RAJ PRAKASH GOHIL

☑ rajgohil021@gmail.com 🔊 +91-9172495348

EDUCATION

Indian Institute of Technology, Kanpur Kanpur, UP

Master of Technology (Research) in Electrical Engineering

2019-2021

Vidyalankar Institute of Technology, Mumbai University Mumbai, India

Bachelors of Technology in Electronics Engineering

2014-2018

WORK EXPERIENCE

Samsung RnD Institute Bangalore India

Senior Machine Learning Engineer

August 2022 - Present

- Independent researcher to implement and improve the voice assistant in the day to day life.
- Working on end-to-end conformer based speech recognition for Indian Languages, achieving 8% WER for day to day conversation. The model can work on device for offline speech recognition.
- Worked and implemented tasks like Diarization, post stitching of the ASR output, Speaker attributed ASR, ASR Error Correction.
- o Algoritms like RPN, Spectral clustering were used for Diarization. Models like BERT were trained for ASR error correction.
- Worked on POC for Code Switched ASR
- o Commercialization of Streaming ASR for server and On Device for Indian English.
- o Worked on DIPLACE Challenge (2nd Place), Low Resource Speech2Speech (1st Place, Paper details mentioned Below)
- o Contributed to various challenges hosted Interspeech, ICASSP, DCASE.
- Major contribution in developing far field source localization.
- o Dockerized various algorithm for the usage by the other team members.

Fraunhofer IDMT Germany

Early Stage Researcher in Hearing, Speech and Audio Technology

Januray 2022 - August2022

- o Implement various state of the art algorithm for speech signal processing.
- Worked on Audio Event Detection and Localization using various directional features.
- Various localization models like SELD, and attention based models are implemented.

Gnani AI India

Speech Processing

August 2021 - October 2021

- o To develop a system for voice cloning to adapt a unseen speaker using less than 2mins of reference audio.
- Speaker embeddings like xvectors, dvectors and others were studied to use it for speaker adaptation task.
- Various Text-to-Speech models like Tacotron, Fastspeech, Melgan were explored so as to get clean and cloned speech from the text.

SONY Research India India

Speech Processing Intern

October 2020 - July 2021

- Built a custom data-set for Hindi-English mix audio.
- o Build an end to end ASR for Hindi-English with custom acoustic and language model based on Chain models in Kaldi.
- o Transformer based LM, CNN based AM is built using and pytorch.
- Various other toolkits like wav2letter, DeepSpeech was also implemented.
- Various pre-processing techniques like noise cancellation, speech activity detection, speaker diarization was also studied and used.
- Tried using various features like Wav2Vec2 in place of MFCC.
- Python wrappers for kaldi like PyKaldi were also studied.
- The WER on the validation data was 13% and the WER on the real dataset (Bollywood-Movies) was 35%.

NMIMS India

Visiting Faculty

- o Taught a course on Speech Signal Processing of 60hrs spanned over 4 months.
- Delivered knowledge of speech signals, machine learning, deep learning, speech recognition, how to read research papers.
- The students at the end of the course were able to successfully build a Kaldi based ASR model with in-depth understanding of Language model, acoustic model and their interation.
- Also conducted a viva-voice session where the students presented summary of all the selected speech processing task by
 efficiently reading the research papers based on the tasks.

Publications

Ensemble of Incremental System Enhancements for Robust Speaker Diarization in Code-Switched Real-Life Audios SPECOM 2023

- o Study of various pre-trained speaker embedding and vad models.
- Applied novel spectral clustering for speaker diarization.
- Secured second place in the DISPLACE Challenge conducted at Interspeech 2023

SRI-B's Systems for IWSLT 2023 Dialectal and Low-resource Track: Marathi-Hindi Speech Translation

Association for Computational Linguistics (IWSLT SYSTEM PAPER)

2023

2022

Joint DOA Estimation in Spherical Harmonic Domain using Low Complexity CNN **SPCOMM**

o Joint DOA estimation is done using low complexity Spherical Harmonics features.

- o CNN based learning is adapted so as to capture all information from the features.

An outlier robust Gross error has been used to measure the accuracy of the DOA estimation.

National Conference on Communications(*NCC-2021*)

July 2021

- o A direction based features was extracted from audio simulated using Uniform and Co-Prime conformal microphone array(C3M) at different SNR for all the source angles.
- Joint estimation of azimuth and elevation has been done using robust direction based GCC-PHAT features.

Learning Based Method for Robust DOA Estimation using Co-prime Circular Conformal Microphone Array

o A state-of-art method to calculate DOA based on Least Squares was compared with the existing DNN based classifier and with our CNN based classifier.

LEARNING BASED DOA ESTIMATION IN ADVERSE ACOUSTIC ENVIRONMENT USING CO-PRIME CIRCULAR MICROPHONE ARRAY

Asia-Pacific Signal and Information Processing Association

December 2020

- A direction based features was extracted from audio simulated using Uniform and Co-Prime microphone array at different SNR for all the source angles (0 to 360).
- A state-of-art method to calculate DOA based on Least Squares was compared with the existing DNN based classifier and with our CNN based classifier.
- o Technology and Software used: Matlab, Python, DNN, Multi-Layer perceptron, CNN, Multi-Class Classification, Tensorflow,data generation, MicrophoneArray.

SKILLS & OTHERS

Data Sciene Library: Pandas, NumPy, SciPy, SciKit-Learn, Matplotlib, ggplot2

Machine Learning: Classification, Regression, Neural Networks, GANs, Autoencoders, Data Visualization

Certifications: Deep Learning AI and Tensorflow and Gudiede projects based on time-series and GAN from Coursera

POSITION of RESPONSIBILITY

Overall Placement Cordinator

Conduct the overall placement and internship drive in the campus

- Led a 4-tier team of 150+ members responsible for the placements of 1050 and internship of 750 students.
- Buddy Mentorship Program Decentralized Work Distribution Improved Company Slotting Mechanism Brought internship opportunities for Post-graduates.
- End-to-End online registration of the companies directly on the portal.

Media Head

Transform and store the events in the digital format

- Led a team of 15 members, specialized in photography, videography and editing.
- Produced and directed a short film based on the events of the annual-fest for publicity.
- Organized two photography events and managed 150+ participants from various colleges.

COURSE WORK

| Speech Signal Processing | Machine Learning for Signal Processing | Advanced Machine Learning |
|-------------------------------------|---|---------------------------|
| Statistical Signal Processing | Discrete Signal Processing | Applied Linear Algebra** |
| Advanced Topics in Machine Learning | Representation and analysis of Random signals | |
| Natural language Processing | Deep Neural Network** | |

^{**=} Online Courses

^{*=}Submitted and under review