

INTRODUCTION

**Research Interests:** Generative AI, LLM-powered agents, human-AI collaboration, and multimodal modeling (speech/text). I am a Master’s student at the University of Southern California (USC) focused on building reliable, deployable agentic systems, with experience spanning speech and multimodal modeling, data pipelines, evaluation, and production ML systems.

EDUCATION

<b>M.S. in Artificial Intelligence</b> <i>University of Southern California (USC)</i>	<b>Sep. 2025 – Present</b> <i>Los Angeles, CA, USA</i>
<b>M.S. in Software Engineering</b> <i>Beijing University of Technology BJUT; GPA: 3.4</i>	<b>Sep. 2016 – May. 2019</b> <i>Beijing, China</i>
<b>B.E. in Computer Science</b> <i>Southwest Minzu University SMU; GPA: 3.66, “Top 5” of 154 students</i>	<b>Sep. 2012 – Jul. 2016</b> <i>Chengdu, China</i>

PROFESSIONAL EXPERIENCE

<b>Part-time Consultant</b> <i>[Mobvoi]</i> <ul style="list-style-type: none"><li><b>Consultant:</b> Optimized production speech models and LLM-based multi-agent workflows; improved offline evaluation, data pipelines, and deployment reliability.</li></ul>	<b>Aug. 2025 – Present</b> <i>Los Angeles, CA, USA</i>
<b>Speech Tech Lead</b> <i>[Mobvoi]</i> <ul style="list-style-type: none"><li><b>Agents:</b> Built and deployed a production-grade LLM-powered agent framework for speech-centric applications (tool calling, multi-agent coordination, safety guardrails, runtime monitoring) <a href="#">[www.moyin.com]</a>.</li><li><b>LLMs &amp; Audio:</b> Integrated LLMs with speech/audio pipelines (ASR, TTS, prompt generation) to ship AI-driven product features in production <a href="#">[www.dupdub.com]</a>.</li><li><b>Data Engineering:</b> Designed and maintained scalable multimodal data pipelines (text, audio, image) for model training and evaluation, including data collection, versioning, and automated quality checks at large scale.</li><li><b>Evaluation:</b> Developed an end-to-end evaluation framework for multimodal generation, combining offline benchmarks and regression testing to ensure model quality and production stability.</li><li><b>Collaboration:</b> Partnered with product managers, researchers, and engineers to turn AI prototypes into production systems; mentored junior engineers and interns.</li></ul>	<b>May. 2023 – Jul. 2025</b> <i>Suzhou, China</i>
<b>Speech Algorithm Engineer</b> <i>[Mobvoi]</i> <ul style="list-style-type: none"><li><b>NLP/Speech Synthesis:</b> Owned and optimized core speech algorithms (multilingual TTS, text normalization, G2P, prosody prediction, emotion analysis, rhythm prediction) for both foundational NLP components and product-facing speech features.</li><li><b>Backend:</b> Designed and built production speech backend services, exposing scalable TTS/NLP APIs that powered a consumer AI product with tens of millions of users.</li><li><b>Services:</b> Improved reliability of large-scale speech services to 99% availability through deployment hardening, monitoring, performance optimization, and production incident response.</li><li><b>Collaboration:</b> Mentored 3 interns on Speech/NLP projects (annual program).</li></ul>	<b>Jul. 2019 – May. 2023</b> <i>Beijing, China</i>
<b>Algorithm Research Intern</b> <i>TAL AI Lab</i> <ul style="list-style-type: none"><li><b>Graph-based Learning:</b> Developed a Deep Knowledge Tracing (DKT) pipeline using graph-based embeddings and distance metrics to model and evaluate student knowledge progression.</li><li><b>Research Outcome:</b> Co-authored a paper accepted at <a href="#">AIED 2019</a>; received the Outstanding Intern award for research contributions.</li></ul>	<b>Aug. 2018 – Dec. 2018</b> <i>Beijing, China</i>

# PROJECTS

- AI-Powered Audio Visualization Platform [QinZhi Creative]**

Nov. 2025 – Present

  - Developed and launched a web-based audio processing platform that utilizes AI algorithms to generate high-fidelity, dynamic visualizations from speech and subtitles.
  - Architected a scalable cloud infrastructure on Google Cloud Platform (GCP); implemented serverless backend services to handle audio rendering and media processing pipelines with high availability.
  - Engineered an end-to-end dubbing workflow, integrating speech API with an interactive frontend to provide professional-grade audio editing and visualization tools for creators.

# SKILLS

- Languages:** Python, C++, Bash, SQL
- Backend / Systems:** backend services, APIs, distributed systems, AI model serving, monitoring, CI/CD
- Data Engineering:** large-scale datasets, dataset construction, data quality and versioning.
- AI / ML:** LLM integration, agent systems, speech & NLP, multimodal learning, benchmark evaluation.
- Dev/ML Tools:** Git, Docker, Linux, Kubernetes, cloud infrastructure; PyTorch, TensorFlow, vLLM

# TEACHING EXPERIENCE

- Lecture - Embedded System Design Practice**

Beijing University of Technology

Supported instruction and labs for 80 M.S. students annually.

2018

Beijing, China
- Company - Speech & NLP**

Mobvoi

Mentored 3 interns on speech/NLP projects (annual program).

2021-2025

Beijing, China

# PUBLICATIONS

- Proceedings**
- Xinsheng W, **Xiaoqin F**, et al. *Spark-TTS: An Efficient LLM-Based Text-to-Speech Model with Single-Stream Decoupled Speech Tokens*. arXiv preprint, 2025. [PDF]
  - Dongsheng W, **Xiaoqin F**, Zeming L, et al. *2M-NER: Contrastive Learning for Multilingual and Multimodal NER with Language and Modality Fusion*. Applied Intelligence, 2024. [PDF]
  - Wenjiang C, **Xiaoqin F**(† equal contribution), Yunlin C, et al. *Multi-granularity Semantic and Acoustic Stress Prediction for Expressive TTS*. APSIPA ASC, 2023. [PDF]
  - Jipeng Z, **Xiaoqin F**. *Prosody Prediction with Discriminative Representation Method*. PRML, 2022. [PDF]
  - Xiaoqin F**, Xie R, et al. *Population Statistics Algorithm Based on MobileNet*. Journal of Physics: Conference Series (JPCS), 2019. [PDF]
  - Zhiwei W, **Xiaoqin F**, et al. *Deep Knowledge Tracing with Side Information*. AIED, 2019. [PDF]
  - Xie Rong, **Xiaoqin F**. *A Method of Quick Edge Detection Based on Zynq*. CCIOT, 2018. [PDF]
  - Sheng J, **Xiaoqin F**, et al. *Research on the Internet of Things Platform for Smart and Environmental Protection*. CCIS, 2018. [PDF]

# Patents

- All relevant to speech/NLP/LLM areas.
- 6 First inventor:** CN111078898A CN110970013A CN111145724A CN115470351A CN115470350A CN115547289A
- 4 Co-inventor:** CN111079428A CN111178042A CN115578998A CN116013251A

# Personal Website

- Website:** [xqfeng-josie.github.io](https://xqfeng-josie.github.io)
- Contains selected projects (agents, speech/LLMs), publications, demos, and updates.