# **Anthony Hong**

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### **Education**

Carnegie Mellon University (CMU)

Pittsburgh, PA, USA

Bachelor of Engineering and Arts, Electrical and Computer Engineering and Music Technology Additional Major: Artificial Intelligence

#### **Relevant Coursework**

Advanced Digital Signal Processing (Graduate) Introduction to Deep Learning Introduction to Computer Systems Machine Learning for Signal Processing (Graduate)

Introduction to Machine Learning

Introduction to Embedded Systems (Spring 2024)

### **Skills**

Technical: real-time DSP, audio prototyping, acoustics, data structures and algorithms, systems programming Musical: FL Studio, Logic Pro, Pro Tools, electronic music production, piano, sound engineering Tools: Matlab, Juce, C/C++, CMake, Python, PyTorch, Git, Google Cloud, HTML/CSS, Java, Max/MSP/Jitter Languages: English (native), Chinese (native), Shanghainese (native)

## **Experience / Projects**

## **Virtual Eurhythmics Tutor**

Sept. 2023 - Present

- Developed a program that plays randomly generated rhythms for users to practice musicianship
- Designed logic to support different BPM's and accurate timings
- Designed algorithms to generate sequences of fixed total duration and record it as text score
- Designed algorithms to mark bars and split and tie notes across downbeats for visual clarity

**Gerzz Interactive** 

May. 2023 - Sept. 2023

Machine Learning and Signal Processing Intern

Shanghai, China

- Modified the existing VITS model for training singing voice conversion, improved clarity of pronunciation and naturalness of unvoiced components
- Implemented real-time reverb and Transformers robot voice effect algorithms in C++
- Developed a real-time AI-based acoustic echo cancellation system

#### Filter-dependent Point Cloud Audio Visualization

Feb. 2023 - Mar. 2023

- Developed an audio filter whose shape also modifies granularity of point cloud visualization
- Programmed the filter to be shaped arbitrarily by mouse
- Written entirely in Max/MSP/Jitter

## **Carnegie Mellon University**

Aug. 2022 - May. 2023

Pittsburgh, PA, USA

Teaching Assistant

- "Signals and Systems": covered system properties, convolution, Fourier/Laplace/Z Transform, sampling
- "Digital Signal Processing": covered multirate DSP, filter design, Discrete/Fast Fourier Transform
- Hosted weekly office hours to help students understand important signal processing concepts
- Created homework assignments and exams that aid students' understanding

## **Course Projects**

Jun. 2022 - May. 2023

- "Introduction to Deep Learning" Course Projects
  - Created Pytorch classes from scratch for loading and preprocessing speech and image data
  - o Implemented various models for different tasks from scratch, using basic Pytorch functions
  - Reviewed multiple research papers describing different model architectures and their performances
  - Attention-based Automatic Speech Recognition
    - Reviewed the "Listen, Attend, and Spell" research paper
    - Implemented an attention mechanism, a pyramidal bi-directional long-short-term memory (pBLSTM)-based encoder (listener), and a LSTM-based decoder (speller)

- Used the Librispeech dataset for training, validating, and testing
- Achieved 8.9 Levenshtein distance during testing
- Convolutional neural network for face classification
  - Reviewed research papers on Convnext and ResNet
  - Experimented with and built Convnext and ResNet from scratch, using pytorch functions
  - Used the VGGFace2 dataset for training, validating and testing
  - Achieved 90% test accuracy using the Convnext model
- Deep neural network for frame-level speech recognition
  - Built a multi-layer perceptron (MLP) for mapping speech to phonemes
  - Experimented with various MLP architectures and other hyperparameters, including but not limited to: dropout probability, learning rate, scheduler, batch size, and optimizer
  - Used the Librispeech dataset for training, validating, and testing
  - Achieved 88% test accuracy
- "Advanced Digital Signal Processing" Matlab Simulations
  - Noise canceller using adaptive filtering and the LMS update algorithm
  - o Linear predictive coding-based vocoder
  - Phase vocoder that changes the time and pitch of music independently
- "Introduction to Computer Systems" Course Projects
  - o Tiny interactive shell that supports background jobs, process interruptions, and zombie reaping
  - Dynamic memory allocator (1300+ lines of C code) achieving high throughput and utilization
  - Cache simulator adopting the LRU replacement policy
  - Fast matrix transposition implemented with blocking techniques

## Carnegie Mellon University AB Tech

Sound Engineer

Aug. 2021 - Dec. 2022 Pittsburgh, PA, USA

- Set up sound and lighting equipments
- Assisted operating sound board during the performance to monitor sound quality and feedback
- Communicated with stage managers to ensure smooth transition between performances

### **Music Producer**

Dec. 2015 - Present

- Produced high-quality original music, solely responsible for all aspects of production: composition, sound design, sound recording, mixing and mastering
- Used signal processing methods to generate real-time audiovisual artwork accompanying original music
- Created a musician website from scratch using HTML/CSS, regularly updating content

#### References

Richard Stern, Professor of Electrical and Computer Engineering at CMU: rs1e@andrew.cmu.edu

Ying Xu, CTO at Gerzz Interactive: <u>ishine2010@gmail.com</u> Tiange Ling, CEO at Gerzz Interactive: <u>tiangeling@gmail.com</u>

#### Awards / Honors

Dean's list, Fall 2022

International Electronic Music Competition Top 25%, Aug. 2019
Black Hole Recordings China Remix Competition Honorable Mention, Feb. 2019

United States Academic Decathlon China Music Section Gold Medalist, Feb. 2019