

# David Valdivia

2nd year PhD student, Audio signal processing and optimal transport, Univ. of Toulouse

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## Education

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**PhD in Audio Signal Processing**, IRIT / Univ. Toulouse, France (2024/27 expected)

- Research: applications of optimal transport for audio
- **MESRI-funded doctoral fellowship** (French Ministry of Higher Education and Research), obtained via competitive selection
- Supervisors: Cédric Févotte, Elsa Cazelles

**MSc in Artificial Intelligence**, PSL University, Paris, France (2023/24)

**Diplôme d'ingénieur (equiv. MEng)**, ISAE-Supaéro, Toulouse, France (2018/22)

## Work experience

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**Data analyst (apprentice), Qobuz** - Paris (Oct 2023 - Sep 2024)

- Implemented z-score-based anomaly detection to flag abnormal coupon usage patterns
- Built Looker dashboards on Snowflake (SQL + Python) regarding music streaming usage
- Documented the Snowflake database to support non-technical users

**Natural language understanding, SoundHound Inc.** - Paris (May 2022 - Mar 2023)

- Implemented C++ NLU interface based on ASR string outputs.
- Built and maintained multilingual commands (English/French/Spanish).
- Developed an internal website enabling spoken input to be routed to the C++ NLU component.

**Test software developer (intern), Ableton AG** - Berlin (Sep 2020 - Feb 2021)

- Developed hardware test software for a music controller's pad using C and Python.
- Implemented test cases and automation scripts to validate the pad's behavior.
- Wrote documentation describing each test.

## Publications

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**D. Valdivia, M. Renaud, E. Cazelles, C. Févotte**, "Audio signal interpolation using optimal transportation of spectrograms", 2025 IEEE Statistical Signal Processing Workshop (SSP), Edinburgh, United Kingdom, 2025, <https://arxiv.org/abs/2502.15430>

- Proposed unsupervised audio morphing method based on optimal transport barycenters of time-frequency energy distributions.
- Introduced a cost matrix that forbids large temporal displacements, leading to a sparse representation that scales to high-sample-rate and multi-second signals.
- **Best student paper award** (finalist): <https://2025.ieeessp.org/best-paper-awards/>
- **Code**: <https://github.com/davidvaldiviad/audio-signal-interpolation-ot>
- **Demo with sounds**: <https://davidvaldiviad.github.io/audio-signal-interpolation-ot/>
- **Poster** presented at IEEE SSP 2025, Edinburgh, United Kingdom
- **Talk** given at GDR-IASIS day on Optimal Transport, 2025, Lyon, France

## Skills

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**Computer stack:** Python (NumPy, SciPy, Matplotlib, Librosa), C, C++, CMake, Git, Bash, Linux/macOS, MATLAB, SQL

**Languages:** English - 990/990 TOEIC (April 2024), Spanish - bilingual, French - bilingual

**Audio practice:** musician, familiarity with timbre, pitch and harmonics, good listening skills

## Teaching experience

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**Algorithms and data structures in C**, ISAE-Supaéro, Toulouse, France

**Statistics**, ENSEEIHT, National Polytechnic Institute of Toulouse, Toulouse, France

**Introduction to Matlab**, ENSEEIHT, National Polytechnic Institute of Toulouse, Toulouse, France