

Valve vs. solid-state microphone preamplifier: a comparative study

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Abstract

This paper outlines research carried out to determine the perceptual and objective differences between a solid-state and a valve preamplifier running at low voltages. ABX testing was employed and showed that there were perceivable differences between the two systems. A comprehensive objective analysis was performed, which utilised tests for total harmonic distortion + noise (THD+N), intermodulation distortion (IMD), THD versus frequency and frequency response in order to ensure the two systems were performing in their linear region. In addition, MIRTtoolbox was utilised to extract low-level features such as spectral centroid, skewness and novelty. The electronic measurements combined with the MIRTtoolbox support the listeners' subjective descriptors that there is a difference in brightness and harmonic content between the two types of preamplifiers. A correlation theory was developed, which linked the objective and the subjective measurements.

Keywords: Valves; solid-state; preamplifiers; objective and subjective measurements; THD+N; IMD; spectral centroid; skewness;

Introduction

Within the audio industry there are ongoing and controversial debates over the differences in tonality and perceived sound quality of valve and solid-state amplification. Microphones are connected to a preamplification stage and the choice of this stage is an important part of the signal chain.

The main focus of this study is to determine whether there are any perceivable differences in sound between valve and solid-state preamplification operating at the same low voltage (24V), and, if so, what the exact variables are that contribute to those differences. Both preamplifiers were analysed while operating in their linear region and keeping both inputs and outputs the same in order for the measurements to solely reflect the differences that might arise due to amplification.

Circuit design

Two preamplifiers were built, one based around a solid-state design, the other valve based. Both circuits utilised identical input and output stages to try to ensure any perceptual differences were due to the amplification stage only.

Input stage

The input stage (Figure 1) features an electronically balanced input, with a transistor long-tailed pair, with constant current source on the collector and a PNP complementary feedback pair.

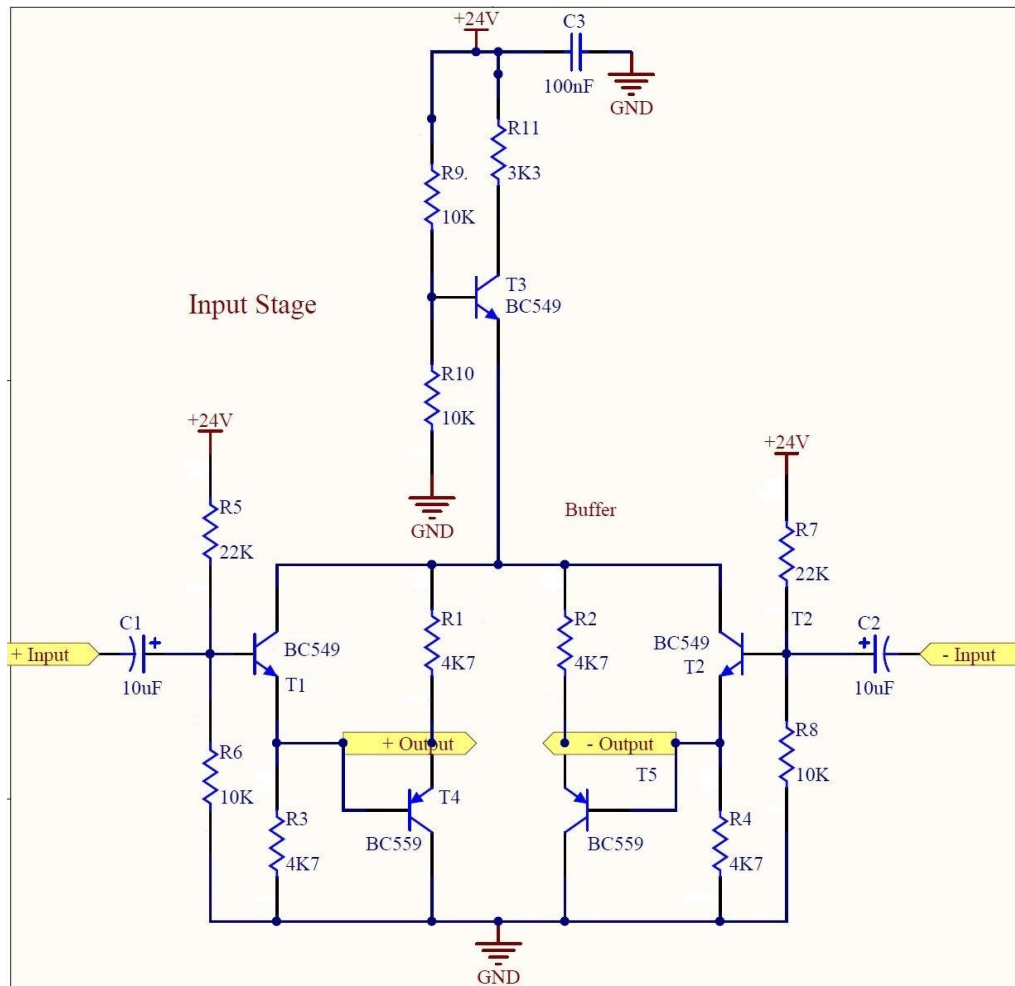


Figure 1: Input Buffer

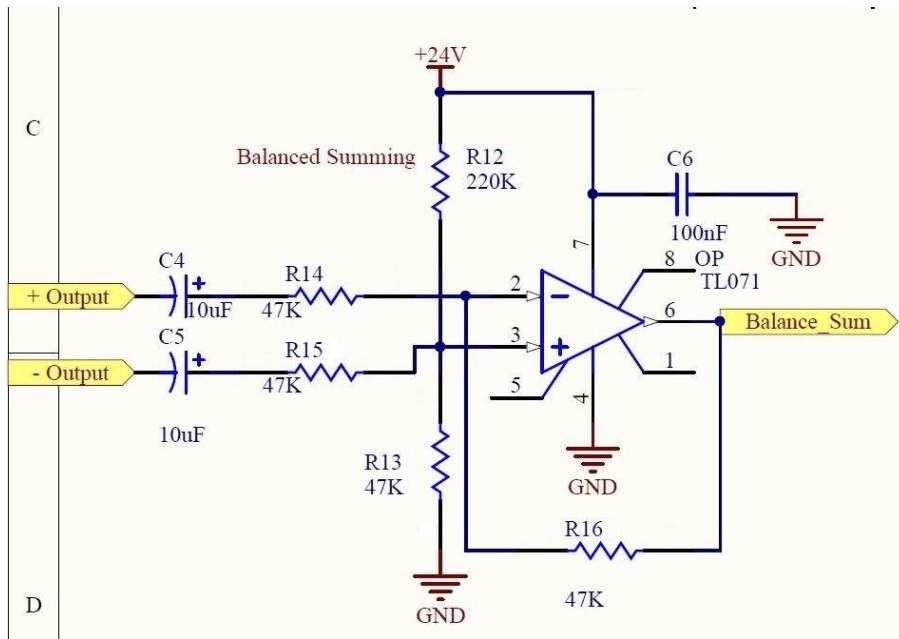


Figure 2: Summing Op-Amp

The stage presented in Figure 2 has the role of summing the signals from the buffer, converting them into a single-ended output.

Amplification stage

Solid-state

In both cases, the amplification stages (Figure 3 and Figure 4) are kept as similar as possible in order for the results to be comparable, so both preamplifiers feature a two-stage amplification topology, AC coupling, adjustable gain and a similar number of components.

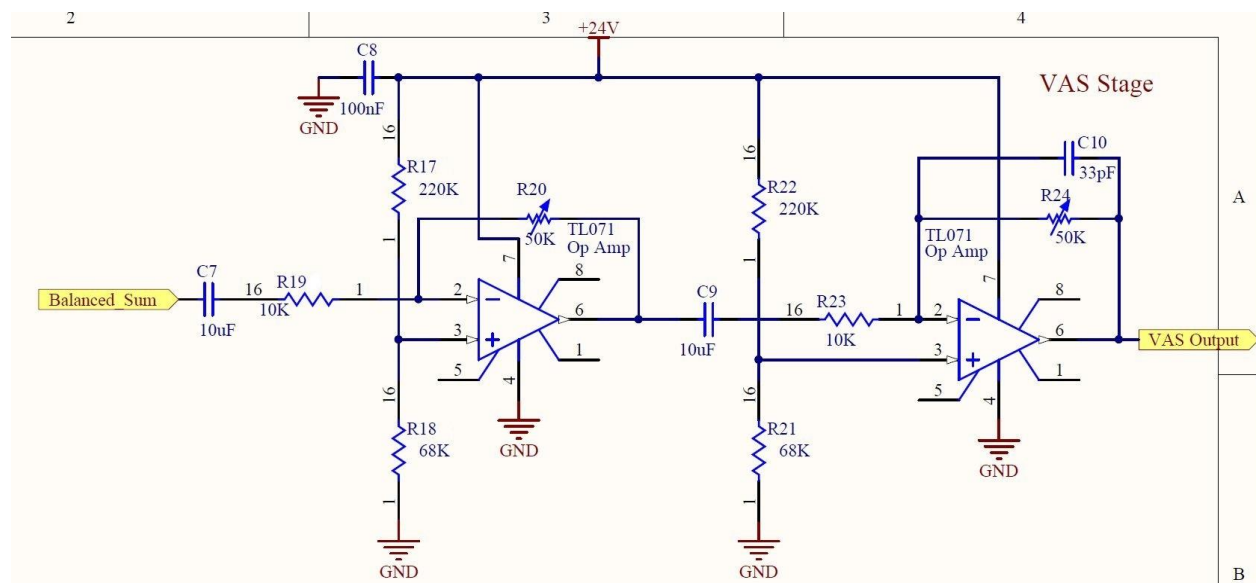


Figure 3: Solid-State Voltage Amplification Stage

Valve

All the working currents were obtained using the load-line from Figure 5, provided by Makarewicz (2014).

