# Khandokar Md. Nayem



#### TECHNICAL EXPERTISE

• Speech Enhancement, Processing & Recognition • Machine Learning • Deep Learning • Natural Language Processing (NLP) • Classification & Regression • Deep Neural Network (DNN) • Recurrent Neural Network (RNN) • Long-Short Term Memory (LSTM) • Convolutional Neural Network (CNN) • Machine Learning Models (Bayesian Network, HMM, Clustering, Decision Tree, Ensemble Methods) • Deep Learning Libraries (TensorFlow, Keras) • NLP Libraries (NLTK, Scikit-Learn) • Human Survey Platform (Qualtrics, Amazon MTurk) • Computer Vision and Graphics Libraries (CImg, OpenCV, OpenGL) • Python • Matlab •  $\mathbb{C} \mathbb{C} + + \mathbb{C}$  Java • HTML, CSS & Javascript • SQL •  $\mathbb{R}$  • Shell Script

## **EMPLOYMENT**

### Indiana University, Bloomington, IN, Research Assistant, ASPIRE research lab

Fall 2016 - Present

- Developed an attention-based monaural <u>speech enhancement</u> model that aims to maximize human perceptual rating of the enhanced speech by incorporating embedding vectors from a human Mean-Opinion Score (MOS) prediction model and jointly training the models on real-world noisy speech data. (INTERSPEECH-2021)
- Proposed and implemented a quantized speech prediction model that classifies speech spectra into a corresponding quantized class and applies a language-style model to ensure more realistic speech spectra. Acceptable quantization level is determined by a listener study ran in <a href="Manazon MTurk"><u>Amazon MTurk</u></a> designed in Qualtrics. (ICASSP-2021, poster, slides, video)
- Designed a recurrent layer named Intra-Spectral Recurrent (ISR) layer that captures spectral dependencies within the magnitude and phase responses of the noisy speech using Markovian recurrent connections, and successfully deployed in a <u>LSTM-based</u> single-channel <u>speech enhancement</u> model. (ICASSP-2020, slides, video)
- Formulated a new type of recurrent output layer that enforces spectral-level dependencies within each spectral time frame modeling the Markovian assumption along the frequency axis in both uni-directional and bi-directional ways, and tested in a magnitude <u>speech enhancement</u> model. (MLSP-2019, poster)
- Engineered a deep architecture named Recurrent Stacked Generative Adversarial Network (RSGAN) which generates video clips based on a pre-condition like a sentence description, action classes, or fMRI signals. (IU-VISION-2017, poster)
- Designed <u>prediction models</u> for nulliparous pregnant women to diagnose gestational diseases like diabetics, pre-eclampsia, and hypertension as part of the NSF Proactive Health Informatics (PHI) project. Also, developed a smart system that tracks the daily physical activity of women collected by wearable devices and helps to diagnose gestational complications.
- Investigated speech emotion detection deep architectures like RNN and CNN in the scale of arousal and valence.
- Developed an automatic <u>classification of rhetorical questions</u> with stress detection on our own collected dataset using Recurrent Neural Network (RNN) and Convolutional Recurrent Neural Network (CRNN) models. (code )
- Implemented an <u>end-to-end speech recognition</u> system for the English language using bi-directional recurrent neural network architecture without any frame wise labeling. (code )

## BOSE Corporation, Boston, MA, Machine Learning/Neural Signal Processing Intern

Summer 2020

• Researched <u>enhancing speech</u> in remote microphone applications by self-speech removal to provide better quality sound with low latency to the hearing aids and voice-assistive wearable devices. LSTM-based architecture with speaker dependent d-vector is used for real-time operation.

## United International University (UIU), Dhaka, Lecturer, Department of CSE

August 2016

• Taught courses of Computer Science curriculum, like C++ Programming language, Algorithms, Digital Logic Design and Pattern Recognition courses in classes of more than 90 undergrads.

## REVE Systems, Dhaka, Jr. Software Engineer, Team Media Gateway

January 2015

• Programmed media gateway controller to facilitate both calls and faxes between the telephone network and VoIP network or another telephone network via <u>Megaco 1.0 protocol</u>. Also designed front-end panel by <u>.JSP framework</u> for VoIP administrators and customers for easy use.

#### **EDUCATION**

Ph.D. in Computer Science,

Indiana University, Bloomington, IN

Fall 2022 (Anticipated) Advisor: Prof. Donald S. Willamson

M.Sc. in Computer Science,

. Donaid S. Williamson

Indiana University, Bloomington, IN

December 2019

B.Sc. in Computer Science & Engineering (CSE),

July 2014