Cascaded to End-to-End: New Safety, Security, and Evaluation Questions for Audio Language Models

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1 Introduction

A growing number of large language models (LLMs) now process audio input alongside text, commonly known as Audio LMs. Although this development is not entirely new—voice assistants have existed for some time—audio was historically processed by language models through a **cascaded** pipeline approach until recently. In this pipeline, audio is first transcribed into text by a separate automatic speech recognition (ASR) model, and then the text is fed into an LLM. Some voice assistants, like Amazon's Alexa, Apple's Siri, ChatGPT's early voice mode [\[1\]](#page-2-0), and others, use such a cascaded pipeline. The cascaded pipeline is limited, though. The transcription step discards rich information in the audio input, such as the speaker's intonation, pronunciation, the presence of multiple speakers, background scene information, and more. Since the LLM does not have access to this information in the cascaded system, it cannot incorporate it into any downstream decisionmaking and processing. This consideration has driven a shift toward **end-to-end** Audio LMs, such as GPT-4o [\[2\]](#page-2-1). These models process audio inputs by directly accessing rich audio features rather than relying on text tokens transcribed by an intermediary ASR model. In this perspective paper, we first underscore novel safety and security risks that could be introduced by the transition from cascaded to end-to-end Audio LMs. We then highlight tensions and gaps in current end-to-end Audio LM evaluation protocols between open and closed-source models. We hope our work spurs a re-alignment in open-source Audio LM safety, security, and capability evaluations.

2 Expanded Safety and Security Risks in End-to-End Audio LMs

End-to-end Audio LMs can directly access rich audio features, making them more amenable to solving a broader range of tasks—a key goal for general-purpose systems. However, direct access to audio leaks sensitive information to the model and expands the model's risk profile.

Sociotechnical Safety Challenges. Audio recordings of human speech contain rich information, much of which might be considered sensitive. Whether correctly or not, a large swath of researchers have tried to infer sensitive attributes from audio data [\[3\]](#page-2-2), including: identity [\[4\]](#page-2-3); demographic attributes like age [\[5\]](#page-2-4), gender [\[6\]](#page-2-5), race and ethnicity [\[7\]](#page-2-6), and socioeconomic status [\[8\]](#page-2-7); and psychophysiological traits like emotions [\[9\]](#page-2-8), personality [\[10\]](#page-2-9), intoxication [\[11\]](#page-2-10), and mental health [\[12\]](#page-2-11).

There is a risk, then, that end-to-end Audio LMs may implicitly use these features in their processing of user requests or make unintended inferences about the speaker, leading to allocative and representational harms [\[2,](#page-2-1) [13\]](#page-2-12). Consider, for example, if a model is asked to find information about potential jobs, but infers the speaker's demographics and adjusts its recommendations towards stereotypical biases. Or a model infers that a user may be vulnerable in some way and manipulates them. While some have begun examining these risks in closed source settings [\[14\]](#page-2-13), or have examined biases arising from ASR errors across disparate groups [\[15\]](#page-2-14), there is still a dearth of open-source evaluations examining these harms in end-to-end Audio LMs.

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Audio LMs could also be explicitly abused for **harmful inference** since they leak potentially sensitive information. If Audio LMs retain the few-shot prompting and adaptation capabilities of text-based LMs, they may enable a wide range of surveillance or privacy-violating uses with relative ease. This capability, a byproduct of shifting toward end-to-end Audio LMs, may come into tension with several laws globally. The EU AI Act explicitly prohibits emotion recognition in educational and professional settings and still classifies it as high risk in scenarios where it is not outright prohibited [\[16\]](#page-2-15). The collection and the use of biometric information—which could include voice data used to make these inferences, or even potentially identifying features stored within Audio LMs—might be subject to the Biometric Information Privacy Act (BIPA) of Illinois [\[17\]](#page-3-0) and the EU General Data Protection Regulation (GDPR) [\[18\]](#page-3-1). Without appropriate safeguards and evaluations, developers and users of Audio LMs may find themselves facing legal repercussions.

Security Risks of Adversarial Attacks. End-to-end Audio-LMs are also more vulnerable to audiodomain adversarial attacks than classical cascaded architectures. For the latter, audio adversarial attacks can, at most, trick the ASR module into generating incorrect transcriptions [\[19\]](#page-3-2), and the impact of attacks is no greater than that of purely textual adversarial attacks. However, end-to-end Audio-LMs, like vision language models, make possible strong gradient-based adversarial attacks [\[20\]](#page-3-3). Consequently, downstream behaviors of the Audio LM can be directly manipulated by adversarially crafted audio inputs. This vulnerability can lead to targeted attacks on the model's output [\[21\]](#page-3-4) or jailbreaking behaviors [\[20\]](#page-3-3).

3 Gaps in Current Audio-LMs Evaluation

The expanded risks introduced by end-to-end Audio-LMs raise new evaluation questions.

How should benchmarks align on safety and capability evaluations? There is an ongoing tension in Audio LM safety versus capability evaluation. Some major Audio LM evaluation benchmarks, such as AIR-Bench [\[22\]](#page-3-5) and AudioBench [\[23\]](#page-3-6), reward improved ability to identify sensitive features from audio, including gender, age, and emotion. Model developers may then explicitly optimize for performance on these tasks. Qwen-2-Audio [\[24\]](#page-3-7), for example, explicitly includes emotion recognition in its training process. In contrast, some closed-source tech companies with proprietary models have adopted more cautious measures to mitigate legal risks. OpenAI, for example, states that GPT-4o is designed to avoid inferring a speaker's race or socioeconomic status and to refrain from prompting emotional reliance [\[14\]](#page-2-13). This discrepancy creates an unsettling outcome: proprietary evaluations test that models *do not engage* in behaviors with some undesirable legal and ethical implications, while open source evaluations test that models *do well* in those tasks.

What legitimate use cases genuinely benefit from the end-to-end approach and justify its additional risks? General-purpose audio capabilities may be desirable. But Audio LM benchmark creators may want to revisit which safe capabilities actually outweigh the increased safety and security risks from an end-to-end system. For conversation-oriented tasks, if the primary intended functions of an Audio LM are already satisfied by a cascaded model, why should we even deploy an end-to-end alternative? If the goal is merely to build an AI assistant capable of responding to voice commands, a cascaded model with a good ASR module may already suffice. In such scenarios, providing the AI assistant with additional audio features that could reveal the speaker's identity and emotional state does not directly contribute to a better response to commands while increasing another risk.

4 Conclusion

Including rich audio features is motivated by capturing paralinguistic information, but is that fundamentally at odds with the risks of introducing sensitive information? This is a fundamental tension that current evaluation frameworks insufficiently address. Notably, some developers report performance on tasks like accent, gender, and emotion recognition [\[25,](#page-3-8) [23\]](#page-3-6). These incentivized benchmark tasks may result in models that bear increased risks in their use of sensitive information. Currently, closedsource models like GPT-4o have spent more effort in benchmarking their Audio LM's potential risks, covering categories like speaker identification, ungrounded inference and sensitive trait attribution, and generating problematic speech content (relevant for audio output models) [\[14\]](#page-2-13). However, none of these (or equivalents) are openly available for other model creators. In all, as we shift from cascading to end-to-end Audio LMs, it's time to make sure that capability and safety evaluations are aligned.

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