To allow for signal headroom, a grid bias voltage of $-0.5V$ was chosen, which, in theory, allows for inputs of up to approximately $500mV_{p-p}$, with the result that the cathode had to be biased at $0.5V$. Based on the load-line, when the swing pulled the anode at $0V$, resulting in $24V$ across the anode resistor, the maximum flowing current would be $1.22mA$.

By using the load-line, the transconductance, plate resistance and amplification factor were calculated according to Elliot (2009) and represent the DC quiescent condition:

$$gm = \frac{\Delta I}{\Delta V} = \frac{0.7-0.2}{1V} = \frac{0.5mA}{1V} = \frac{500\mu A}{V} = 500\mu mhos - \text{calculated at constant plate voltage}$$

$$rp = \frac{\Delta Ep}{\Delta Ip} = \frac{24-12}{1.22-0.2} = \frac{12}{1.02} = 12k\Omega - \text{calculated at constant grid voltage}$$

$$\mu = gm * rp = 6.3$$

$$Av = \mu * \frac{R_{tot}}{rp + R_{tot} + (\mu + 1) * R_k}$$

$$Av = 4.15 / \text{triode}$$

$$R_{tot} = Rp \parallel R_{load} = 19k\Omega \parallel 470k\Omega = 19k\Omega$$

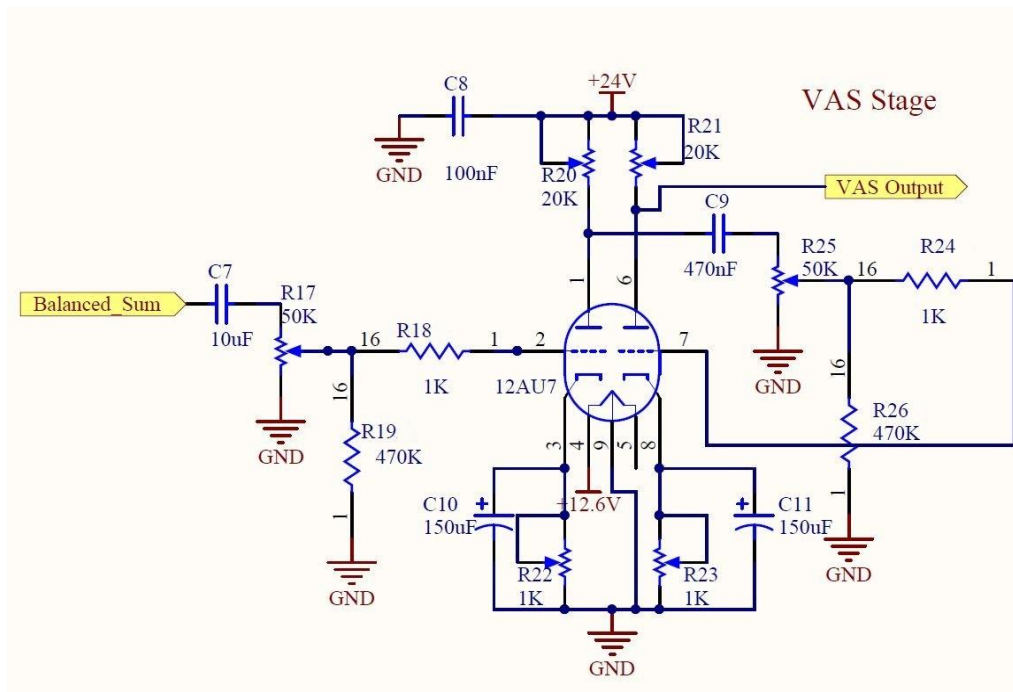


Figure 4: Valve Voltage Amplification Stage

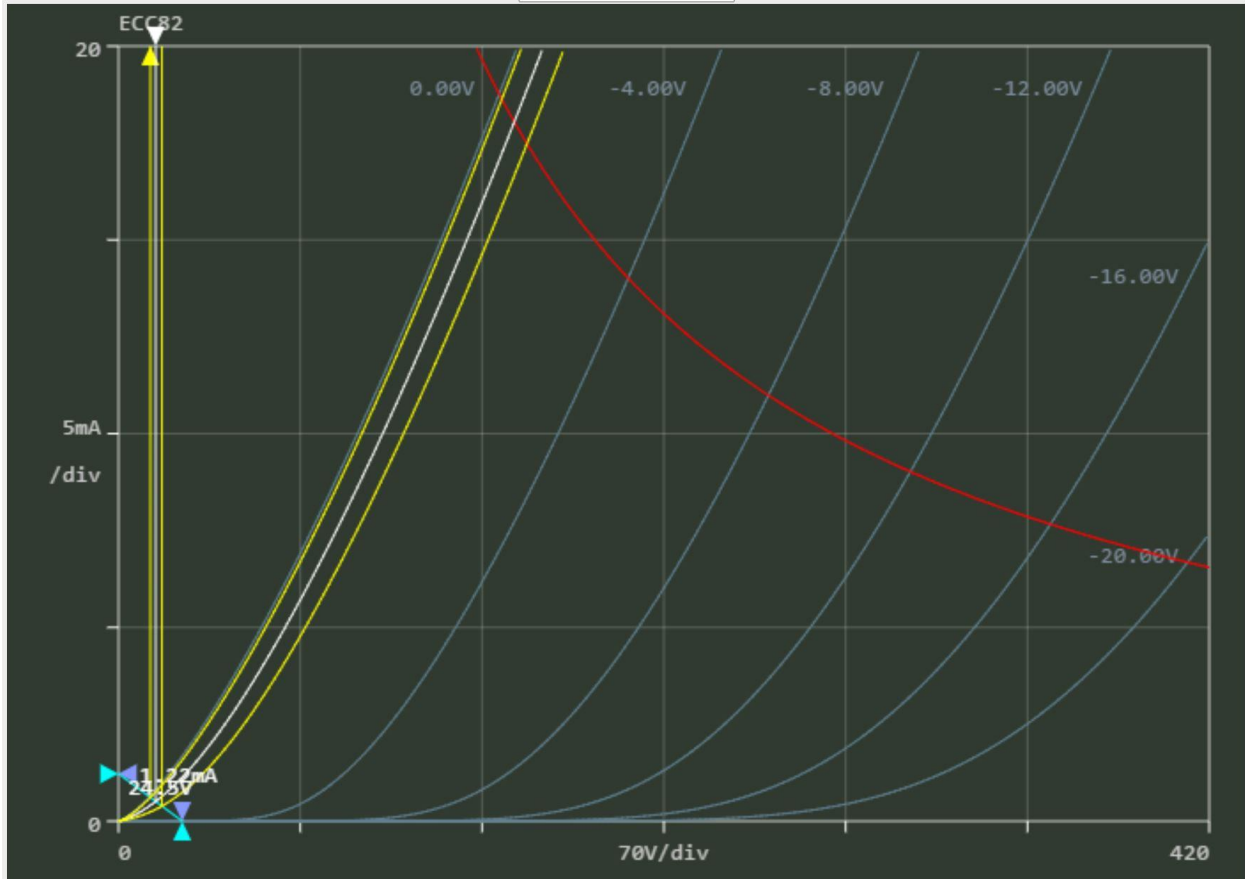


Figure 5: Valve Load-Line

Output stage

The output stage that can be seen in Figure 6 features an op-amp in voltage-follower configuration used primarily to convert a high-input impedance to a low-output impedance. The input impedance is dictated by the equivalent resistor formed by the voltage divider. The output impedance is approximately 500Ω .

