

SIMULATION-BASED PLATE-REVERB PARAMETER ESTIMATION FROM A SINGLE IMPULSE RESPONSE

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ABSTRACT

We present a simulation-trained, non-iterative estimator for Task A of the 1st DAFx Parameter Estimation Challenge. Each unnormalized plate-reverb impulse response is summarized by amplitude, spectral, and decay descriptors, and an ensemble of tree regressors estimates the six target parameters in one pass. Across two independent synthetic validation sets, the normalized models outperform the training-set mean and an earlier raw-regression baseline. On a shared set, the final ensemble also outperforms a single run of the official default PSO at substantially lower inference cost. Since the official labels are hidden, parameter accuracy is measured on simulator-matched data, and the released responses support only audio-side consistency checks. The estimator returns point estimates without uncertainty.

1. INTRODUCTION

The 1st DAFx Parameter Estimation Challenge asks how accurately physical audio-model parameters can be estimated from impulse responses.¹ The two challenge tasks operate at different levels of the same generative process: physical and observation parameters determine a modal response, whose damped components form the impulse response. Task A estimates the compact physical and observation description, whereas Task B estimates the generally much larger set of modal frequencies, decay rates, and amplitudes. This paper addresses Task A, which requires estimating the six-dimensional parameter vector

$$\theta = [\mu, D/\mu, T_0/\mu, L_y, x_o, y_o], \quad (1)$$

where μ is the surface density, D is the plate flexural rigidity, D/μ and T_0/μ are the normalized stiffness and tension parameters, L_y is the plate dimension in the y direction, and (x_o, y_o) denotes the output location at which the plate displacement is observed.

At inference time, each unknown parameter vector must be estimated from a single impulse response. The parameters affect different aspects of the response: the stiffness and tension ratios and the plate dimension shape the modal-frequency pattern, the surface density controls the absolute response scale, and the output location determines the relative visibility of the modes in the measured response [1, 2]. These cues are distributed across a dense

resonant response in which many modes overlap in frequency and time [3]. Estimation therefore requires the overall frequency, amplitude, and temporal structure rather than a few isolated spectral peaks.

One strategy first estimates a modal representation and then maps it to physical parameters. Classical modal analysis represents an impulse response as damped sinusoids and estimates their frequencies, decay rates, and amplitudes through Hankel-matrix, ESPRIT-type, or matrix-pencil formulations [4, 5]. Dense, overlapping resonances make this first stage difficult, and the resulting modal parameters still need to be mapped to the six Task A targets.

An alternative is analysis-by-synthesis, in which candidate physical parameters are passed through a simulator and iteratively updated to reduce the discrepancy between the simulated and target responses. Differentiable signal-processing and modal-simulation frameworks make this procedure amenable to gradient-based optimization [6, 7, 8]. These approaches retain an explicit physical model, but per-instance inference requires repeated simulation and iterative optimization for every target response. The challenge reference baseline follows this route using particle swarm optimization [9].

Learning-based inverse models instead move simulator evaluations to offline training. They can estimate parameters directly or initialize later optimization [10, 11]. We use the direct variant: a compact tree ensemble maps fixed-dimensional response descriptors to the six targets in one pass, and the official simulator then regenerates the corresponding response. We evaluate parameter accuracy on labeled synthetic data and forward consistency on the released responses, whose physical parameters are hidden.

2. METHOD

Figure 1 summarizes the offline training and test-time paths. The complete inference-and-regeneration pipeline is

$$h(t) \rightarrow \phi(h) \rightarrow \hat{z} \rightarrow \hat{\theta} \rightarrow \hat{h}(t), \quad (2)$$

where $h(t)$ is the target response, $\phi(h)$ is its feature vector, $\hat{z}$ is the normalized estimate, $\hat{\theta}$ is the estimate in physical units, and $\hat{h}(t)$ is the regenerated response. Simulator calls are used for offline training and final regeneration. Estimation itself requires only feature extraction and regression. Training pairs are produced by sampling raw parameters, synthesizing responses, and extracting $\phi(h)$. Separate labeled simulations are used for validation because the released labels are hidden.

¹<https://github.com/LOGUNIVPM/1st-DAFx-Challenge>.

