# Umberto Cappellazzo

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# EDUCATION

#### University of Trento

Trento, Italy

Ph.D. in Information Engineering and Computer Science

Nov. 2021-ongoing

- Advisors: Daniele Falavigna, Alessio Brutti
- Research interests: continual learning for audio and speech processing; multi-modal continual learning;
   parameter-efficient transfer learning of audio/speech and audio-visual (e.g., Adapters, Mixture of Adapters, LoRA)

#### University of Padua

Padua, Italy

MSc in Telecommunication Engineering

2016-2019

- Advisors: Michele Rossi, Matteo Gadaleta
- Thesis Title: A Deep Learning-Based ECG Delineator: Evaluation and Comparison on Standard Databases

#### University of Padua

Padua, Italy

BSc in Information Engineering

2013-2016

- Advisor: Nicola Laurenti
- Thesis Title: Message Authentication over an Ideal or Noisy Channel

### Work Experience

Jelinek Summer Workshop on Speech and Language Technology (JSALT) Le Mans, France Junior researcher in the FST group

June 2023 –August 2023

- Junior researcher for the "Finite state methods with modern neural Architectures for speech applications and beyond" group at JSALT2023 in Le Mans, France. I worked on the integration of early-exit techniques to make the training and inference of CTC/MMI systems dynamical. Our group included people from Google, JHU, Telecom Paris to name a few. More information available here.

#### Imperial College London

London, UK

Research Intern, Audio-Visual PETL

February 2024 –May 2024

Supervisor: Stavros Petridis (ICL/Meta AI)

 The internship revolves around the study of parameter-efficient transfer learning (PETL) methods (e.g., adapters, mixture of adapters, LoRA) for audio-visual tasks.

## SKILLS

- LANGUAGES
- Programming Languages: Python (advanced), Java (basic), HTML (basic), Matlab (basic)
- ML/DL Toolkits/Libraries: PyTorch (advanced), NumPy, HF Transformers, Matplotlib. Good experience with CL libraries like Continuum and Avalanche
- ASR Frameworks: good experience with SpeechBrain and K2/icefall
- Other: Git, Docker

- Italian: mother tongue
- English: C1
- **TOEFL:** 100/120

# MENTORSHIP & PROFESSIONAL SERVICES

- Reviewer: ICASSP 2024 Workshop XAI-SA
- Co-supervision: I co-supervised a MSc student from the University of Bologna (thesis title: "On the use of Prompting for Fine-Tuning Neural models for Speech Processing")

#### **PUBLICATIONS**

- [1] G. A. Wright, U. Cappellazzo, S. Zaiem, D. Raj, L. Ondel Yang, D. Falavigna, and A. Brutti, "Training dynamic models using early exits for automatic speech recognition on resource-constrained devices", Self-supervision in Audio, Speech and Beyond (SASB) Workshop, ICASSP, 2024.
- [2] M. Yang, U. Cappellazzo, X. Li, S. Watanabe, and B. Raj, "Improving continual learning of acoustic scene classification via mutual information optimization", to appear at ICASSP 2024, 2024.
- [3] U. Cappellazzo, D. Falavigna, and A. Brutti, "An Investigation of the Combination of Rehearsal and Knowledge Distillation in Continual Learning for Spoken Language Understanding", *Interspeech (Poster)*, 2023.
- [4] U. Cappellazzo, D. Falavigna, A. Brutti, and M. Ravanelli, "Parameter-Efficient Transfer Learning of Audio Spectrogram Transformers", arXiv preprint, 2023.
- [5] U. Cappellazzo, E. Fini, M. Yang, D. Falavigna, A. Brutti, and B. Raj, "Continual Contrastive Spoken Language Understanding", arXiv preprint, 2023.
- [6] U. Cappellazzo, M. Yang, D. Falavigna, and A. Brutti, "Sequence-Level Knowledge Distillation for Class-Incremental End-to-End Spoken Language Understanding", *Interspeech (Oral)*, 2023.

See Google Scholar for my Google Scholar profile.