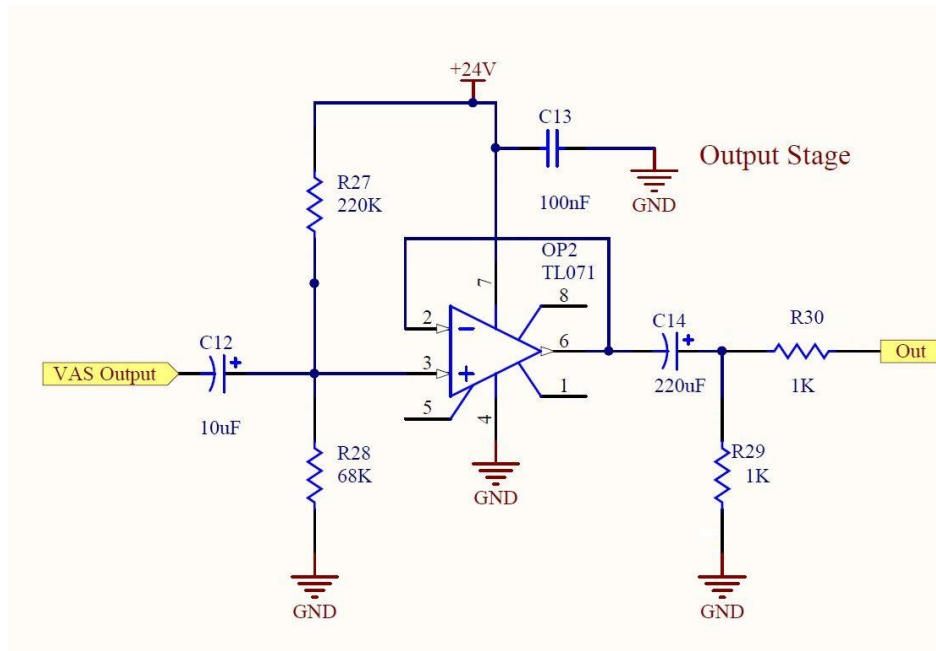


Figure 6: Output Stage

Key audio measurements

Objective measurements

Objective measurements employed in this study consisted of both electronic measurements and the use of a low-level feature extraction tool called MIRTtoolbox (Lartillot, 2013).

All electronic measurements were carried out using a PRISM dScope III audio analyser, and parameters Total Harmonic Distortion + Noise (THD+N), intermodulation distortion (IMD), individual harmonic magnitude and frequency response were measured. The systems were tested for their transient response using a square wave generator. All measurements were taken under a load of 10k Ω .

The THD parameter is one of the most important as it provides information on the harmonic distortion of a system over the gain range or the frequency range and is used to characterise the linearity of audio systems and the power quality of electric power systems. As noted by Metzler (1993), it is a measurement of the harmonic distortion present in the system and is defined as the ratio of the sum of the powers of all harmonic components to the power of the fundamental frequency.

THD+N is measured in the same way as THD, but the root mean square (RMS) voltage of the noise is also added, as stated by Palmer (2005). This test is used as valves usually exhibit higher noise magnitudes than solid-state, which can have a potential impact on the subjective perception.

The first three harmonics were also measured individually for each preamplifier, as they are the most representative, having the highest distortion values and theoretically being responsible for any perceived differences.

The frequency response factor can add to the sound timbre depending on the amplitude of certain frequencies, and can introduce perceivable differences if one lacks low-frequency response or high-frequency response.

As Elliot (2012) specifies, the IMD test is more meaningful than a THD analysis because it gives the distortion values for the products which are not harmonically related to the pure input signal. The Society of Motion Picture and Television Engineers (SMPTE) specifies the two signals to be at 60Hz and at 7kHz which are summed at a 4:1 amplitude ratio (12dB ratio), as described by Bohn (2000). Metzler (1993) summarises IMD as the amplitude modulation of signals containing two or more frequencies, which is caused by having a non-linear behaving system.

Square wave testing is very meaningful for checking the system's transient response according to Elliot (2015), being particularly suitable for correlating the findings with the spectral centroid or the skewness, tests that are presented below.

Using MIRToolbox, attack slope was measured in order to detect the attack/rise times of the notes played on an instrument. This test, together with the novelty curve, is used to determine whether there are any differences in transient response of the two preamplifiers. Spectral centroid measure was carried out, as recommended by Zacharov and Bech (2006). The test looks into the shape of a distribution through the use of its moments, and is a measure that has been shown to relate to the perceived brightness of an audio signal. Skewness was also measured, from which transient response can be predicted, by identifying where the most energy is centred, depending on the skew, which can be either positive or negative.

Listening experiment

Proposed stimuli

Four samples were utilised for testing. A piano was used for its rich harmonic content, an electric bass for the low frequencies, and a guitar with simple struck chords for its envelope characteristics and high number of harmonics distribution. In addition, a cello was used for its sustained notes to see how the system reacts to constant energy levels, the difference between the cello and bass samples being the higher harmonics distribution of the former. All samples utilised in the testing were loudness normalised using a ITU-1770-4 standard loudness meter.

Test method

An ABX testing method was used for this project, in line with the ITU BS.1116 recommendations. The subjects were asked to listen to stimuli A and B, one of which represented the sample recorded through the valve preamplifier and the other the same sample recorded through the solid-state preamplifier; subjects were not told which

sample was which. The subjects were also given another stimulus, X, which was chosen randomly from either A or B, and were asked to say if X was A or B. In order for the measure to be considered valid, the subjects underwent 10 trials per set of samples to ensure that the final result was neither random nor based on luck. The subjects did not know what the test was investigating, or that samples A and B were different, thus reducing biasing effects. Also, the afferent deviation was calculated according to instructions provided by Koch (n/a)

After the test had ended, the subjects were asked to comment on the main differences perceived between the solid-state and the valve sound (as they were told at the end which was which). From this, a list of descriptors was collected.

