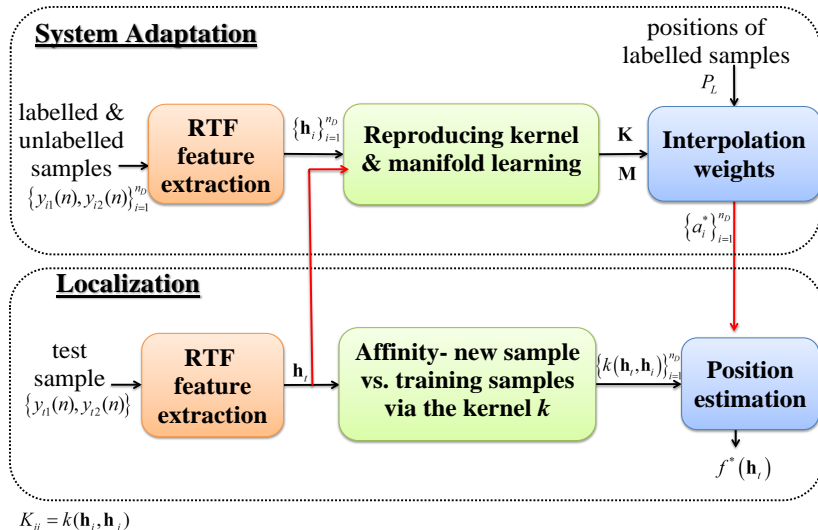
$$f(\mathbf{h}) = \sum_{i=1}^{n_D} a_i k(\mathbf{h}_i, \mathbf{h}) \quad \Rightarrow \quad \text{closed-form solution for } \mathbf{a}^*$$

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



Manifold Regularization for Localization (MRL)

[Laufer-Goldshtein et al., 2016c]



Bayesian Inference on Manifold [Laufer-Goldshtein et al., 2016a][Sindhwani et al., 2007]


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↓
Search in RKHS defined by the kernel k

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 Search in RKHS defined by the kernel k

$$f^* = \operatorname{argmin}_{f \in \mathcal{H}_{\tilde{k}}} \underbrace{\frac{1}{n_L} \sum_{i=1}^{n_L} (\bar{p}_i - f(\mathbf{h}_i))^2}_{\text{Cost Function}} + \underbrace{\gamma_k \|f\|_{\mathcal{H}_{\tilde{k}}}^2}_{\mathcal{H}_{\tilde{k}} \text{ norm}}$$


 Search in RKHS defined by the kernel \tilde{k}

Bayesian Inference on Manifold [Laufer-Goldshtein et al., 2016a][Sindhwani et al., 2007]

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↓ Search in RKHS defined by the kernel \tilde{k}

$$\underbrace{p(f|P_L, H_L, H_U)}_{\text{Posterior}} \propto \underbrace{p(P_L|f, H_L)}_{\text{Likelihood Function}} \cdot \underbrace{p(f|H_L, H_U)}_{\text{Manifold-Based Prior}}$$

↓

f is a Gaussian Process
with Covariance \tilde{k}

Localization

MAP/MMSE estimator:

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

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

- The posterior $p(f(\mathbf{h}_t) | P_L, H_L, H_U)$ is a multivariate Gaussian with:

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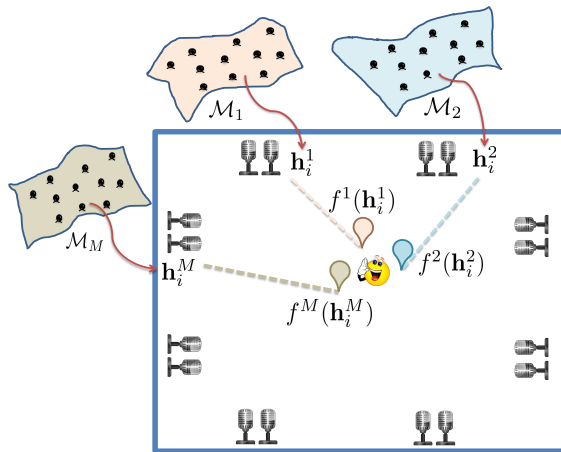
The MAP/MMSE estimator of $f(\mathbf{h}_t)$ is given by:

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Outline

- 1 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

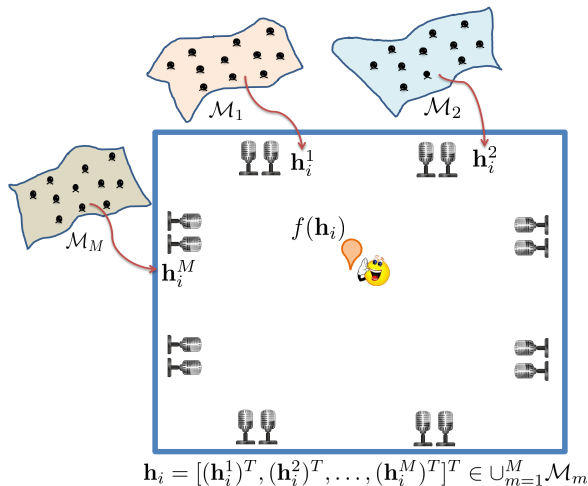
Source Localization with Ad Hoc Array [Laufer-Goldshtein et al., 2016d]



Each node:

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

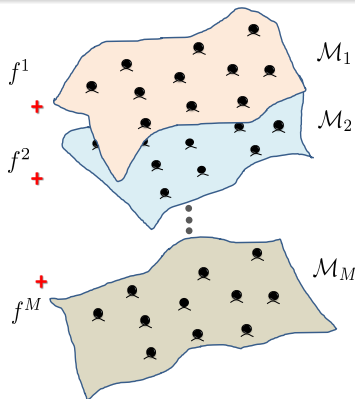
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How to **fuse** the different views in a unified mapping $f : \cup_{m=1}^M \mathcal{M}_m \mapsto \mathbb{R} ?$

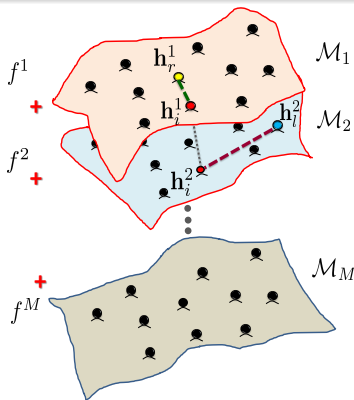
Multiple Manifold Gaussian Process (MMGP)

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



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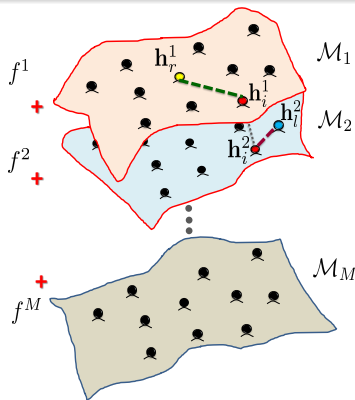


The covariance between $p_r = f(\mathbf{h}_r)$ and $p_l = f(\mathbf{h}_l)$

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

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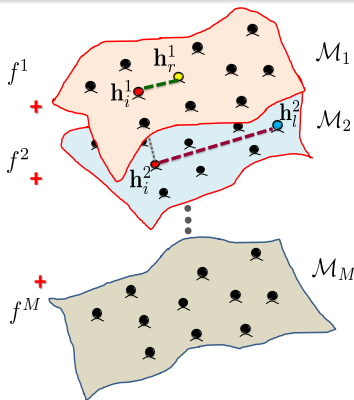


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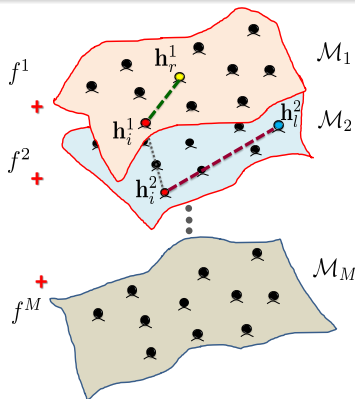


