

# Sreyan Ghosh

☎ +1 240 899 7685 | ✉ [sreyang@umd.edu](mailto:sreyang@umd.edu) | 🔗 [LinkedIn](#) | 🐙 [GitHub](#) | 🌐 [Website](#) | 📄 [Google Scholar](#) | 📍 College Park, USA

## EDUCATION

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**University of Maryland** (Advised by [Dr. Dinesh Manocha](#))

College Park, USA

*Ph.D. in Computer Science; GPA: 3.9/4.00*

*Aug 2022 – May 2027*

*M.S. in Computer Science; GPA: 3.9/4.00*

*Aug 2022 – May 2024*

**Christ University**

Bangalore, India

*B.Tech in Computer Science and Engineering; GPA: 8.7/10*

*June 2016 – Jun 2020*

## CONFERENCE PAPERS (IN CHRONOLOGICAL ORDER)

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[AdVerb: Visually Guided Audio Dereverberation](#)

Sanjoy Chowdhury\*, *Sreyan Ghosh\**, Subhrajyoti Dasgupta, Anton Ratnarajah, Utkarsh Tyagi, Dinesh Manocha

**ICCV 2023**

[MMER: Multimodal Multi-task Learning for Speech Emotion Recognition](#)

*Sreyan Ghosh*, Utkarsh Tyagi, S Rameswaran, Harshvardhan Srivastava, Dinesh Manocha

**InterSpeech 2023**

ACLM: A Selective-Denoising based Generative Data Augmentation Approach for Low-Resource Complex NER

*Sreyan Ghosh\**, Utkarsh Tyagi\*, Manan Suri, Sonal Kumar, S Rameswaran, Dinesh Manocha

**ACL 2023**

[BioAug: Conditional Generation based Data Augmentation for Low-Resource Biomedical NER](#)

*Sreyan Ghosh\**, Utkarsh Tyagi\*, Sonal Kumar\*, Dinesh Manocha

**SIGIR 2023**

[MAST: Multiscale Audio Spectrogram Transformers](#)

*Sreyan Ghosh\**, Ashish Seth\*, S. Umesh, Dinesh Manocha

**IEEE ICASSP 2023**

[SLICER: Learning universal audio representations using low-resource self-supervised pre-training](#)

Ashish Seth\*, *Sreyan Ghosh\**, S. Umesh, Dinesh Manocha

**IEEE ICASSP 2023**

[data2vec-aqc: Search for the right Teaching Assistant in the Teacher-Student training setup](#)

Lodagala V S V Durga Prasad\*, *Sreyan Ghosh\**, S. Umesh

**IEEE ICASSP 2023**

[PADA: Pruning Assisted Domain Adaptation for Self-Supervised Speech Representations](#)

Lodagala V S V Durga Prasad, *Sreyan Ghosh*, S. Umesh

**IEEE SLT 2022**

[CCC-wav2vec 2.0: Clustering aided Cross Contrastive Self-supervised learning of speech representations](#)

Lodagala V S V Durga Prasad, *Sreyan Ghosh*, S. Umesh

**IEEE SLT 2022**

[Span Classification with Structured Information for Disfluency Detection in Spoken Utterances](#)

*Sreyan Ghosh*, Sonal Kumar, Yaman Kumar Singla, Rajiv Ratn Shah, S. Umesh

**Interspeech 2022**

[DeToxy: A Large-Scale Multimodal Dataset for Toxicity Classification in Spoken Utterances](#)

*Sreyan Ghosh*, Samden Lepcha, Sakshi, Rajiv Ratn Shah, S. Umesh

**Interspeech 2022**

[End-to-end Named Entity Recognition from English Speech](#)

Hemant Yadav, *Sreyan Ghosh*, Yi Yu, Rajiv Ratn Shah

**Interspeech 2022**

## JOURNAL AND WORKSHOP PAPERS (IN CHRONOLOGICAL ORDER)

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[Decorrelating Feature Spaces for Learning General Purpose Audio Representations](#)