Results

Subjective results

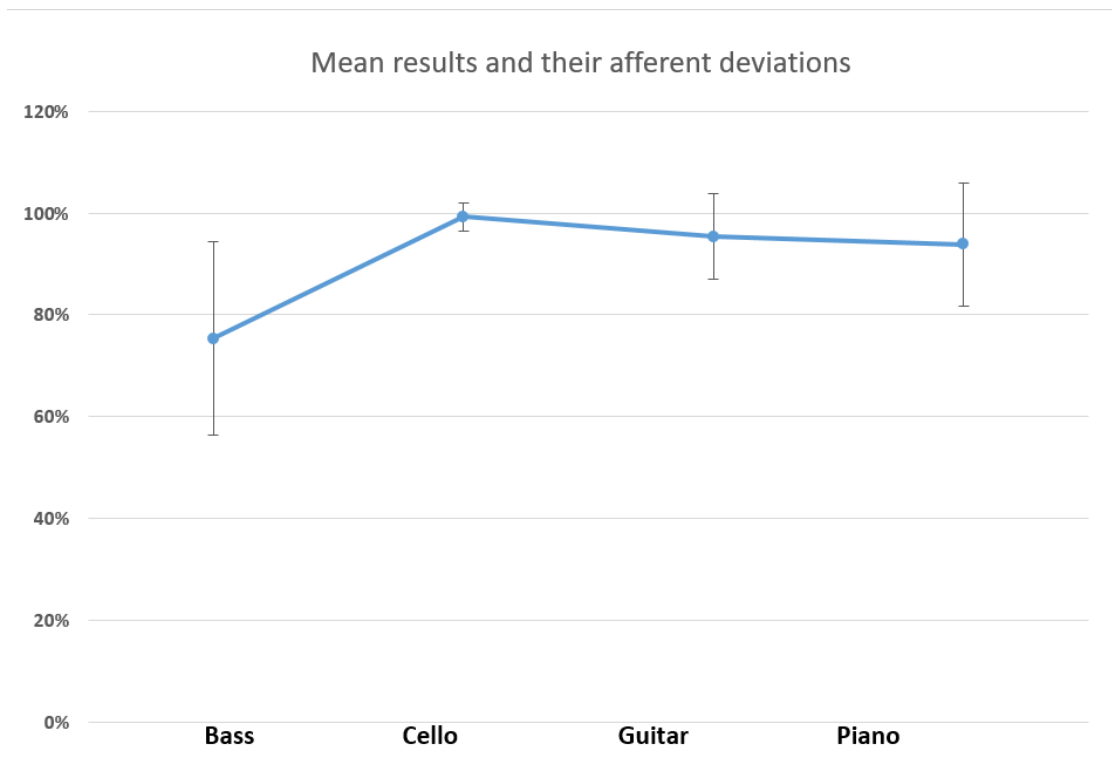


Figure 7: Mean results and their afferent deviation

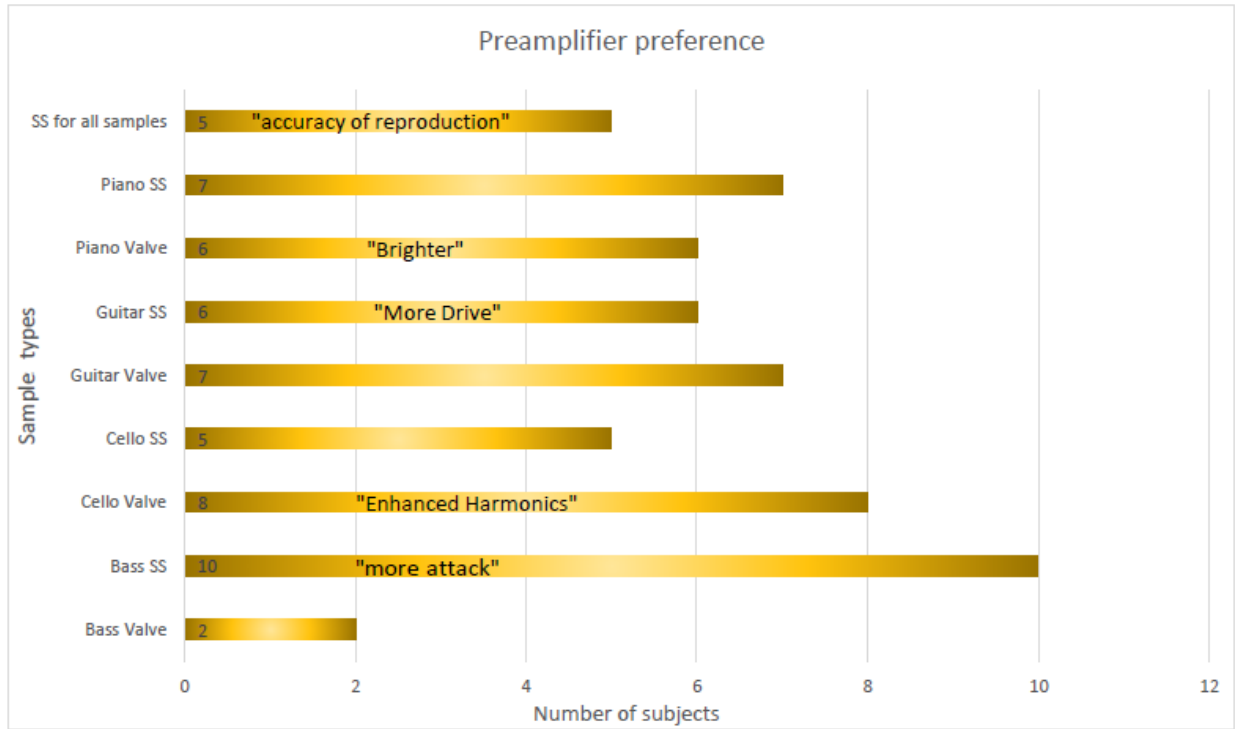


Figure 8: Listeners Preference and Descriptors Chart

Figure 8 summarises all descriptors that were provided by the listeners.

Objective results

THD+N

Values were measured for a 100mVp-p input at 1kHz with maximum gain and the results are shown in Table 1.

	Valve Preamplifier	Solid-State Preamplifier
THD+N absolute, 100mV input, max gain	-53.5dBu	-58.7dBu
THD+N relative, 100mV input, max gain	0.56%	0.136%

Table 1: THD+N levels for the two preamplifiers under test

Second, third and fourth individual harmonics

	2nd Harmonic	3rd Harmonic	4th Harmonic
Valve	-33.48 dB	-34.04 dB	-39.37 dB
Solid-state	-39.25 dB	-33.32 dB	-34.66 dB

Table 2: Individual Harmonics Measurement

Frequency response

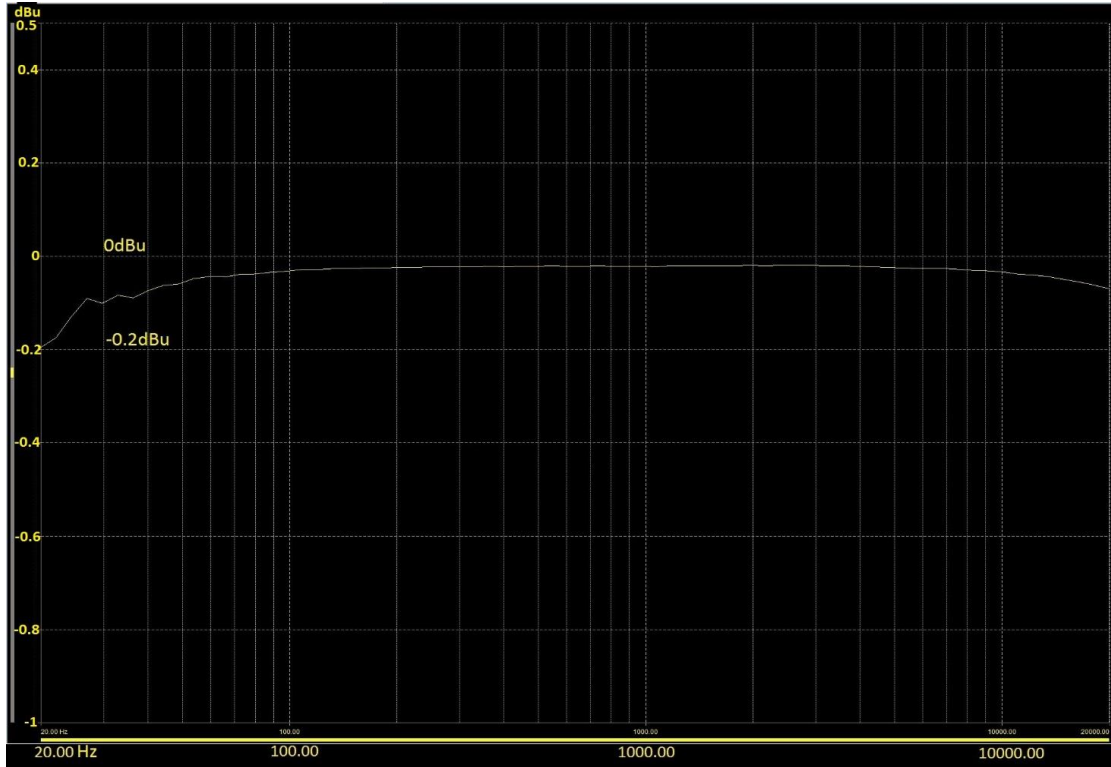


Figure 9: Solid-State Frequency response

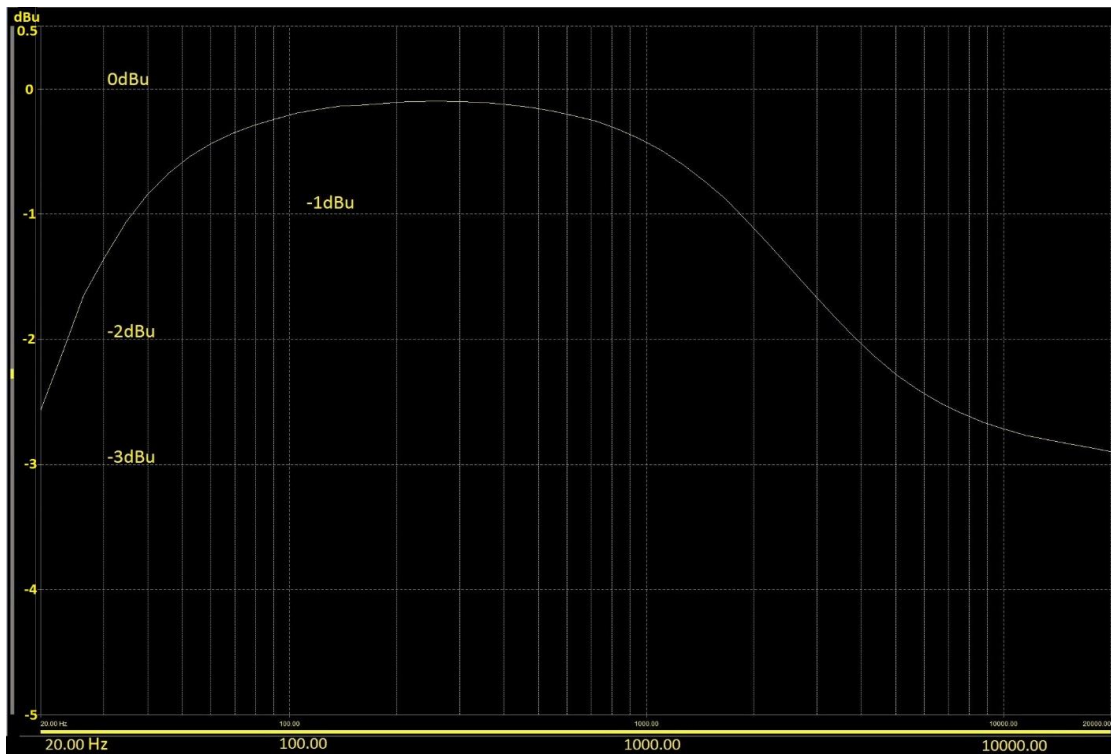


Figure 10: Valve Frequency Response

IMD – SMPTE Standard

Valve preamp	0.03%, 60Hz, 7kHz, 4:1 ratio, maximum gain
Solid-state preamp	0.0057% 60Hz, 7kHz, 4:1 ratio, maximum gain
In-Out configuration	0.0055% 60Hz, 7kHz, 4:1 ratio, maximum gain