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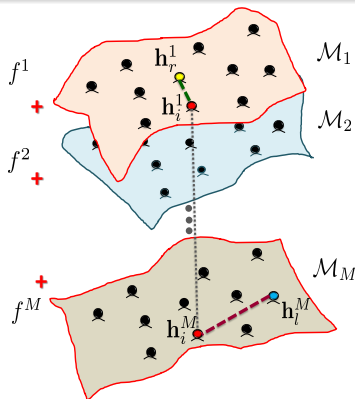


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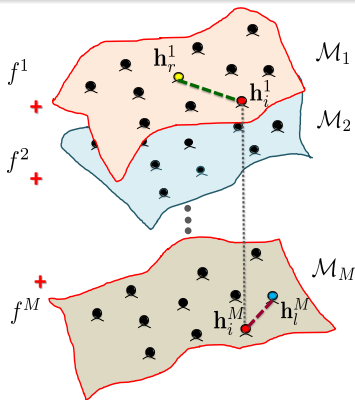


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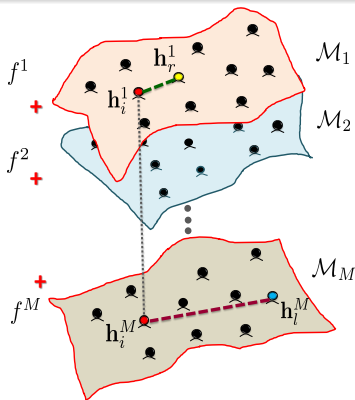


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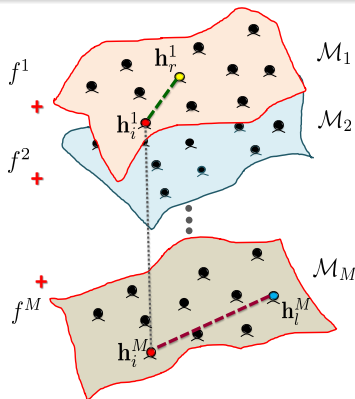


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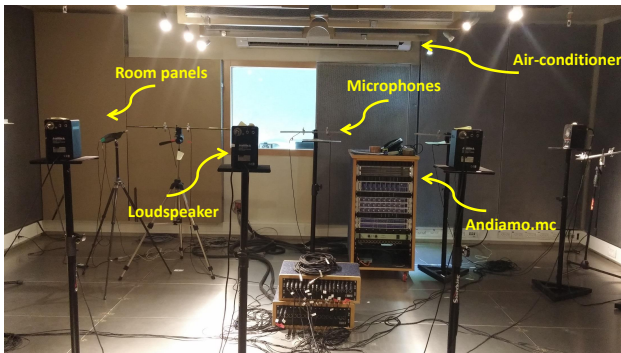
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Recordings Setup

Setup:

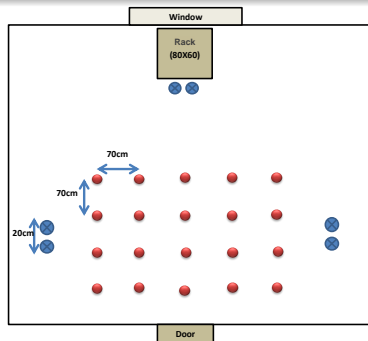
- Real recordings carried out at Bar-Ilan acoustic lab
- A $6 \times 6 \times 2.4\text{m}$ room controllable reverberation time (set to **620ms**)
- Region of interest: Source position is confined to a $2.8 \times 2.1\text{m}$ area
- 3 microphone pairs with inter-distance of 0.2m (**position unknown**)



Recordings Setup

Setup:

- Real recordings carried out at Bar-Ilan acoustic lab
- A $6 \times 6 \times 2.4\text{m}$ room controllable reverberation time (set to **620ms**)
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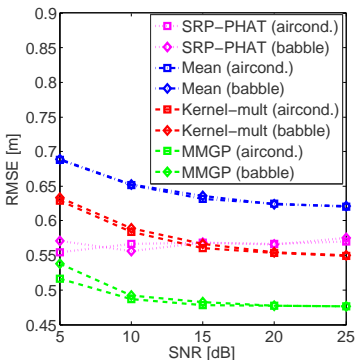
Experimental Results [Laufer-Goldshtein et al., 2016d]

