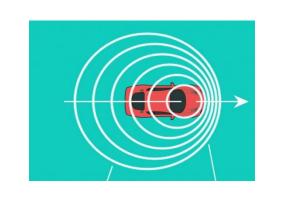


# Teaching Physical Awareness to LLMs through Sounds

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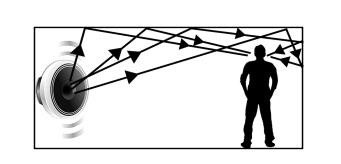


# 1. How Human Ears Understand Physical World?



**Doppler Effect** Identifies whether a

car is approaching



**Multipath Effect** Distinguishes indoor from



**Binaural Hearing** 

Enables localization of sound sources

Sounds inherently carries rich physical information

outdoor environments

### 2. Can Audio LLM Hears like Human Ears?

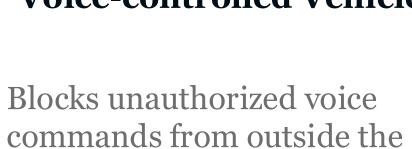


While Audio LLMs perform well on speech content, they lack physical understanding

# 3. Why We Need Physical Understanding?



**Voice-controlled Vehicle** 



vehicle



**Embodied AI Systems** 

Uses sound localization to make systems more human-like



Siren Detection and Localization

Prevents "deaf driver" behavior, enhancing safety and awareness

#### 4. Model Architecture Audio + Text Follow the voice command from the input audio **Prompt** Text 1.Audio Encoder: Converts Encoder raw audio into tokens 2.Text Encoder: Converts text Concatenated **Text Tokens Audio Tokens** input into tokens **Tokens** . LLM: Generates responses based on combined input Large Language Model **Text Response** awareness> No LOS detected, sound source is blocked. Rejecting command as it's outside the car. </physical-awareness: Command rejected. Please give commands from inside the vehicle. Following common practices, we adopt a common end-to-end architecture

## 5. Challenge I: Dataset Construction

### How to collect and annotate a large-scale dataset?

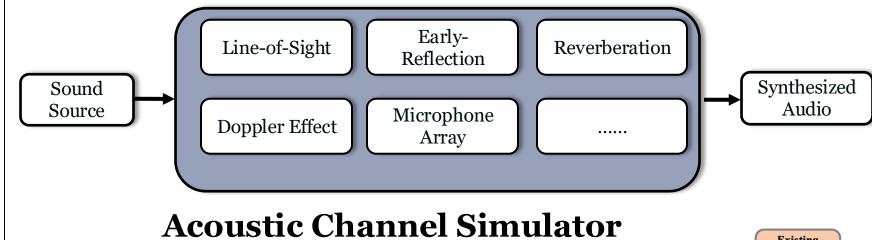
- Data Collection? This requires extensive deployment of recording devices across various environments and conditions, which is expensive and not scalable
- Data Annotation? Unlike text or images where humans can directly annotate content, audio physical cues cannot be labeled easily by humans.

**Key Insight**: The sound that we hear or microphones capture can be decomposed into two independent components:

$$y = h * x$$

- x the sound source
- h the physical channel through which it travels

Solution: Synthesize audios by convolving real sounds with simulated channels



We generate SFT data by convolving existing

Using known simulation parameters, we

Each datapoint is a < Audio, Question,

**Answer>** triplet, enabling grounded physical

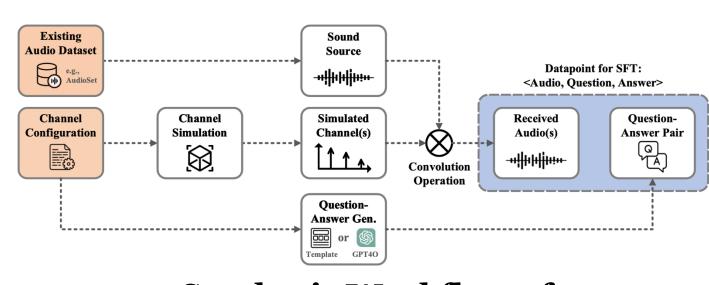
automatically create aligned QA pairs.

sounds with simulated channels.

understanding.