Table 3: IMD Values

Square wave response

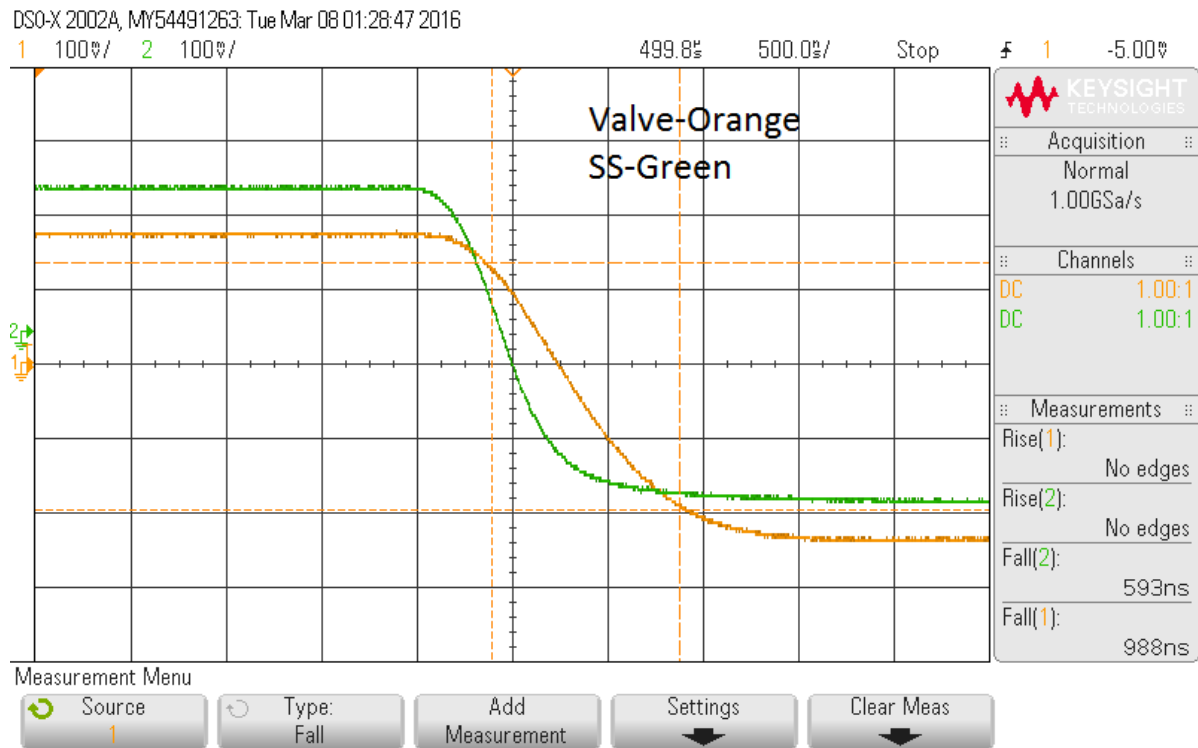


Figure 11: Fall Times

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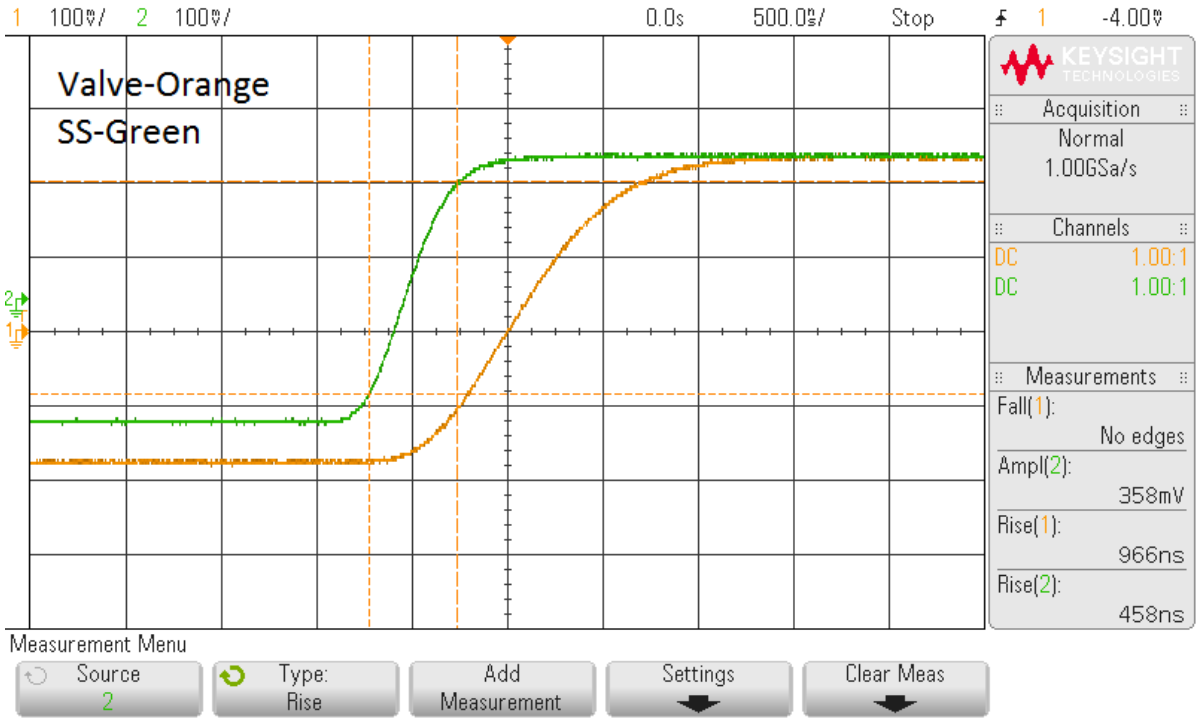


Figure 12: Rise Times

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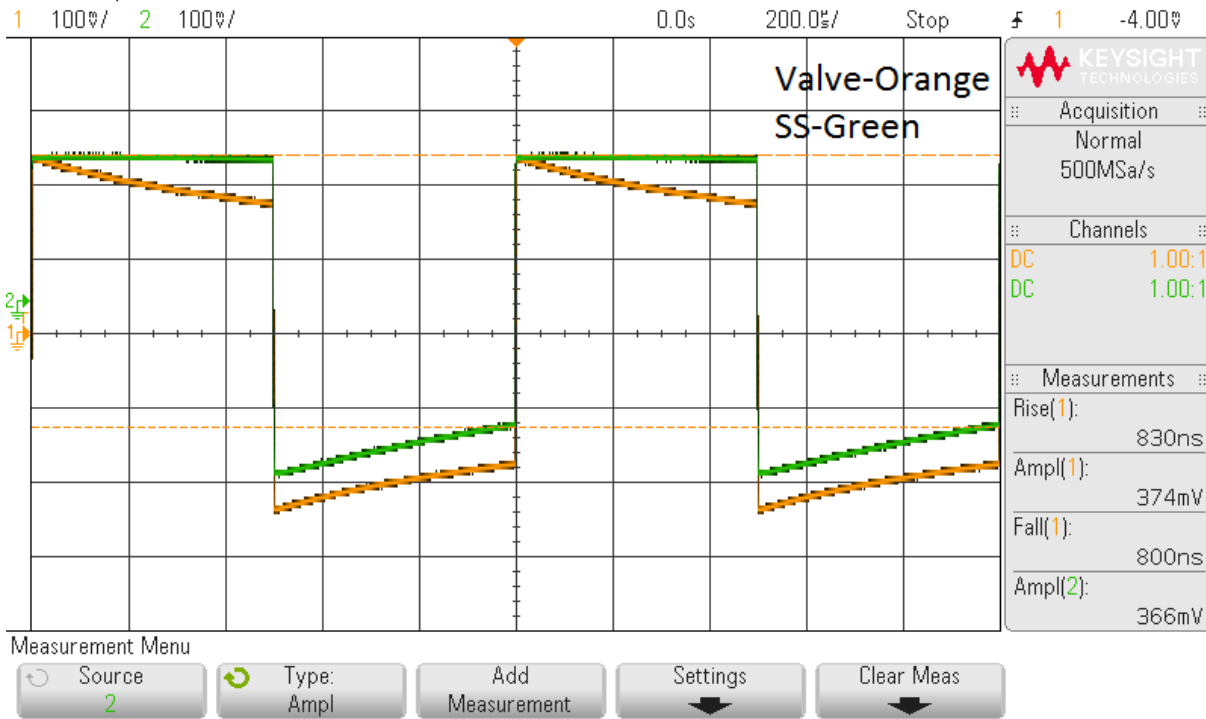


