Ramin Anushiravani

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Skills

- Deep Learning Frameworks: PyTorch, TensorFlow, Keras, TFLite, Sklearn, HuggingFace
- Foundation and Multimodal AI: <u>Classical ML</u>, <u>Self-Supervised Learning</u>, <u>Reinforcement learning</u>, <u>Few-shot</u> inference, Vision Transformer, EfficientNet, Flamingo, Audio tokenizer, Wav2Vec, Model soups, Conformer, YamNet, Agentic AI
- Search: RAG, Vectorized Search, Entity recognition, query understanding, recommendation systems
- Model optimization: LoRA, prompt engineering and instruction fine-tuning, quantization, knowledge distillation, pruning
- Audio Processing: Signal processing, <u>Blind source separation (NMF)</u>, dereverberation, denoising, feature engineering, <u>3D</u> audio
- MLOps and deployment: AWS (S3, EC2, SageMaker Pipelines), MLFlow, Flask, FastAPI, GitHub Actions, Dockers, Optuna

Experience

Precision Neuroscience, New York, NY - Staff Machine Learning Scientist - 11/2023 to Present

- Implemented a novel **transformer-based multitask spatiotemporal foundation model** for ECoG data, pretrained using both contrastive (SimCLR with InfoNCE) and reconstruction-based (Masked Autoencoders, MAE) **self-supervised** objectives, then fine-tuned on supervised tasks to produce high-quality embeddings.
- Developed real-time few-shot inference for hand gesture classification from motor cortex activity, achieving 85% F1 score.
- Developed real-time multitask cursor control regression with custom learnable loss function models achieving 79% in R^2.
- Developed data and electrode patch selection methods based on sample impact on training loss and electrode patch impact on validation performance to improve model decodability and generalization.
- Reduced model latency by 4× and compressed model size by 20x with minimum performance loss, using neural architecture search and quantization for custom hardware.
- Developed model interpretability tools using saliency and attention maps to assess electrode contributions to decoding.
- Fine-tuned SOTA ASR models to annotate speech data collected from operating rooms and align it with neural data.
- Built scalable and reusable machine learning and signal processing pipelines to process and visualize terabytes of high-dimensional time series data.

United HealthGroup, San Mateo, CA - Sr Principal ML Engineer - 01/2021 to 10/2023

- Led a team of data and ML engineers to develop, and maintain text understanding models for consumer search products.
- Developed and maintained multilingual auto-correct using character-level bidirectional LSTMs and N-grams.
- Developed auto-complete and auto-suggest algorithms using FSTs and fine-tuned GPT-2 on healthcare queries.
- Created Al-powered search tools **serving 40 million active members** directly driving significant improvements in click-through rates and user satisfaction, leading to a 5x increase through A/B testing.
- Pre-trained and fine-tuned several encoders (BERT, RoBERTa, DistillBERT) to generate **sentence embeddings** to enable vectorized search functionality and other downstream tasks such as entity recognition.
- Benchmarked **ASR models**, including wav2vec 2.0 and NVIDIA NeMo, and deployed **conversational AI agents** for call routings and abstractive summarization using T5, enhancing customer service efficiency.

CurieAI, Menlo Park, CA - Machine Learning Scientist - 04/2018 to 01/2021

- Developed novel hybrid on-device and cloud **audio understanding** for monitoring chronic respiratory diseases in challenging acoustic environments, achieving an 80% increase in recall and an 86% improvement in precision over existing licensed models.
- Spearheaded **machine learning life cycles**, from data collection and annotation to signal processing and continuous model training, driving significant improvements in model performance and efficiency.
- Developed an AI-driven course of action recommendation system, leveraging patient history and engagement data.

DSP Concepts, Santa Clara, CA - *Algorithm Engineer* - 09/2017 to 04/2018

- Engineered noise reduction and dereverberation algorithms for improving wake-word detection on smart speakers.
- Automated testing protocols for audio algorithms, ensuring robust performance across various acoustic conditions.

Dolby Labs, San Francisco, CA - Audio Engineer - 09/2016 to 09/2017

- Developed an automated system for detecting infringements of Dolby audio codecs.
- Delivered expert tutorials and white papers on cutting-edge audio processing and deep learning, educating senior executives on emerging technologies.
- Managed extensive patent portfolio, drafting claims and responding to complex office actions.

Prior roles: Adobe (Audio editing), GN-ReSound (Hearing aids), Advanced Digital Science Center (Microphone arrays, Singapore)

Written Work & Publications

- Granted: Sound Enhancement through Reverberation Matching
- Granted: Methods for Explainability of Deep-Learning Models
- Granted: Intelligent Health Monitoring
- Granted: Design of Stimuli for Symptom Detection
- Pending: <u>Domain aware autocomplete</u>
- Pending: Graph-based data compliance using natural language text
- Pending: Interactive map-based visualization system related to multichannel search for complex search domains
- Pending: Machine learning techniques for generating domain-aware query expansions
- Pending: Multi-channel search and aggregated scoring techniques for complex search domains
- Pending: <u>Text embedding-based search taxonomy generation and intelligent refinement</u>

What is attention?	How does ChatGPT work?	Self-supervised learning
3D Audio	Fine-tuning LLMs	Sound Source Localization
<u>Transformers</u>	Neural Architecture Search	Acoustics Matching of Recordings
Example Based Audio Editing	Knowledge Distillation	Quantization

Education

08/2011 - 12/2016

M.S. & B.S., Electrical & Computer Engineering, University of Illinois at Urbana-Champaign (GPA: 3.97/4, 3.86/4)