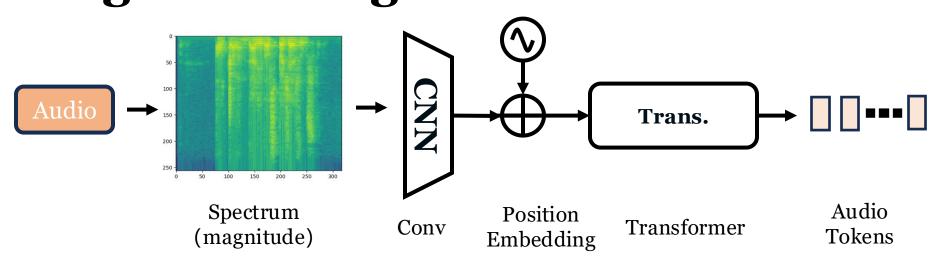
simulator, which models the full life cycle of sound — from emission to reception.

We develop a acoustic channel



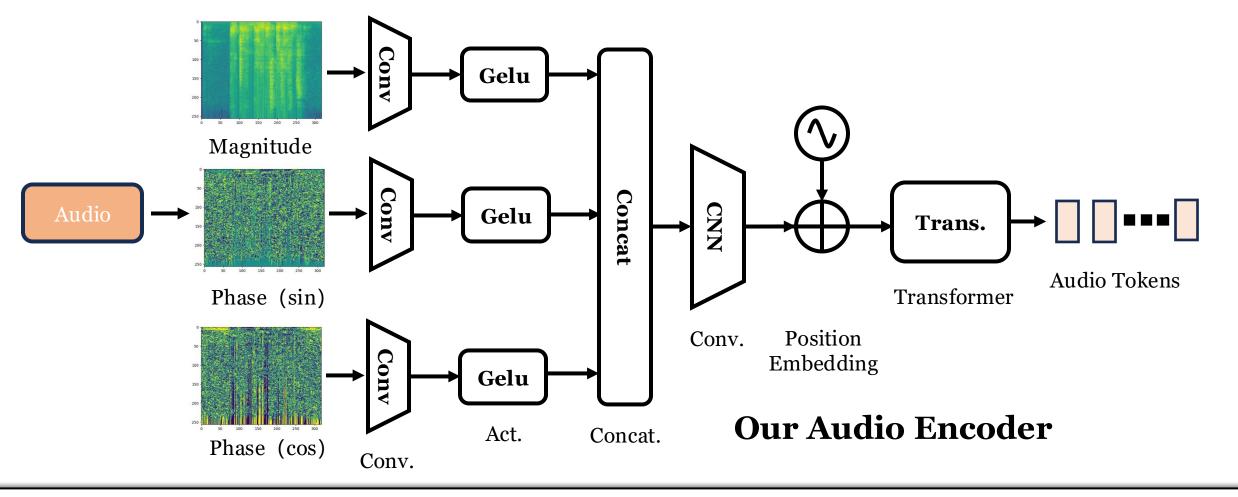
**Synthesis Workflow of** <Audio, Question, Answer>

# 6. Challenge II: Fine-grained Feature Extraction



### Audio Encoder (OpenAI Whisper)

**Problem:** Audio encoders like Whisper fall short for physical understanding. Whisper mainly captures magnitude features, which work well for speech recognition—but lack the finegrained phase information needed for physical cues



Solution: Our encoder incorporates both magnitude and phase (sin, cos) to retain physical characteristics of sound.

### 7. Main Results

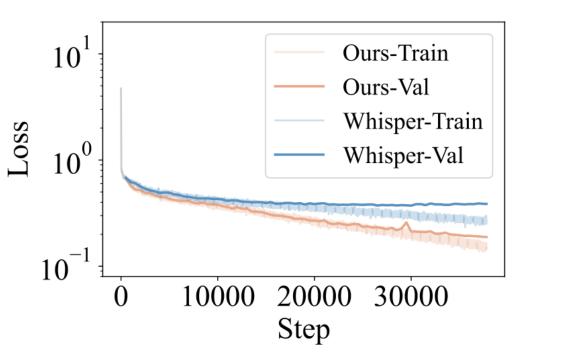
Table 2. Overall Performance. Values are presented as (Merged | Sole) where "Merged" indicates models trained on combined dataset and "Sole" indicates models trained separately for each task. By default, we focus on Merged results, with Sole results provided for reference.

