

# Khandokar Md. Nayem

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## RESEARCH INTEREST

• Deep Learning • Machine Learning • Large Language Model (LLM) • Natural Language Processing (NLP) • Digital Signal Processing (DSP) • Speech Enhancement, Translation & Recognition • Digital Health • Wearable Computing

## EDUCATION

<b>Ph.D. in Computer Science</b> <a href="#">Indiana University</a> , Bloomington, IN, USA	Fall 2023 (Anticipated) Advisor: <a href="#">Prof. Donald S. Williamson</a>
<b>M.Sc. in Computer Science</b> <a href="#">Indiana University</a> , Bloomington, IN, USA	December 2019
<b>B.Sc. in Computer Science &amp; Engineering (CSE)</b> <a href="#">Bangladesh University of Engineering &amp; Technology (BUET)</a> , Dhaka, Bangladesh	July 2014

## PUBLICATIONS

### Peer-reviewed & Accepted (Journal):

- **Khandokar Md. Nayem**, Donald S. Williamson, “Attention-based Speech Enhancement Using Human Quality Perception Modelling”, in Proceedings of journal IEEE/ACM [Transactions on Audio Speech and Language Processing \(TASLP\)](#) 2023.  
Journal Impact factor: 3.919, SCImago Journal Rank (SJR): 0.92

### Peer-reviewed & Published (Conference):

- **Khandokar Md. Nayem**, Ran Xue, Ching-Yun Chang, Akshaya Vishnu Kudlu Shanbhogue, “Knowledge Distillation on Joint Task End-to-End Speech Translation”, in Proceedings of the Annual Conference of the [International Speech Communication Association \(ISCA\)](#), INTERSPEECH 2023. ([paper](#), [poster](#))  
Conference Impact factor: 5.14, SCImago Journal Rank (SJR): 0.689
- **Khandokar Md. Nayem**, Donald S. Williamson, “Incorporating Embedding Vectors from a Human Mean-Opinion Score Prediction Model for Monaural Speech Enhancement”, in Proceedings of the Annual Conference of the [International Speech Communication Association \(ISCA\)](#), INTERSPEECH 2021. ([paper](#), [slides](#), [video](#))  
Conference Impact factor: 5.14, SCImago Journal Rank (SJR): 0.689  
Paper Citation in Google Scholar: 4
- **Khandokar Md. Nayem**, Donald S. Williamson, “Towards an ASR approach using Acoustic and Language Models for Speech Enhancement”, in Proceedings of IEEE [International Conference on Acoustics, Speech and Signal Processing \(ICASSP\)](#) 2021. ([paper](#), [poster](#), [slides](#), [video](#))  
Conference Impact factor: 4.65, SCImago Journal Rank (SJR): 0.546  
Paper Citation in Google Scholar: 2
- **Khandokar Md. Nayem**, Donald S. Williamson, “Monaural speech enhancement using intra-spectral recurrent layers in the magnitude and phase responses”, in Proceedings of IEEE [International Conference on Acoustics, Speech and Signal Processing \(ICASSP\)](#) 2020. ([paper](#), [slides](#), [video](#))  
Conference Impact factor: 4.65, SCImago Journal Rank (SJR): 0.546  
Paper Citation in Google Scholar: 1
- **Khandokar Md. Nayem**, Donald S. Williamson, “Incorporating Intra-Spectral Dependencies with a Recurrent Output Layer for Improved Speech Enhancement”, in Proceedings of IEEE [International Workshop on Machine Learning for Signal Processing \(MLSP\)](#) 2019. ([paper](#), [poster](#))  
Conference Impact factor: 1.81, SCImago Journal Rank (SJR): 0.283  
Paper Citation in Google Scholar: 6

## Poster:

- Shujon Naha, **Khandokar Md. Nayem**, Md. Lisul Islam, “RSGAN: Recurrent Stacked Generative Adversarial Network for Conditional Video Generation”, presented at IU computer vision project showcase, 2017. ([paper](#), [poster](#))
- **Khandokar Md. Nayem**, Mir Toornaw Islam, Md. Monirul Islam, “Handwritten Writer Independent Bangla Character Recognition”, presented at undergrad thesis at BUET, 2014. ([paper](#), [poster](#))

