

## Research Interests

Neural Network Perception | Multimodal Machine Learning | Computer Vision

## Education

2019 – **Master of Science**, *Artificial Intelligence*, University of Erlangen-Nürnberg, Germany

Present Started with Communications and Multimedia Engineering. Switched to AI in 2021

2013 – 2018 **Bachelor of Science**, *Computer Science*, Sharif University of Technology, Iran

Scored **top 1%** in the national university entrance exam, entered the **highest-ranking** university in Iran

## Projects

1-Person **Functional Specialization between Music and Speech in Convolutional Neural Networks**, Master Project 2

- Trained 4 dual-task **neural networks** on music and speech to **compare perception of brain and CNNs**
- Designed **lesioning experiments** to quantify segregation between music and speech in CNNs
- Analyzed impact of batching method and network architecture on the level of task segregation

2-Person **Biological Learning**, Master Project 1

- Reproduced two **biologically-inspired neural networks** following Hebb's rule
- Adapted a convolutional version and scaled up the models to Tiny ImageNet dataset
- Demonstrated a decrease in performance and an increase in generalizability of Hebbian learning

3-Person **Musical Instrument Separation**, Music Processing course project at FAU

- Implemented two signal processing methods for musical instrument **source separation** in Python
- Used spoken speech spectrogram of singer as template matrix for non-negative matrix factorization

## Work Experience

2022 – 2023 **Fraunhofer IIS**, Student Research Assistant, Quantum Reinforcement Learning

- Developed a **reinforcement learning agent** environment for quantum circuit optimization
- Boosted circuit scanning speed by 57 times using local gate search

2020 – 2022 **Fraunhofer IIS**, Student Research Assistant, Deep Learning for Video Compression

- Trained >10 deep learning models to improve **hybrid video compression** methods
- Tested three convolutional models to predicting **optical flow** with TensorFlow and PyTorch

2019 – 2020 **International AudioLabs Erlangen, Fraunhofer IIS**, Student Research Assistant, Music and Speech Processing

- Developed a **speech alignment** method with Dynamic Time Warping with an error of .25 seconds
- Wrote a Jupyter notebook to compare performance of filterbanks in SciPy and Librosa

## Technical Skills

**Programming** Python, Java, MATLAB, R

**Deep Learning** PyTorch, TensorFlow, Keras

**Other** Pandas, Librosa, TensorBoard | Excel, Word, PowerPoint,  $\LaTeX$  | Linux, Windows | Git

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## Languages

- **English:** Fluent
- TOEFL iBT: 103/120
- **Persian:** Native
- **German:** Currently at B2 level