

<b>EDUCATION</b>	<p><b>Columbia University</b>  <i>M.S. Computer Science (Machine Learning Track)</i></p> <ul style="list-style-type: none"> <li>• Coursework included: artificial intelligence, machine learning, advanced topics in neural networks and deep learning, advanced topics in spoken language processing and speech recognition, natural language processing, computer vision, databases, and analysis of algorithms.</li> <li>• Conducted research advised by Professor Peter Belhumeur on developing self-supervised optical character recognition model whose training procedure is guided by separate language model via student-teacher knowledge distillation paradigm without use of ground-truth labels.</li> </ul> <p><i>B.A. Drama and Theatre Arts</i></p>	<p>New York, NY  Dec. 2022</p> <p>May 2018</p>
<b>EXPERIENCE</b>	<p><b>TikTok/ByteDance</b>  <i>Research Scientist</i>  <i>Speech Synthesis Engineer</i>  <i>Software Engineer Intern</i></p> <ul style="list-style-type: none"> <li>• Conduct research on <a href="#">Seed/Doubao</a> team (previously Speech, Audio, and Music Intelligence) on state-of-the-art neural speech synthesis and audio processing algorithms for various TikTok and ByteDance products to empower content creation and consumption.</li> </ul>	<p>San Jose, CA  Jul. 2024 – Present  Dec. 2022 – Jun. 2024  May 2022 – Nov. 2022</p>
<b>RESEARCH</b>	<p>My research interests include deep generative modeling, self-supervised representation and transfer learning, zero-shot learning, and knowledge distillation. I’m especially interested in neural end-to-end learning for audio and natural language processing tasks.</p>	
<b>PUBLICATIONS</b>	<p><b>Seed Team</b>, ByteDance, “<i>Seed-TTS: A family of high-quality versatile speech generation models</i>,” arXiv:2406.02430, Jun. 2024. [<a href="#">Paper</a>, <a href="#">Demo</a>, <a href="#">Code</a> (1K+ stars)]</p> <p><b>Philip Anastassiou*</b>, Zhenyu Tang*, Kainan Peng, Dongya Jia, Jiaxin Li, Ming Tu, Yuping Wang, Yuxuan Wang, Mingbo Ma (*equal contribution), “<i>VoiceShop: A unified speech-to-speech framework for zero-shot voice editing</i>,” arXiv:2404.06674, Apr. 2024. [<a href="#">Paper</a>, <a href="#">Demo</a>]</p>	
<b>PATENTS</b>	<p><b>Philip Anastassiou</b>, Zhenyu Tang, Jiaxin Li, Kainan Peng, Dongya Jia, Qiao Tian, Mingbo Ma, Yuping Wang, Yuxuan Wang, “<i>Identity-preserving zero-shot many-to-many accent and speech style conversion via bottleneck-to-bottleneck and diffusion-based modeling</i>,” CN202311659609.5, pending CN patent application filed by ByteDance Ltd., 2024.</p>	
<b>PROF. SERVICES</b>	<p><b>Program Committee:</b> <a href="#">AAAI Conference on Artificial Intelligence</a> (2025); <b>Paper Reviewer:</b> <a href="#">IEEE Signal Processing Letters</a> (2024).</p>	
<b>PRESENTATIONS</b>	<p><b>Nokia Bell Labs</b>  <i>Experiments in Art and Technology Program (“Squashed” Software)</i></p> <ul style="list-style-type: none"> <li>• Developed software for Nokia Bell Labs E.A.T. program at Columbia Computer Music Center to produce musically desirable digital artifacts in audio signals with lossy LAME MP3 encoder.</li> </ul> <p><b>Lamont-Doherty Earth Observatory</b>  <i>Research as Art Exhibition (“Novel Song” Software)</i></p> <ul style="list-style-type: none"> <li>• Developed software to convert text of fictional novels into music with VADER Python package for sentiment analysis to interpolate emotional valence scores into harmonic relationships in RTemix.</li> </ul>	<p>Murray Hill, NJ  Feb. 2019</p> <p>Palisades, NY  May 2018</p>
<b>PROJECTS</b>	<p><b>VAE-GAN for Speech-to-Speech Style Transfer</b></p> <ul style="list-style-type: none"> <li>• Implemented proposed variational autoencoder-generative adversarial network (VAE-GAN) architecture with domain-specific decoders for non-autoregressive speech-to-speech style transfer based on <a href="#">AlBadawy, et al. (2020)</a> and <a href="#">Bonnici, et al. (2021)</a>. [<a href="#">Code</a>]</li> </ul> <p><b>Deep Convolutional Spectral Autoencoder for Neural Audio Synthesis</b></p> <ul style="list-style-type: none"> <li>• Implemented deep convolutional autoencoder in Python using TensorFlow for audio synthesis of musical notes based on <a href="#">Engel, et al. (2017)</a>, trained on subset of Google Magenta’s <i>NSynth</i> corpus.</li> </ul>	<p>Dec. 2021</p> <p>May 2021</p>
<b>SKILLS</b>	<p><b>Languages:</b> Python, Java, C, MATLAB, SQL, Unix shell scripting.  <b>Software:</b> Git, PyTorch, PyTorch Lightning, TensorFlow, Keras, Pandas, SciPy, NumPy, Matplotlib, Scikit-Learn, FFmpeg, Librosa, Kaldi, ESPNet, Praat, NLTK, Conda, <math>\LaTeX</math>.</p>	