## Archive:

- **Khandokar Md. Nayem**, Donald S. Williamson, “Attention-based Speech Enhancement Using Human Quality Perception Modelling”, in arxiv, 2023. ([paper](#))

## RESEARCH EXPERIENCE

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### Indiana University, Bloomington, IN, USA

Summer 2018 - Now

*Research Assistant, [ASPIRE research lab](#)*

- Developed a novel [speech enhancement](#) model that aims to maximize human perceptual satisfaction of speech after cleaning real-world noise from the captured audio. We incorporate perceptually important features learned from a separate perception rating prediction model in achieving this task.
- Proposed & implemented a quantized speech prediction model that classifies speech spectra into a corresponding quantized class and applies a language-style model to generate more realistic speech. Acceptable quantization level was determined by listener subjective study conducted on [Amazon MTurk](#), designed using [Qualtrics](#).
- Formulated a new type of recurrent output layer in the context of deep learning architecture that shows significant improvement in [speech enhancement](#). This new design utilizes the internal relations along the frequency axis between speech frequency points.
- Engineered a deep architecture named Recurrent Stacked Generative Adversarial Network ([RSGAN](#)) which [generates video](#) clips given different types of signal like human's brain fMRI signal or sentence description of the video clips.
- Developing a [smart system](#) that tracks the daily physical activity and sleep activity of pregnant women collected by wearable devices and helps to diagnose gestational complications, like gestational diabetics and pre-eclampsia. This project is part of the ongoing [NSF Proactive Health Informatics \(PHI\)](#) project.
- Researched on [speech emotion](#) detection systems that analyze speech to monitor human emotions, a valuable cue for tackling sensitive emotional situations and maintaining a healthy conversation.

### Amazon Services LLC, Seattle, WA, USA

Summer 2023

*Applied Scientist Intern, Consumer SPIRIT*

- Conducting research on the application of the [Large Language Model](#) (LLM) for class labeling on closed taxonomy utilizing product descriptions, while also generating [chain-of-reasoning](#) explanations for improved overall comprehension.

### Amazon Services LLC, Cambridge, MA, USA

Fall 2022

*Applied Scientist Intern, Alexa AI*

- Researched the development of a real-time, end-to-end compressed multi-lingual [speech translation](#) system. Investigated the use of [Large Language Models](#) (LLMs) and applied [knowledge distillation](#) approach to transfer their performance to smaller models with 50% and 75% fewer parameters.

### Microsoft Corporation, Redmond, WA, USA

Summer 2022

*Research Intern, Audio and Acoustics Research Group*

- Focused on analyzing and improving the performance of [speech enhancement](#) algorithms to generate high-fidelity (Hi-Fi) speech by removing distortions and extending speech bandwidth. Applied real-time [deep learning](#) models with various data augmentation techniques to recover codec and clipping distortions, and performed deep noise suppression.

### BOSE Corporation, Boston, MA, USA

Summer 2020

*Machine Learning/Neural Signal Processing Intern*

- Researched on [enhancing speech](#) in remote microphone applications by removing self-speech in order to provide better quality sound with low latency to hearing aids and voice-assistive wearable devices. Utilized a [deep learning](#) architecture with speaker-dependent features for [speaker identification](#), to ensure real-time operation.

- Programmed controller of ‘media-gateway’ server to facilitate both phone calls and faxes between the telephone network and VoIP network or another telephone network. Also designed a front-end panel of controllers for VoIP administrators and customers.

## TEACHING EXPERIENCE

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- Mentored 2 undergrad students [Muhammad Asghar](#) (currently works as Software Engineer at [L3Harris](#)) and [Daniel Quintans](#) (currently works as Software Development Engineer Intern at [Amazon](#)) as a part of the **NSF Research Experience for Undergraduates (REU)** program.
- Mentored **summer visitor undergrad** student [Chitrang Gupta](#) (currently works at the [University of Texas, Austin](#)) in audio focused machine learning research project.
- Taught over 50 undergrads in Data Science course - Data Representation (D-321) and coordinating AIs.
- Taught over 25 undergrads on Computer Science course - Machine Learning (B-455).
- Taught over 60 undergrads on Computer Science course - Python (A-201).
- Taught over 60 undergrads on Computer Science course - Discrete Mathematics (C-241).