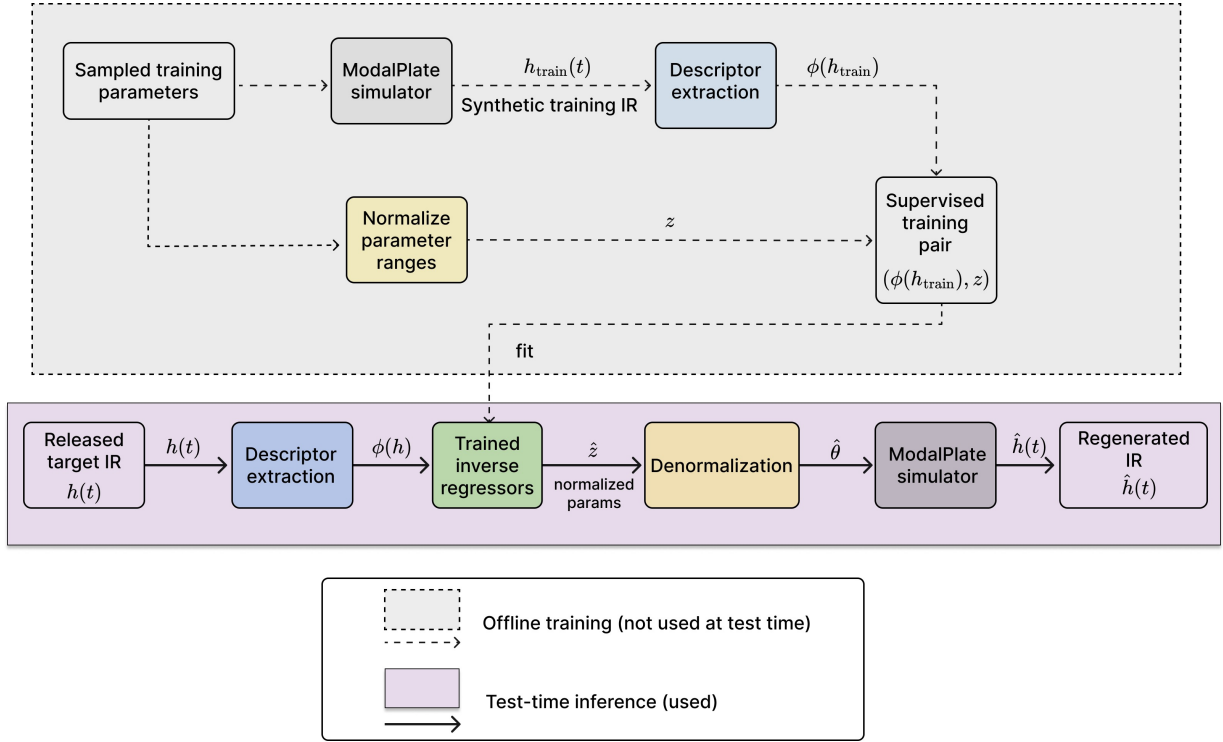


Figure 1: Overview of the Task A pipeline. Solid arrows show prediction for a released challenge response. Dashed gray arrows show how simulated training examples are made and used to fit the regression model.

2.1. Simulator-based data generation

Training data are generated with the public ModalPlate simulator². Variable raw simulator parameters are sampled from the public ranges, with $L_y \in [1.1, 4.0]$, $h \in [10^{-3}, 5 \cdot 10^{-3}]$, $T_0 \in [10^{-2}, 10^3]$, $\rho \in [2430, 21230]$, $E \in [6.7 \cdot 10^{10}, 2.2 \cdot 10^{11}]$, and $x_o, y_o \in [0.51, 1.0]$. We synthesize 5 s displacement impulse responses matching the released target duration. The corresponding identifiable parameters θ are retained as supervised labels. The final estimator is trained on 1000 synthetic examples. We use two independently generated labeled sets that are not included in training. *Validation 1* contains 24 responses and was used during model development. *Validation 2* contains 12 responses and was generated only after the final estimator was fixed, providing a fresh check against dependence on one favorable validation set.

2.2. Response feature extraction

Each impulse response is mapped to a fixed-length feature vector designed to summarize response scale, spectral shape, and decay behavior. The final feature vector contains 372 scalar values: global time-domain and energy summaries, early-to-late energy ratios, full-response spectral centroid and tilt, statistics from nine broad spectral bands, group-delay summaries over four bands, and six broad frequency-region summaries with band-limited decay statistics. These features are computed directly from the unnormalized response, without using target metadata.

