

Echo-aware signal processing for audio scene analysis

Diego Di Carlo December 3, 2020

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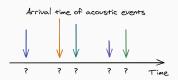
Université de Rennes 1, IRISA/INRIA, Panama research group

Acoustic Echo Estimation

Acoustic Echo Retrieval

Estimating early (strong) reflections for microphones recordings, i.e.,

$$\{\tilde{x}_i\}_i \longrightarrow \{\tau_i^{(r)},\alpha_i^{(r)}\}_{i,r}$$



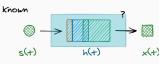
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Arrival time of acoustic events V V V V ? ? ? ? Time

Two scenarios:

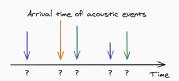


- intrusive or specific setups
- non-blind problem (Applications: sonar, measurements, etc.)

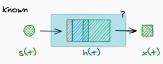
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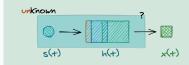
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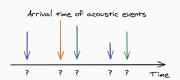
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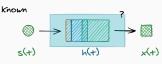
- passive and more common setups
- blind inverse problem (harder)
 (Applications: recording on smart speakers, etc.)

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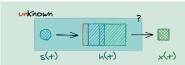
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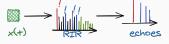


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- blind inverse problem (harder) (Applications: recording on smart speakers, etc.)

Our case: signal source and passive microphone array

Passive Acoustic Echo Retrieval

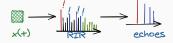
RIR-based approaches



RIR-agnostic approaches





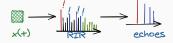


- 1. Discrete optimization \implies RIRs
- 2. Peak picking \implies Echoes

RIR-agnostic approaches



1. Direct estimation of $\left\{\tau_i^{(r)},\alpha_i^{(r)}\right\}$ e.g., with maximum-likelihood

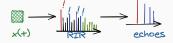


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- ✓ BCE is well and known studied
- ✓ reasonably good for some application [Crocco and Del Bue, 2016]

RIR-agnostic approaches



- 1. Direct estimation of $\left\{\tau_i^{(r)},\alpha_i^{(r)}\right\}$ e.g., with maximum-likelihood
- ✓ No full RIRs & no peak picking
 - → lower complexity
 - $\rightarrow \ \ \text{less hyperparameters}$



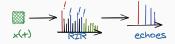
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- X Peak picking has hyperparameters
- X Issues due to discrete estimation

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RIR-agnostic approaches



- 1. Direct estimation of $\left\{\tau_i^{(r)},\alpha_i^{(r)}\right\}$ e.g., with maximum-likelihood
- ✓ No full RIRs & no peak picking
 - → lower complexity
 - → less hyperparameters
- exploratory
 (no standard solver, few works on audio)

Proposed appoach RIR-agnostic & continuous:

- 1. Learning-based approach
- 2. Analytical approach

Signal model

$$x_1 = h_1 \star x$$

$$x_2 = h_2 \star x$$

Ideas:

- 1. Echo TOAs \propto sampling frequency
- 2. Find echoes \rightarrow find sparse non-negative vectors h_1,h_2 of length L
- 3. Modeled as Lasso-like problem

$$\hat{h}_1, \hat{h}_2 \in \mathop{\arg\min}_{h_1,h_2 \in \mathbf{R}^n} \|x_1 \star h_2 - x_2 \star h_1\|_2^2 + \lambda \mathcal{P}(h_1,h_2) \quad \text{s.t.} \quad \mathcal{C}(h_1,h_2)$$

$$\mathcal{P}(h_1,h_2) \longrightarrow \text{sparse promoting regularizer} \qquad \mathcal{C}(h_1,h_2) \longrightarrow \text{constraints e.g.} \quad \underset{\text{anchor}}{\text{nonnegativity anchor}}$$

$$\checkmark$$
 [Tong et al., 1994] \checkmark [Lin et al., 2008] \checkmark [Aissa-El-Bey and Abed-Meraim, 2008] \checkmark [Kowalczyk et al., 2013] \checkmark [Crocco and Del Bue, 2016]



Convolving with filters:

$$\begin{aligned} h_2 \star x_1 &= h_2 \star h_1 \star x \\ h_1 \star x_2 &= h_1 \star h_2 \star x \end{aligned}$$