Sreyan Ghosh\*, Ashish Seth\*, S. Umesh

**IEEE JSTSP Special Issue on Self-Supervised Learning for Speech and Audio Processing**

[DeLoRes: Decorrelating Latent Spaces for Low-Resource Audio Representation Learning](#)

Sreyan Ghosh\*, Ashish Seth\*, S. Umesh

**SAS Workshop @ AAAI 2022**

[Cisco at SemEval-2021 Task 5: What's Toxic?: Leveraging Transformers for Multiple Toxic Span Extraction from Online Comments](#)

Sreyan Ghosh, Sonal Kumar

**SemEval-2021 @ ACL 2021**

[Cisco at AAAI-CAD21 shared task: Predicting Emphasis in Presentation Slides using Contextualized Embeddings](#)

Sreyan Ghosh, Sonal Kumar, Harsh Jalan, Hemant Yadav, Rajiv Ratn Shah

**CAD-21 @ AAAI 2022**

## PROFESSIONAL WORK EXPERIENCE

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### **NVIDIA**

Bangalore, India

*Research Scientist*

*April 2022 – August 2022*

- Worked as a senior solutions architect in the professional services team at NVIDIA. Responsible for delivering deep-learning based NLP solutions to NVIDIA's premier customers around the globe.
- Contributed to AI R&D at NVIDIA. Published 2 papers at **IEEE SLT 2022**.

### **Cisco Systems**

Bangalore, India

*Software Engineer II*

*Aug 2020 – March 2022*

- Worked as a senior software engineer in the automation and orchestration team under the Customer Experience BU. Built network assurance solutions for Cisco's telecom customers, leveraging state-of-the-art algorithms for anomaly detection at scale. Built a critical component in Cisco's first telemetry-based network assurance solution.
- Lead the development of an AI-based network security system for one of Cisco's telecom customers.
- Was part of the AI team that developed Cisco's first contact center solution, leveraging state-of-the-art NLP algorithms.
- Contributed to AI R&D at Cisco by representing Cisco at various conferences.

## RESEARCH AND TEACHING EXPERIENCE

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### **Gamma Lab @ UMCP**

College Park, Maryland, USA

*Research Assistant*

*Fall 2022 – Present*

- My primary research focuses on low-resource (labeled data and compute) learning with applications in speech, NLP or vision. In this area, I solve problems using self-supervised learning, synthetic data augmentation, etc.
- Advised by [Dr. Dinesh Manocha](#).

### **University of Maryland**

College Park, Maryland, USA

*Teaching Assistant*

*Fall 2022 – Fall 2022*

- TA for Introduction to NLP (CMSC 470) for Fall 2022

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College Park, Maryland, USA

*Teaching Assistant*

*Fall 2022 – Fall 2022*

- TA for Introduction to NLP (CMSC 470) for Fall 2022

### **Speech Lab, Indian Institute of Technology Madras**

Chennai, India

*Project Associate (Research)*

*June 2021 – August 2022*

- Worked under the supervision of **Dr. S. Umesh** in the area of self-supervised learning for Speech and Audio processing. Exploring techniques to devise lighter-weight models and efficient algorithms to make supervised and self-supervised learning in speech and audio amenable to resource-constrained scenarios (both data and compute). Paper accepted to **SAS Workshop at AAAI 2022**, **Interspeech 2022**, **IEEE JSTSP Special Issue** and **IEEE SLT 2022**.

### **MIDAS Labs, IIIT-Delhi**

Delhi, India

*Research Assistant (part-time)*

*December 2019 – August 2022*

- Worked under the supervision of **Dr. Rajiv Ratn Shah** in the areas of Speech and Natural Language Processing. Worked on building ASR systems for low-resource Indian Languages (mono and multi-lingual) and Indian Accented English which served as a critical component for other systems built by the lab. Worked in the field of content moderation in modalities of both speech and text. Currently exploring multi-modal techniques for identifying disfluencies in spoken utterances. Published papers at **AAAI 2021**, **ACL 2021** and **Interspeech 2022**.