²<https://github.com/LOGUNIVPM/1st-DAFx-Challenge/tree/main/ModalPlate>.

2.3. Predicting normalized parameters

The six target variables have different scales, so each target is first mapped to its normalized position inside its allowed range,

$$z_i = \frac{\theta_i - \theta_i^{\min}}{\theta_i^{\max} - \theta_i^{\min}}. \quad (3)$$

The derived ranges induced by the raw simulator bounds are $\mu \in [2.43, 106.15]$, $D/\mu \in [0.28, 201.2]$, and $T_0/\mu \in [9.4 \cdot 10^{-5}, 411.5]$. For L_y and (x_o, y_o) , which are already target variables in the simulator, we use their public bounds directly. These ranges follow from the public raw-parameter bounds using $\mu = \rho h$, $D = Eh^3/[12(1 - \nu^2)]$, and therefore $D/\mu = Eh^2/[12(1 - \nu^2)\rho]$ and $T_0/\mu = T_0/(\rho h)$, where the Poisson ratio is fixed at $\nu = 0.25$.

The final estimate averages two regressors trained on the same feature-target pairs, namely an ExtraTrees (ET) regressor [12] and a histogram gradient-boosting (HGB) regressor [13]. We train one model for each of the six normalized parameters. The ET model uses 500 trees, minimum leaf size 2, and robust feature scaling. The HGB model uses 250 boosting iterations, learning rate 0.04, ℓ_2 regularization 10^{-3} , and standard feature scaling. For a target response,

$$\hat{z}_r = F_r(\phi(h)), \quad r \in \{\text{ET}, \text{HGB}\}, \quad (4)$$

$$\hat{\theta}_r = C_\theta \left(\theta^{\min} + \hat{z}_r \odot \Delta\theta \right), \quad (5)$$

$$\hat{\theta} = C_\theta \left(\frac{\hat{\theta}_{\text{ET}} + \hat{\theta}_{\text{HGB}}}{2} \right), \quad (6)$$

where F_r denotes the six per-parameter regressors in model family r , $\Delta\theta = \theta^{\max} - \theta^{\min}$, C_θ clips each component to its allowed physical range, and $\odot$ denotes element-wise multiplication.

2.4. Forward regeneration for submission

The challenge also requires an impulse response generated from each estimate. Because the simulator accepts raw parameters such as (ρ, h, E, T_0) rather than $(\mu, D/\mu, T_0/\mu)$, we choose a feasible raw tuple matching the estimated invariants and synthesize its response. This is forward regeneration, not a separate audio-matching optimization.

3. EXPERIMENTS

Parameter accuracy is evaluated on Validation 1 and Validation 2, since the physical parameters of the released targets are hidden. For a set of N responses, we report

$$\text{NMSE} = \frac{1}{6N} \sum_{n=1}^N \sum_{i=1}^6 \left(\frac{\hat{\theta}_{n,i} - \theta_{n,i}}{\theta_i^{\max} - \theta_i^{\min}} \right)^2, \quad (7)$$

where lower is better. We compare the mean of the 1000 training targets, an earlier raw multi-output ExtraTrees model, normalized per-parameter ExtraTrees and HGB models, and their final ensemble. The earlier ExtraTrees candidate uses 246 generic descriptors rather than the final 372-feature set, so it is a historical system baseline rather than a controlled normalization ablation. On Validation 2, we additionally run the official PSO baseline with its public default configuration: 10 particles, five iterations, $(w, c_1, c_2) = (0.2, 2, 1)$, and multi-scale spectral loss with STFT configurations (512, 128), (2048, 512), and (8192, 2048). We evaluate one fixed-seed realization of this stochastic baseline.

We separately assess forward-response consistency on the 16 released targets. Using the unnormalized target and regenerated responses, we report absolute RMS-level error in dB. We also report log-spectrum RMSE up to 12 kHz after independently peak-normalizing both spectra and applying an -80 dB floor. Temporal energy distribution is measured by $|\tau(\hat{h}) - \tau(h)|/\tau(h)$, where the energy-time centroid is

$$\tau(h) = \frac{\sum_n n h[n]^2}{f_s \sum_n h[n]^2}. \quad (8)$$

These are local audio-side diagnostics, not official parameter scores.