Figure 13: Frequency response predictable from the slopes

MIRToolbox measurements

Attack slope

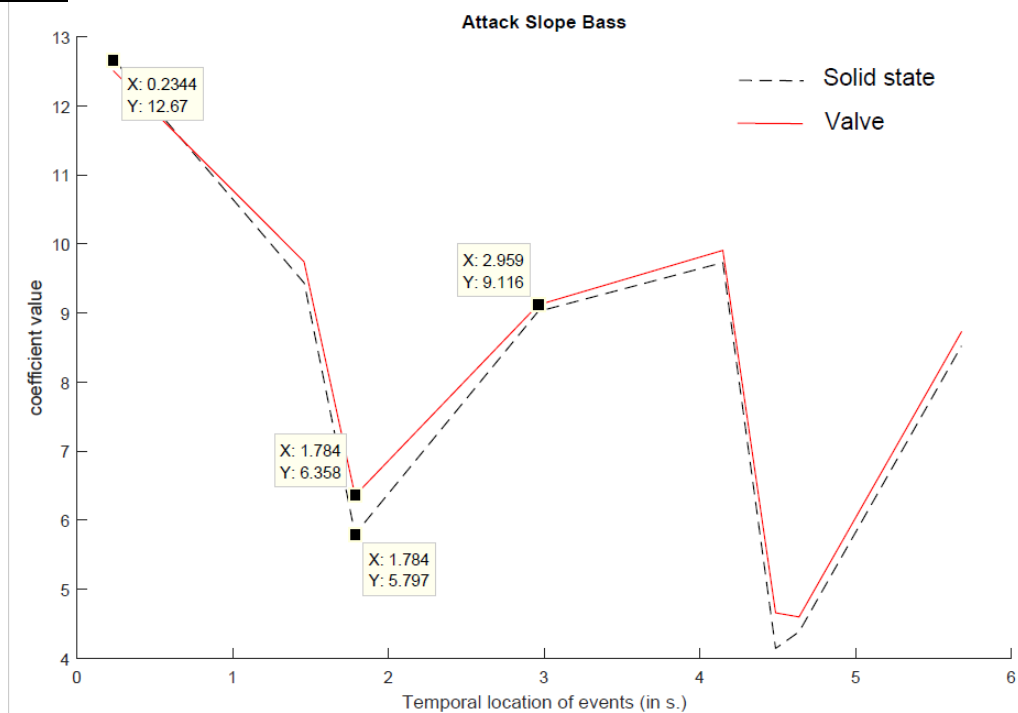


Figure 14: Bass attack slope comparison

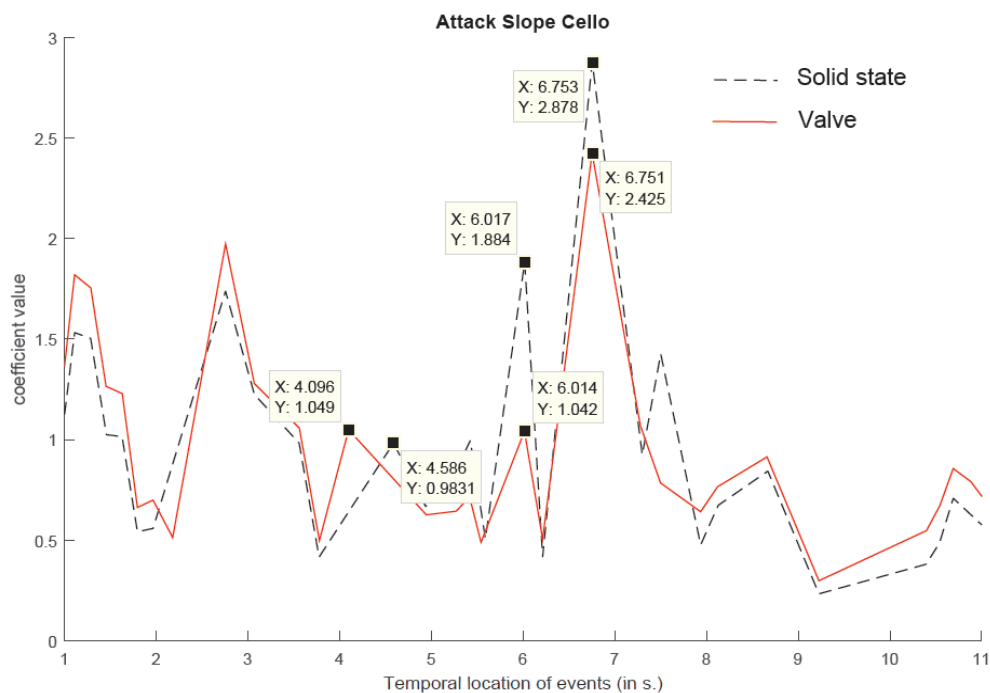


Figure 15: Cello attack slope comparison

Novelty curve

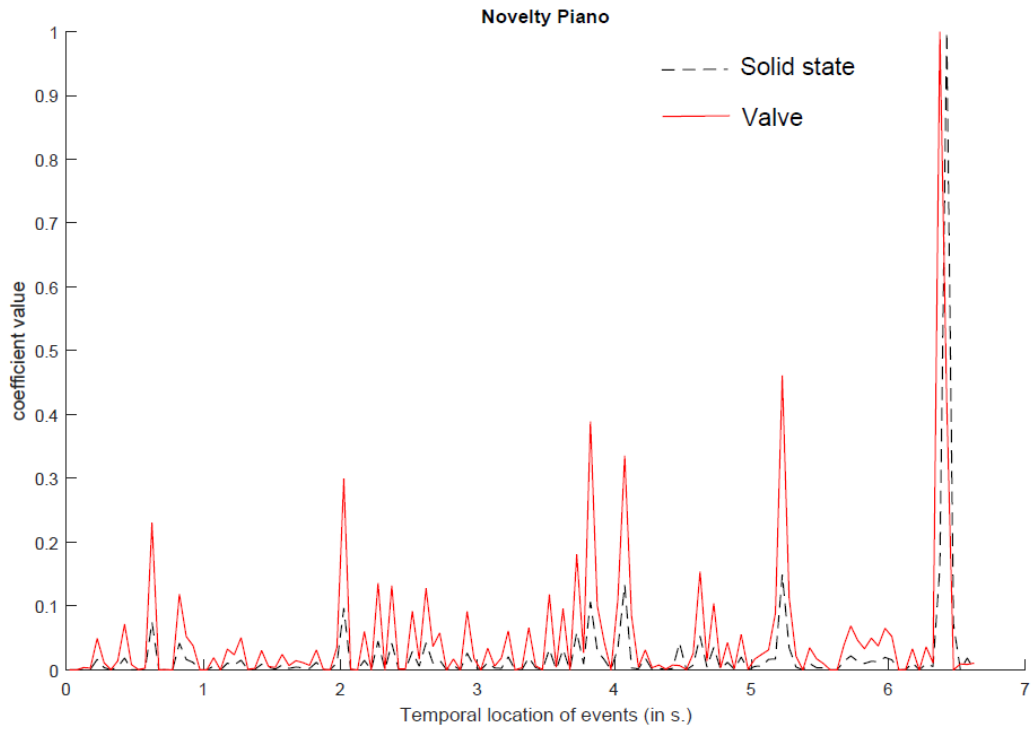


Figure 16: Piano novelty difference

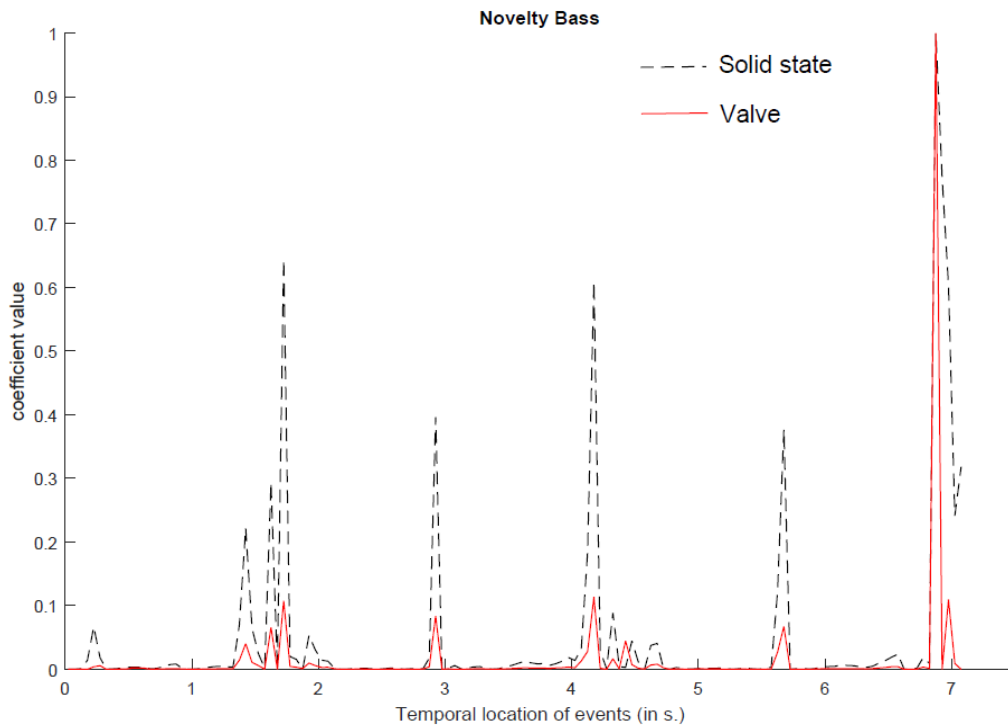


Figure 17: Bass novelty difference

Spectral centroid

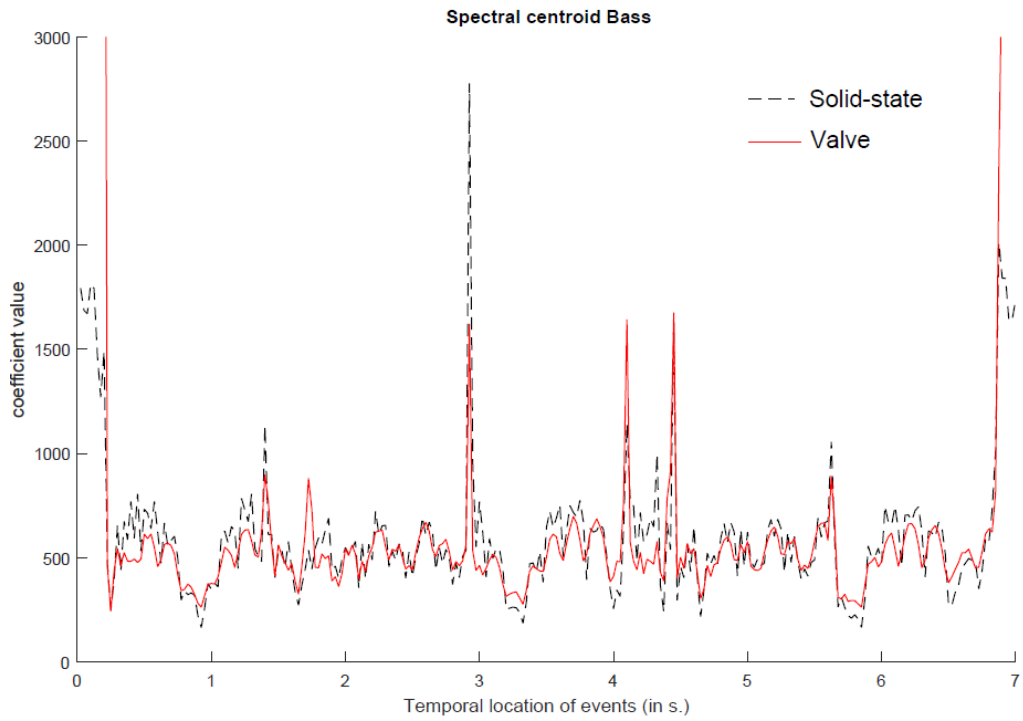


Figure 18: Bass Centroid

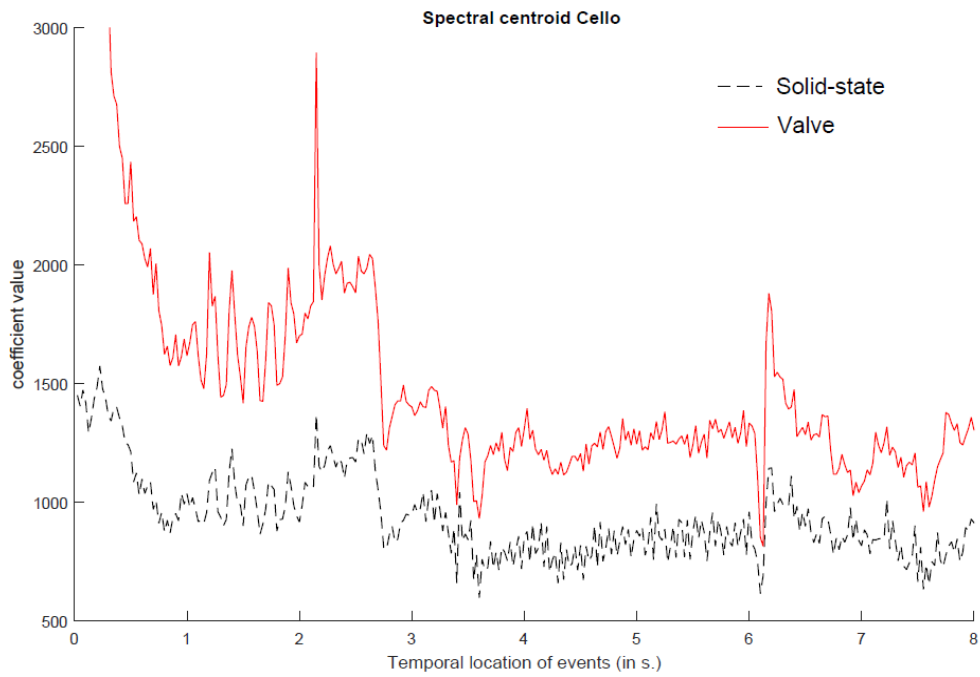


Figure 19: Cello Centroid

Skewness

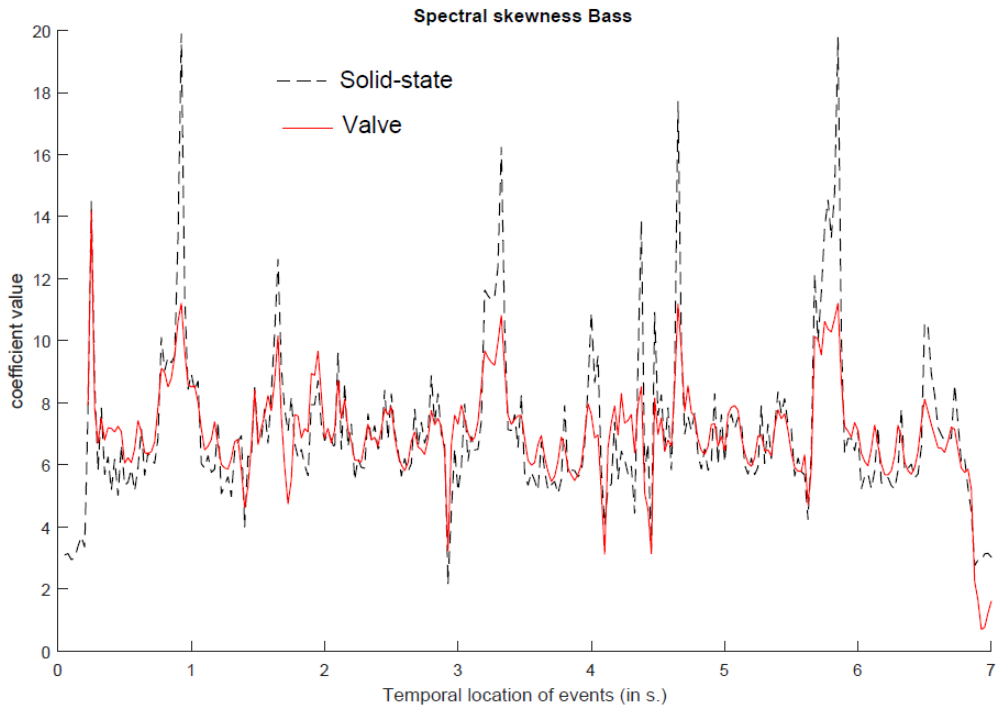


Figure 20: Bass Skewness

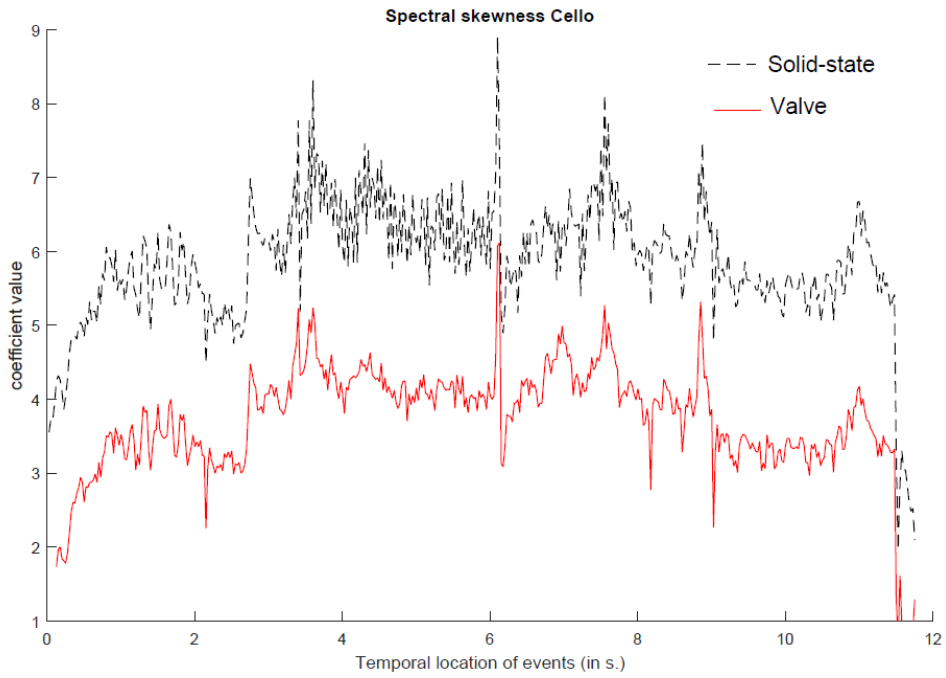


Figure 21: Cello Skewness

Analysis of results and correlation theory

First of all, it is important to discuss the results from Figure 7 and Figure 8. The poorer bass identification percentage was expected before testing because the bass is a stream of continuous energy with constant RMS, so there are no sudden changes that might trigger different preamplifier behaviour. However, some subjects were still able to identify the difference and also provided interesting comments on their perceived dissimilarity. The biggest deviation from the mean can be seen in the bass sample. This can be explained with reference to the nature of human hearing, which is less sensitive to low frequencies, which do not, or should not, present any difference when they are reproduced by different systems. The cello sample presents the lowest deviation, even though the valve piano sample was subjectively regarded as being the brightest. This could be explained by the fact that the cello sample had just a couple of sustained notes, as opposed to the piano, which had many more notes played in the same period of time, thus preventing the listener from focusing on timbre. Therefore, listeners paid more attention to the character of the cello sample. But even though the deviation is quite large on the piano, taking into consideration the mean result and the subjective opinions provided by the listeners, the result can be regarded as significant and relevant.