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

- 1 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
- $\mathbf{h}(t) = [\mathbf{h}(t)^1]^T, \dots, [\mathbf{h}(t)^M]^T]^T$ - a concatenation of the RTF vectors from all nodes
- $p_c(t) = f(\mathbf{h}(t)), c \in \{x, y, z\}$ - mapping of the concatenated RTF vector to position (for brevity $p_c(t) \equiv p(t)$)

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

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

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

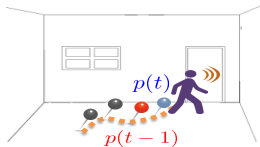
Standard (Nonlinear) State-Space Model

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Propagation Model

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



Bayesian Inference for Source Tracking

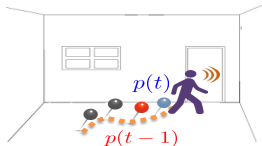
Standard (Nonlinear) State-Space Model

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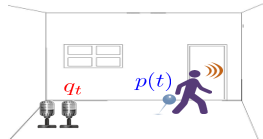
Propagation Model

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



Observation Model

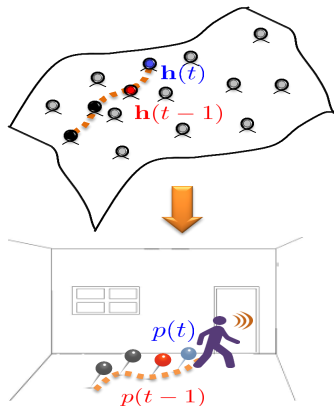
- Relate current position to measurements
- Examples: TDOA or steered response power readings
- Noise statistics is unknown



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

Propagation Model - Local

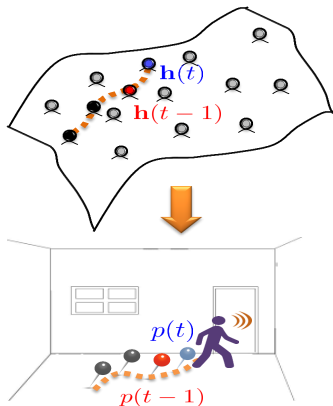
Transform nonlinear regression of high-dimensional RTFs to linear transition of source positions



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

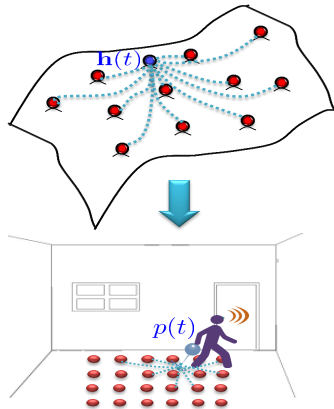
Propagation Model - Local

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:

$$\begin{bmatrix} p(t) \\ p(t-1) \end{bmatrix} \sim \mathcal{N} \left(\mathbf{0}, \begin{bmatrix} \tilde{\Sigma}_{t,t} & \tilde{\Sigma}_{t,t-1} \\ \tilde{\Sigma}_{t,t-1} & \tilde{\Sigma}_{t-1,t-1} \end{bmatrix} \right)$$

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

State Space Representation (2)

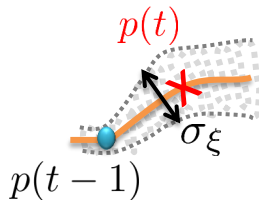
Propagation Model:

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
- $\xi_t \sim \mathcal{N}(0, \sigma_\xi^2)$ 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 Motion Model:

- $\bar{\mathbf{p}}_L = [\bar{p}_1, \dots, \bar{p}_{n_L}]^T$ - 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 $\bar{\mathbf{p}}_L$ 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:

$$p(t) | \bar{\mathbf{p}}_L \sim \mathcal{N} \left(\tilde{\Sigma}_{Lt}^H \left(\tilde{\Sigma}_L + \sigma^2 \mathbf{I}_{n_L} \right)^{-1} \bar{\mathbf{p}}_L, \tilde{\Sigma}_{t,t} - \tilde{\Sigma}_{Lt}^H \left(\tilde{\Sigma}_L + \sigma^2 \mathbf{I}_{n_L} \right)^{-1} \tilde{\Sigma}_{Lt} \right)$$

