

# TANMAY KHADELWAL

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## EDUCATION

### New York University (NYU), Courant Institute of Mathematical Sciences

Sep 2023 - May 2025

Master of Science in Computer Science (Concentration in AI) **GPA: 4.0/4.0**

New York, USA

**Relevant Courses:** Deep Learning (Yann LeCun), Natural Language Processing, Machine Learning, Computer Vision, Reinforcement Learning

### Birla Institute of Technology and Science (BITS), Pilani

Aug 2017 - Jul 2021

Bachelor of Engineering in Electrical Engineering **GPA: 8.43/10.0 (First Division)**

Pilani, India

**Relevant Courses:** Information Retrieval, Data Mining, Neural Networks and Fuzzy Logic, Object-Oriented Programming, Data Structures

## EXPERIENCE

### Amazon

June 2024 - Aug 2024

Applied Scientist Intern

Sunnyvale, California

- Improved **real-time search and recommendation** for Amazon Music by leveraging a **transformer-based two-tower network** with **LoRA adaptors** and **mixture of experts (MoE)**, achieving a **61.5% accuracy boost** on low-performing queries to enhance relevance, user engagement, and satisfaction across **millions of searches**.
- Boosted **search ranking** performance with a **popularity-weighted contrastive loss** function, leading to a **25.17% increase in Recall@20** by incorporating **synthetic data generation** and **user search logs**.
- Deployed a **guardrailed** system on AWS, leveraging **large language model (LLM) fine-tuning** with **DPO** for user-aligned responses and **RAG** for key phrase generation and indexing from **lyrics**, achieving **sub-80ms** response time.

### Music and Audio Research Lab (MARL)

Sep 2023 - Present

Machine Learning Research Assistant

New York, USA

- Developing a lightweight latent **diffusion transformer (DiT)** for **text-to-audio generation** on **resource-constrained devices**, leveraging **T5 embeddings**, **LoRA adaptors** and **post-training quantization (PTQ) methods** for efficient, high-quality synthesis.
- Contributed to open-source **Python** library: **Soundata** [\[Source\]](#) [\[Paper\]](#) with **47k downloads**, introducing data loaders, interactive visualization tools, and standardized usage for enhanced reproducibility; published in **journal of open source software (JOSS)**.

### Fortemedia Inc.

Sep 2021 - Jul 2023

Machine Learning Engineer

Singapore, Singapore

- Devised **multi-task learning (MTL)** framework [\[Paper\]](#) with two-stage **semi-supervised learning (SSL)** system for **speech-to-text software**, employing **transformer** and **conformer** to model global and local sequences; Improved PSDS by **45.5%**.
- Engineered **low-complexity** acoustic event detection system [\[Paper\]](#) for **speech recognition** embedded devices, integrating **attention modules** and **Bi-GRU**, resulting in a **34.1%** improvement in PSDS metrics while reducing model complexity by **27.6%**.
- Developed and **deployed** a scalable, **real-time** infant cry detection system [\[Paper\]](#) utilizing **depthwise-separable convolutions** and **REST APIs** for backend services; Accomplished an F-score of **0.738** on the curated dataset.

### Bajaj Finserv Health Limited

Jul 2020 - Aug 2021

Software Developer Intern

Pune, India

- Engineered personalized medication **recommendation system** with past history and patterns-based quick suggestions using **Elasticsearch** and **SQL**, reducing E-consult time with **response time < 50ms**.
- Pioneered highly scalable **microservices** for a doctor's practice management system [\[Link\]](#) in **SpringBoot** and **NodeJs**, combined with **MongoDB** and **Redis**, using **Docker** and **Kubernetes** for **1000+ daily** E-consult users.

## PUBLICATIONS

- "A Multi-Task Learning Framework for Sound Event Detection using High-level Acoustic Features", **INTERSPEECH** [\[DOI\]](#)
- "Audio-Tagging Assisted Sound Event Detection using Weak Generated Labels and Frequency Dynamic Convolutions", **IEEE SSP** [\[DOI\]](#)

## TECHNICAL SKILLS

**Programming Languages:** Python, C++, Java, JavaScript, Typescript, Matlab, Scala, ML

**Tools:** GitHub, Apache Kafka, Spark, Hadoop, Airflow, Kubernetes, Docker, Git, GPU, Nvidia CUDA, MySQL, JIRA

**Frameworks:** PyTorch, TensorFlow, PyTorch Lightning, Keras, OpenCV, Hugging Face Transformers, Scikit-learn, Matplotlib, NumPy, Pandas, ONNX, NLTK, spaCy, MLflow, XGBoost, Prometheus, Grafana, OpenAI API, Microsoft Azure, AWS (Sagemaker, Lambda, EC2)

## PROJECTS

- Built a deep learning model with **U-Net** and **SimVP** for **22nd- video frame prediction** and **segmentation** from first 11 frames, achieving **0.455 IoU** using **pseudo-labeling** and **fine-tuning** to enhance accuracy on unlabeled data; Attained **1st position**. [\[Code\]](#) [\[Report\]](#)
- Established a **multi-modal visual question answering (VQA)** system using **CNNs**, **LSTMs**, and a **stacked attention network (SAN)**, achieving state-of-the-art accuracy **80.5%** on the VQA-v2 dataset and the Radiology dataset. [\[Code\]](#) [\[Report\]](#)
- Formulated a sound event detection (SED) system leveraging **bi-directional encoder representation from audio transformers (BEATS)** embeddings with **frequency dynamic convolutions (FDY-CNN)**; achieved **4th position**. [\[Report\]](#)