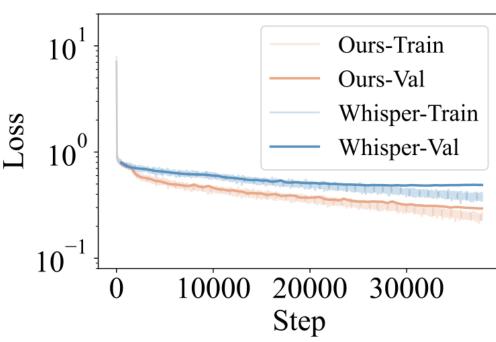
<b>Model Architecture</b>		Task Performances (Merged   Sole)				
Audio Encoder	LLM	<b>LOS Detection</b> BCA (†)	<b>Doppler Estimation</b> $MAE_f(\downarrow)$	<b>DoA Estimation</b> $MAE_t (\downarrow)$	Multipath Analysis TCA (†)	Range Estimation REP $(\downarrow)$
Whisper	Llama3.1-8B	0.867   0.906	1.213   3.147	5.585   5.601	0.845   0.889	12.572   17.182
	Qwen2-7B	0.881   0.910	1.042   0.575	2.716   6.873	0.848   0.897	10.609   12.901
ACORN	Llama3.1-8B	0.920   <b>0.965</b>	0.791   0.557	1.423   1.349	0.890   <b>0.945</b>	1.764   <b>1.446</b>
	Qwen2-7B	<b>0.924</b>   0.962	0.181   0.263	0.907   1.167	<b>0.903</b>   0.944	<b>1.599</b>   1.751
Performance on Open QA (Our Encoder + Qwen2-7B)		0.898   0.953	0.487   0.398	2.314   2.043*	0.906   0.908	2.852   1.900*
Random Baseline**		0.50	10.00	66.67	0.33	33.33

We compare two audio encoders: OpenAI's Whisper and our encoder proposed. We pair each encoder with two different large language models (LLMs): Llama3-8B and Qwen2 with 7B

#### **Key Findings:**

- 1. the feasibility of teaching LLMs to understand physical phenomena through sound
- 2. the superiority of our audio encoder over Whisper
- 3. the model-agnostic nature of our approach, evidenced by similar performance of different LLM architectures



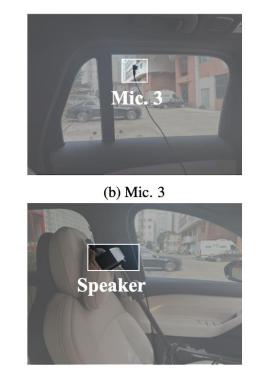


(a) Qwen2-7B

(b) Llama3.1-8B **Loss History** 

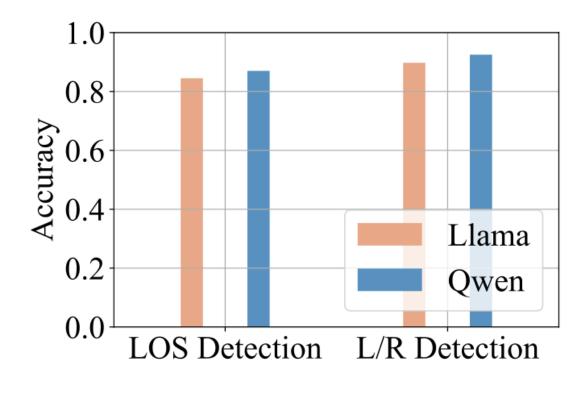
• Our approach achieves **faster convergence** and **lower final loss values** during training across both Llama and Qwen architectures





(e) Speaker at front seat





### **Real-World Deployment**

**Results** 

The results show the **practical viability** of our approach in the real world.

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