All runtimes were measured on a MacBook Pro with an Apple M1 Pro 10-core CPU and 16 GB unified memory under macOS 14.5, without an external CUDA GPU. Timing for the learned estimator excludes synthetic-data generation and training. It processed all 16 released responses in 15.03 s. Since inference uses one feature-extraction and regression pass, the reported number of optimization iterations/trials is zero.

4. RESULTS

Table 1 compares parameter accuracy for the final ensemble, its component models, and the baselines.

The normalized per-parameter models outperform the training-set mean and raw multi-output baseline on both sets. On Validation 2, the final ensemble also reduces NMSE by 71.9%

Table 1: NMSE on two independent synthetic sets. Lower is better. Bold marks the column minimum, and “–” denotes an unevaluated configuration.

Model	Validation 1	Validation 2
Training-set mean	0.047229	0.047080
Official PSO (default)	–	0.046023
Raw ExtraTrees candidate	0.026776	0.022619
Normalized ExtraTrees	0.012163	0.014946
Normalized HGB	0.012426	0.012063
Final tree ensemble	0.011886	0.012935

relative to the official default PSO run. The ensemble is best on Validation 1, whereas HGB is slightly better on Validation 2, so averaging is not uniformly beneficial.

The PSO run required 2252.59 s for Validation 2, compared with 10.75 s for the final ensemble, including model loading and feature extraction. This is a 210-fold reduction in parameter-estimation time at the tested default PSO budget.

Table 2 breaks down the final ensemble error by parameter. The normalized stiffness and tension terms have the lowest errors on both sets. The largest error is associated with an output-position coordinate, although the harder coordinate differs between sets.

Table 2: Final ensemble per-parameter NMSE. Lower is better. Bold marks the column minimum.

Parameter	Validation 1	Validation 2
μ	0.009007	0.009952
D/μ	0.000445	0.000059
T_0/μ	0.000910	0.000732
L_y	0.016485	0.021350
x_o	0.032235	0.022773
y_o	0.012235	0.022743

Table 3 summarizes forward-response consistency over all 16 released targets. IR 0012 is the largest mismatch for all three diagnostics. Four illustrative listening, waveform, and spectrogram comparisons are available at <https://minhuilu.github.io/dafx26-taska-demo/>.

Table 3: Audio-side diagnostics on the 16 released targets.

Diagnostic	Median	Largest mismatch
Absolute RMS-level error	1.04 dB	18.56 dB, IR 0012
Log-spectrum RMSE	4.63 dB	13.66 dB, IR 0012
Energy-time error	1.23%	25.19%, IR 0012

5. DISCUSSION

The similar ensemble NMSE on the two independent synthetic sets supports repeatability within the matched simulator distribution rather than dependence on one favorable split. The much smaller errors for normalized stiffness and tension than for output position suggest that global modal-frequency patterns are easier to identify

than mode-visibility patterns from one observation point. The reversal between HGB and the ensemble across sets also cautions against claiming a uniform benefit from model averaging.

The released-target diagnostics show that forward-response consistency varies across targets, but they cannot establish parameter accuracy because the labels are hidden. Parameter error and response similarity are related but not interchangeable: small parameter errors can shift modal frequencies and accumulate phase differences, whereas different estimates may retain similar spectral or temporal characteristics. The audio diagnostics therefore complement rather than replace the synthetic parameter evaluation.

The official PSO comparison shows that amortizing simulator calls across offline training can outperform a low-budget per-target search at much lower test-time cost, consistent with learned inverse estimation [10, 11]. It is nevertheless one stochastic realization. Larger swarms, more iterations, or restarts may improve PSO, while our timing excludes data generation and training. We also did not compare with differentiable refinement [7, 8]. Both validation sets are small and simulator-matched, the historical baseline is not a controlled ablation, and the estimator provides no uncertainty. Future work should isolate each design choice and study uncertainty-aware refinement, mismatched or measured plates, and multiple observation locations.

6. CONCLUSION

We presented a fast, simulation-trained Task A estimator. On two independent synthetic sets, the normalized tree models outperform the local baselines. On a common synthetic validation set, the final ensemble also surpasses an official default PSO run at much lower inference cost. The pipeline provides a reproducible baseline for single-response plate-reverberation inversion and could serve as a fast initializer for uncertainty-aware physical refinement. Uneven parameter errors and released-target consistency also motivate evaluation on measured plates and with multiple observation points.

7. REFERENCES

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