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Commutativity of convolution:

$$h_2 \star x_1 = h_2 \star h_1 \star x$$
$$h_1 \star x_2 = h_2 \star h_1 \star x$$

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Subtraction

$$\left. \begin{array}{l} h_2 \star x_1 = h_2 \star h_1 \star x \\ h_1 \star x_2 = h_2 \star h_1 \star x \end{array} \right\} \rightarrow x_1 \star h_2 - x_2 \star h_1 = 0$$

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Proposed approach: analytical & continuous

C. Elvira.

Observation 1: the cross-relation remains true in the continuous frequency domain

$$\mathcal{F}x_1\cdot\mathcal{F}h_2({}^n\!/{}_{\!F_s})=\mathcal{F}x_2\cdot\mathcal{F}h_1({}^n\!/{}_{\!F_s}) \qquad n=0\dots N-1$$



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Observation 3: $\mathcal{F}\boldsymbol{x}_i$ can be (well) approximated by DFT

$$\mathbf{X}_i = \mathrm{DFT}(x_i) \simeq \mathcal{F} \tilde{x}_i(nF_s) \qquad n = 0 \dots N-1$$

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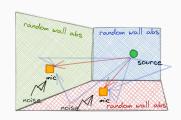
$$\mathbf{X}_i = \mathrm{DFT}(x_i) \simeq \mathcal{F} \tilde{x}_i(nF_s) \qquad n = 0 \dots N-1$$

$$\underset{h_{1},h_{2}\in\underset{\text{space}}{\text{measure}}}{\arg\min}\ \frac{1}{2}\left\|\mathbf{X}_{1}\cdot\mathcal{F}h_{2}(f)-\mathbf{X}_{2}\cdot\mathcal{F}h_{1}(f)\right\|_{2}^{2}+\lambda\left\|h_{1}+h_{2}\right\|_{\mathrm{TV}}\quad\text{s.t.}\ \begin{cases} h_{1}(\{0\})=1\\h_{l}\geq0 \end{cases}$$

 \sim Lasso, but $\mathcal{F}h_i(f)$ is a continuous function \rightarrow BLasso [Azais et al., 2015]

Syntetic Dataset at 16 kHz

- 2 microphones, 1 sound source (noise and speech)
- shoebox with random geometry
- $\mathcal{D}^{\mathrm{SNR}} \colon \, \mathrm{SNR} \in [0, 20] \; \mathrm{dB} \text{, } \mathrm{RT}_{60} = 400 \; \mathrm{ms}$
- $\mathcal{D}^{\text{RT}}_{60}$: $\text{RT}_{60} = [100, 1000] \text{ ms, SNR} = 20 \text{ dB}$



Baselines: discrete RIR-based methods based on LASSO

- BSN: Blind, Sparse and Non-negative¹
- IL1C: iteratively-weighted ℓ_1 constraint $^2 o$ State of the Art

hyperparameters and peak-picking tuned via cross-validation

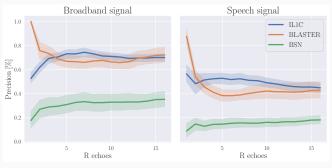
Proposed method: Blind and Sparse Technique for Echo Retrieval (Blaster)

¹[Lin et al., 2007]

²[Crocco and Del Bue, 2015]

□ Precision per # of echoes

Metric: Precision = how many estimated echoes are correct (within 2 samples)



$$(RT_{60} = 400 \text{ ms and SNR} = 20 \text{ dB.})$$

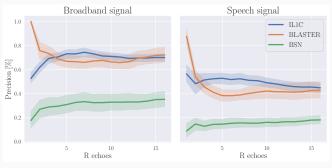
X Sensitive
kto # echoes

X Sensitive source signal

Good for
2 echoes
[Scheibler et al., 2018,
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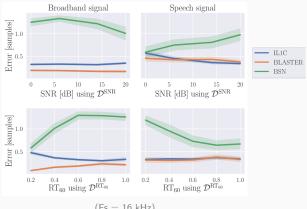
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Lerror per Dataset/Signal while recovering 7 echoes