- Evaluated and judged undergrad final research thesis projects.
- Coached undergrad students for **programming contest** at south-asia region of [International Collegiate Programming Contest \(ICPC\)](#), 2016.
- Taught over 2000 undergrads on Computer Science programming courses - Structured Programming Language (CSE-1111), Structured Programming Language Laboratory (CSE-1112), Object Oriented Programming (CSE-1115), and Object Oriented Programming Laboratory (CSE-1116).
- Taught over 800 undergrads on Computer Science hardware courses - Digital Logic Design (CSE-1325) and Digital Logic Design Laboratory (CSE-1326).
- Taught over 60 undergrads on Computer Science algorithm course - Data Structure and Algorithms I (CSE-2215).
- Taught over 250 undergrads on Computer Science artificial intelligence (AI) courses - Artificial Intelligence (CSE-3811) and Artificial Intelligence Laboratory (CSE-3812).

## ARTICLE, TALKS & PRESENTATION

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- Article published in **Amazon Science** on “Knowledge Distillation on Joint Task End-to-End Speech Translation”, 2022. ([link](#))
- Title “Knowledge Distillation on Joint Task End-to-End Speech Translation”, Presenter in **Amazon Services LLC**, 2022.
- Title “Unified Speech Enhancement Approach for Speech Degradations and Noise Suppression”, Presenter in **Microsoft Corporation**, 2022.
- Title “Incorporating Embedding Vectors from a Human Mean-Opinion Score Prediction Model for Monaural Speech Enhancement”, Presenter in conference **INTERSPEECH 2021**. ([video](#))
- Title “Towards an ASR approach using Acoustic and Language Models for Speech Enhancement”, Presenter **ICASSP 2021**. ([video](#))
- Title “Speech Enhancement”, Presenter in **BOSE Corporation**.
- Title “Monaural speech enhancement using intra-spectral recurrent layers in the magnitude and phase responses”, Presenter in conference **ICASSP 2020**. ([video](#))

- Title “Speech Emotion Recognition”, Guest Lecturer in course **CSCI-B 659**, 2020.
- Title “Machine Learning & Deep Learning”, Guest Lecturer in **NSF Research Experience for Undergraduates (REU) program**, 2019.
- Title “Improved Speech Enhancement by Incorporating Intra-Spectral Dependencies”, Presenter in workshop **Midwest Music and Audio Day (MMAD)**, 2019.
- Title “Incorporating Intra-Spectral Dependencies with a Recurrent Output Layer for Improved Speech Enhancement”, Presenter in conference **MLSP 2019**.

## AWARDS & SERVICES

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- **Reviewer** for the **International Speech Communication Association (ISCA)** at **INTERSPEECH 2023** in Dublin, Ireland.
- **Student Volunteer** responsible for managing the virtual platform at **ISCA INTERSPEECH 2021** in Brno, Czech Republic.
- Recipient of a **Travel grant** from Indiana University to attend **IEEE ICASSP 2020** in Barcelona, Spain.
- **Panelist** at the career development session titled “Career Choice: Academia and Industry” organized by **NSF Research Experience for Undergraduates (REU) program** in 2019.
- Recipient of a **Travel grant** to attend at the Annual Conference of **ISCA INTERSPEECH 2019** in Graz, Austria.
- **Student Volunteer** involved in organizing conference at the Annual Conference of **ISCA INTERSPEECH 2019** in Graz, Austria.
- **President** of the **Bangladesh Student Association (BDSA)** at Indiana University, Bloomington, IN, overseeing a 50+ member student organization in 2020.
- Served as **Vice-President, General Secretary, and Treasurer** of **BDSA** at Indiana University, Bloomington, IN, consecutively from 2017 to 2019.
- Attained the rank of **Brown belt** in Taekwondo and served as a judge for **Yellow belt** students at the **Indiana University Taekwondo club** from 2017 to 2020.