The standalone THD+N measurements seen in Table 1 show a difference of approximately 5dB between the valve and the solid-state preamplifier, a variance that cannot alone account for the brightness difference. Moreover, the individual harmonics presented in Table 2 show that the 3rd harmonic, which is usually regarded as non-musical, presents close values (within 1dB difference) for both preamplifiers, whereas the 2nd harmonic is higher for the valve preamplifier and the 4th harmonic is higher for the solid-state. The 5dB difference between the two cases could explain some of the difference in brightness. Figure 10 shows that the valve preamplifier presents a low- and high-frequency roll-off, which, in theory, would be equivalent to a boost in the mid-frequencies. Taking into consideration the results from the frequency response and the individual harmonics tests, it can be stated as a first fact that the reported difference in brightness could be explained by these two factors.

Moreover, as Metzler (1993) specifies, the intermodulation distortion test is more meaningful than a THD analysis as it gives the distortion values for the products, which are not harmonically related to the pure input signal. Table 3 shows that the valve's IMD value is five times that of the solid-state, indicating that the content created by the non-harmonically related frequencies is much higher in the valve preamplifier.

So far, electrical measurements were taken into consideration, especially the THD measurements and it was concluded that one possible cause for the brightness could be the combined effects of frequency response, 2nd order harmonic and IMD. The square wave test is also very important, providing an overview of the expected system's frequency response. Figure 13 shows the systems' responses, and by analysing the two slopes of the orange waveform, it can be predicted that the valve

preamplifier would suffer both low- and high-frequency roll-off, and this is proved by the frequency response graph.

More important in the square wave test are the rise and fall times, which give an indication of the transient response of the system, as stated by Elliot (2015). It can be seen from Figure 11 and Figure 12 that the solid-state exhibits faster rise and fall times, indicating a better transient response that is approximately equivalent to an amplifier's slew rate. The difference in rise and fall times may explain some of the listeners' claims of a difference in 'drive' and 'attack' of the notes. It can be explained by thinking of the higher inner capacitance which produces a higher Miller effect, the final total capacitance being represented by the plate-grid capacitance from the datasheet multiplied by the gain. Combining this theory with the information provided by the attack slopes in Figure 14 and Figure 15 and listeners' comments from Figure 8, it can be concluded that, particularly for the bass, the difference arises from the different transient response of the two systems. Although the bass test shows a high deviation from the mean (see Figure 7), the test is still valid as the majority of participants could describe a difference between the two samples.

The novelty results from Figure 16 and Figure 17 also support comments such as 'more attack', 'faster attack' by showing that there are quite big differences between the two preamplifiers at the same moments of time. This contrast may be due to both the different transient response and the low-frequency roll-off exhibited by the valve preamplifier. Until now, based on the measurements taken, the 'accuracy of reproduction' claimed by five of the listeners who would choose the solid-state preamplifier for all the instruments can be attributed to the better overall frequency response, lower THD, lower IMD and better transient response of this preamplifier, which translates into a much cleaner, neutral sound.

The overall subjective increase in brightness for the valve preamplifier can also be acknowledged through the spectral centroid measurement. Figure 18 and Figure 19 show that in the case of the cello, the coefficient values for the centroid are greatly elevated, indicating that most of the energy at that particular instant of time was in the high-frequency register. The bass sample presents almost identical centroid measurements for the preamplifiers, showing, as reported by the listening tests, that the bass is not brighter.

The transient response can also be estimated by looking at the skewness results (Figure 20 and Figure 21). It can be seen that in both cases, the skewness coefficient values are higher for the solid-state preamplifier, meaning that they are positively skewed with regard to the valve's values. A positive skew means that the system exhibits a higher concentration of energy on the left side of the mean value, as stated by Lartillot (2013). This result relates to the transient response and the rise time, meaning that a positive skew translates to a higher energy in the first fraction of the transient. The reported 'more attack' is further emphasised by the skewness test: Figure 20 clearly shows that the main difference in the skewness coefficient value is on each transient that is represented by the played notes. In the guitar's case (Figure 21), the

two inverse U shapes represent the strings being struck and, as in the case of the bass, an elevated skewness value can be seen. Therefore, the claim of 'more drive' can be proved. In the other samples, an overall accentuated coefficient value for the solid-state preamplifier was identified.

Correlation theory

During the analysis of both the subjective and the objective measurements, different aspects of the systems have been brought into the discussion. It has been seen that although the overall difference in THD+N between the two systems is only around 5dB, when the systems were tested for the evolution of THD over frequency, a much greater difference was identified. Moreover, in order to prove that frequency response does not have a high impact on the difference between the two preamplifiers, the original cello sample (the one used for recording through both preamplifiers) was equalised with a curve like the one in the valve's response. Comparing the resulting equalised sample to the original in MATLAB showed a very slight difference in spectral centroid between the two. The difference alone could not be held responsible for the overall increase in brightness due to frequency response difference.

The increase in THD vs. frequency in the valve preamplifier can be attributed to the low working plate voltage, which is right at the bottom of the valve's capabilities. Basically, the valve is near the edge of cut-off, but still working linearly for line-level amplification. Moreover, the IMD measurement, which indicates the distortion caused by the product of the non-harmonically related frequencies, shows that the valve produces 5.2 times more distortion than the solid-state. The IMD can be caused by the high internal capacitance of the valve compared to the solid-state's inner capacitances. By combining the increased THD, IMD and low- and high-frequency roll-off in the valve preamplifier, it can be concluded that these are the main differences that explain the overall increased brightness in the valve samples. The rise and fall times explain the better transient response in the solid-state preamplifier. The THD also explains one of the listener's comments for the cello valve sample: 'enhanced harmonics'. The reported brightness is also backed up by the spectral centroid tests.

The high inner capacitance of the valve that gives rise to an increased Miller effect and the grid's high-input impedance could be responsible for the overall slow transient response confirmed by the rise and fall times of the valve preamplifier. The combined effect of the two causes gives rise to an increased Resistor-Capacitor (RC) time. This is the reason why the subjects reported the feeling of more drive and more attack for the guitar and bass in the solid-state samples. The increase in attack transient for the solid-state was also demonstrated through the attack slope and skewness tests.

Conclusion

This paper details a comprehensive set of tests to determine whether there are perceptual differences between solid-state and valve preamplifiers. Both preamplifiers work according to the schematic's calculations and in the linear region, ensuring that they can be fully compared. The results indicate that there are clear differences between the two systems, and both the electronic and MATLAB measurements support the subjective findings reported by the listeners. An identification rate of 70%, and in some cases even above 90%, coupled with an extensive number of trials, proves that participants were able to tell the difference between valves and solid-state amplification at low voltages, the reasons being the increase in THD vs. frequency, IMD, frequency and transient response. Participants' preferences regarding the two preamplifiers differed from instrument to instrument. While five people would choose the solid-state for all of the instruments, the other seven had split opinions over the preference for solid-state and valve, depending on the audio material. Six people would choose a preamplifier that brightens the piano, so chose the valve. The rich harmonic content of the cello led eight people to choose the valve preamplifier, as it enhances the harmonics. Conversely, 10 people chose the solid-state for the bass due to the better transient response. The six people who chose the solid-state for the guitar reported that it had more drive, while the other seven felt that the valve enhanced the guitar's body (500Hz – 1.5kHz).

Further work

The work carried out on this topic covers most possibilities extensively, but further work could be carried out in order to find more reasons for the differences between the two preamplifiers. The internal architecture of the valve could be further analysed to find out what differences there are in construction between the solid-state and the valve that could explain the dissimilarity.

According to previous research on this subject, there is a reported difference between the same valve model manufactured by different companies. Another enhancement to the project would be for both the input and output stages to be changed and replaced with matched pairs of transistors, which are usually available as a package (like SSM2012). By having the same brand of matched pairs, it would be possible to confirm that any change arises from the amplification stage. The frequency response can be further tuned in order to match it equally between the two systems and to scientifically prove that it does not make any difference. Moreover, the samples could be further improved by choosing all of them as either having continuously played notes or struck notes, depending on what is to be demonstrated through the listening test.

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