## INTERNSHIPS

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<b>Adobe</b> <i>Research Scientist Intern</i>	Seattle, WA, USA May 2023 – Present
<ul style="list-style-type: none"> <li>• Working as a research scientist intern at Adobe in the multi-modal understanding group using Large Language Models.</li> </ul>	
<b>Google Summer of Code</b> <i>Open Source Developer</i>	Remote April 2022 – August 2022
<ul style="list-style-type: none"> <li>• Working on building deep learning based NLP (speech and text) notebooks using Tensorflow and Keras.</li> <li>• Link to PRs and code contributed on personal website.</li> </ul>	
<b>Cisco Systems</b> <i>Software Developer Intern</i>	Bangalore, India January 2020 – June 2020
<ul style="list-style-type: none"> <li>• Worked on a project, End-to-End Named Entity Recognition from English Speech under the guidance of <b>Dr. Rajiv Ratn Shah</b> as part of my bachelors thesis. Paper accepted at <b>Interspeech 2020</b>.</li> </ul>	
<b>MIDAS Labs, IIIT-Delhi</b> <i>Research Intern</i>	Delhi, India January 2020 – June 2020
<ul style="list-style-type: none"> <li>• Worked on building a VOIP (Voice Over IP) Traffic Analyzer to detect anomalous SIP messages using machine learning.</li> </ul>	
<b>Noodle.ai</b> <i>Data Science Intern</i>	Bangalore, India December 2019 – December 2019
<ul style="list-style-type: none"> <li>• Worked on multivariate time-series anomaly detection in high-frequency IoT sensor data obtained from steel manufacturing machines.</li> </ul>	
<b>TEG Analytics</b> <i>Data Science Intern</i>	Bangalore, India April 2019 – May 2019
<ul style="list-style-type: none"> <li>• Worked under the healthcare intelligence division to provide insights from insurance plan enrollment data, for private insurance companies in the US.</li> <li>• Used Machine Learning and Deep Learning techniques to predict plan enrollment for insurance companies.</li> </ul>	

## AWARDS & ACHIEVEMENTS

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**Recognised by Cisco CX CTO** and higher management on multiple occasions for my research and innovation initiatives.

Awarded the **Graham Bell Award** for being one of the most competitive undergraduates to have graduated in the year 2020.

**Winner of Cisco Collab Hacks 2020.**

**Winner of P&G Global Innovation Challenge 2020.**

Appeared on the cover page of Analytics India Magazine twice for winning national level hackathons in 2020 (TEG Analytics and Uber Hackathon)

**Winner of Hindustan Unilever BFS Technology Hackathon.**

**Winner of various inter-college and intra-college hackathons sponsored by MNCs and the Government (Including a bronze medal at Kaggle).**

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## MAJOR PROJECTS

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**LAPE** | [GitHub](#)

- LAPE is an easy-to-use toolkit for audio processing. In its initial release, LAPE supports Self-Supervised Learning (SSL)-based Upstream Pre-training and Downstream Fine-tuning. LAPE, originally introduced in this paper, integrates all our research on low-resource audio processing in one unified framework. We open-source LAPE to promote more research in this space.

**Intelligent Minutes of the Meeting**

- Used Automatic Speech Recognition and Speaker Diarization together with transformer-based text summarization and Named-Entity-Recognition to create a web app for intelligent Minutes of the Meetings.
- Application has capabilities to take meeting recordings and output summary, diarized transcripts, and key points from the meeting, including dated items and to-dos.

## SKILLS

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**Programming:** Python, Java, C, MySQL, HTML, CSS

**Frameworks:** PyTorch, Keras, Tensorflow, Django, Flask, Spark

**Tools:** GIT, Android, Tableau, Power BI, AWS, GCP, Rest API, Docker, K8s

**Concepts:** Speech and Natural Language Processing, Software Development, Functional programming, Object-oriented programming, Machine Learning, Deep Learning, Image Processing, Cloud Computing

## COMMUNITY SERVICE

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**Reviewer for:** EMNLP 2023, ACL 2023, ICASSP 2023, InterSpeech 2023, AAAI 2023, ACL 2021, HASOC 2021, Electronic Letters (IET)

**Team Lead and Co-founder for:** Neuron, Christ University's first AI club focused on research, served as the first Vice President of the club.

**Lecturer of:** SLP at University of Buffalo, New York.