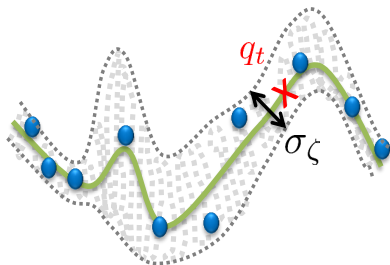
State-Space Representation (4)

Observation model:

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

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

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



The corresponding observation model:

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

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

Tracking Algorithm

Space-State Representation:

The proposed state-space model is given by:

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

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

Kalman Filter

Time Propagation

- Predicted Position:

$$\hat{p}(t|t-1) = b_t \cdot \hat{p}(t-1|t-1)$$

- Predicted Covariance:

$$\gamma(t|t-1) = g_t^2 \gamma(t-1|t-1) + \sigma_{\xi}^2$$

Measurement Update

- Kalman Gain:

$$\kappa(t) = \frac{\gamma(t|t-1)}{\gamma(t|t-1) + \sigma_{\zeta}^2}$$

- Updated position estimate:

$$\hat{p}(t|t) = \hat{p}(t|t-1) + \kappa(t)(q_t - \hat{p}(t|t-1))$$

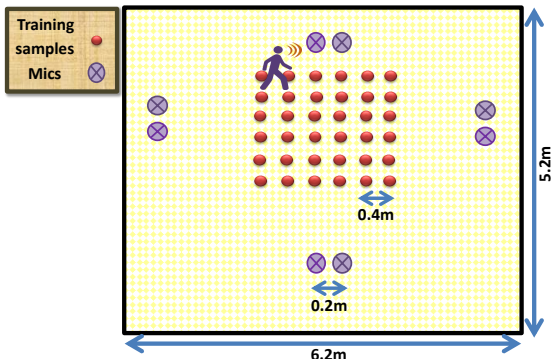
- Updated Covariance:

$$\gamma(t|t) = (1 - \kappa(t))\gamma(t|t-1)$$

Experimental Results

Setup:

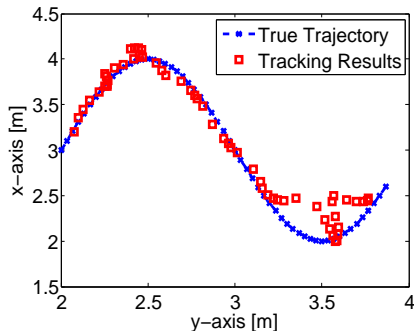
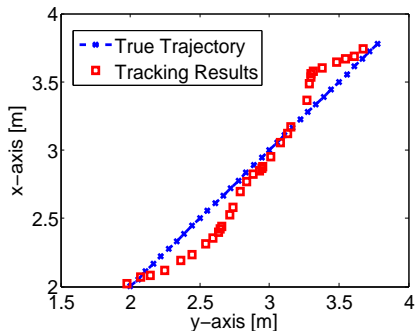
- A $5.2 \times 6.2 \times 3\text{m}$ room with $T_{60} = 300\text{ms}$
- $M = 4$ nodes with 0.2m distance between microphones
- **Region of interest:** a $2 \times 2\text{m}$ square region
- **Training:** 36 samples (0.4m resolution)



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 [Lauffer-Goldshtein 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]

Manifold Regularization

Measuring smoothness over \mathcal{M} :

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

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

which is a **global measure of smoothness** for f

- **Stokes' theorem** links gradient and Laplacian:

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

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

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

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 $S_{ii} = \sum_{j=1}^{n_D} W_{ij}$.
- Smoothness functional of G :

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

where $\mathbf{f}_D = [f(\mathbf{h}_1), \dots, f(\mathbf{h}_{n_D})]$

- It can be shown: [▶ Back](#)

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

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