Metric: RMSE on matched echoes = error on the correct guess



(Fs = 16 kHz)

✓Lower RMSF

Robustness to SNR and RT₆₀ Source signal dependent

Learning-based approach

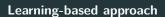
Idea:

- 1. Use virtually supervised deep learning models
- 2. Estimate first echo (simple but important) (used in the next section)
- 3. Only 2 microphones attending 1 sound source

Motivations:

- $\begin{array}{l} \bullet \quad x_i \to \tau_i^{(r)} \text{ is difficult, while } \tau_i^{(r)} \to x_i \text{ "is not"} \\ \to \text{ acoustic simulators: mic/src/room geometry } \longrightarrow \left\{\tau_i^{(r)}, \alpha_i^{(r)}\right\}, \ \tilde{h}_i, \quad \tilde{x}_i \end{array}$
- Acoustic simulator are "simple", versatile and fast

 → allow to create large dataset





Interchannel level and phase difference features from

$$R[f] = \underset{t}{\text{avg.}} \frac{X_2[f,t]}{X_1[f,t]} \approx \underset{t}{\text{avg.}} \frac{H_2[f]S[f,t]}{H_1[f]S[f,t]}$$

 \approx the relative transfer function \rightarrow remove source dependency

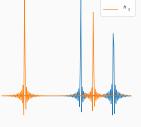
Outputs:

Inter and intra Time Difference of Arrivals (TDOAs)

 $\textbf{HP:} \ \mathsf{close}\text{-}\mathsf{surface} \ \mathsf{scenario} \colon \mathsf{first} \ \Leftrightarrow \mathsf{strongest} \ \mathsf{echo}$

Loss Function

- 1. RMSE (Multi-label regression) \rightarrow TDOAs
- 2. Gaussian log-likelihood $o \left\{\mu_{ au}, \sigma_{ au}^2\right\} \, \forall au \in \mathsf{TDOAs}$
- 3. Student log-likelihood $\rightarrow \{\mu_{\tau}, \lambda_{\tau}, \nu_{\tau}\} \ \forall \tau \in \mathsf{TDOAs}$





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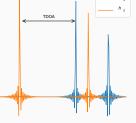
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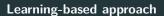
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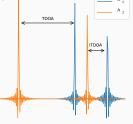
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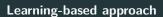
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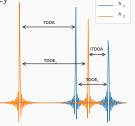
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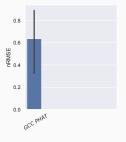


A Experimental results

Proposed Method: MLP, CNN, CNN $_{\!\mathcal{N}}$, CNN $_{\!\mathcal{T}}$

Baseline: GCC PHAT [Knapp and Carter, 1976]

 $\textbf{Metrics:} \ \, \text{normalized RMSE} \ \, (0 = \text{best fit, } 1 = \text{random fit})$



More echoes



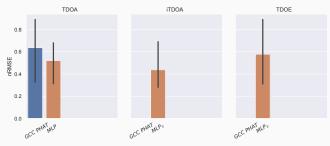


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 $\textbf{Metrics:} \ \, \text{normalized RMSE} \ \, (0 = \text{best fit, } 1 = \text{random fit})$



Observation:

 $\checkmark\,$ MLP outperforms GCC PHAT on TDOA estimation

More echoes



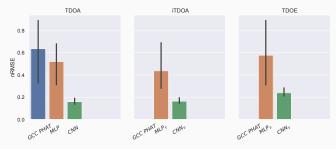


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- ✓ MLP outperforms GCC PHAT on TDOA estimation
- ✓ CNN outperforms MLP (lower error and smaller variance)





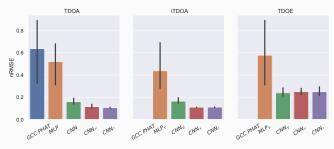


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Observation:

- ✓ MLP outperforms GCC PHAT on TDOA estimation
- ✓ CNN outperforms MLP (lower error and smaller variance)
- ✓ $\mathtt{CNN}_{\mathcal{N}}$ and $\mathtt{CNN}_{\mathcal{T}}$ outperform \mathtt{CNN} (lower error and smaller variance)
- X TDOA between DP and 1st echo more difficult

More echoes



References i



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