

Output Impedance Shaping for Frequency Compensation of MOS Audio Power Amplifiers

Ronan van der Zee, *Member, IEEE*, and Fred Mostert

Abstract—A frequency compensation technique for MOS audio power amplifiers is presented that allows the frequency compensation capacitors around the power transistors to be smaller than the circuit parasitics without power or stability penalty. Stability is analysed by inspecting the output impedance of the closed loop amplifier, instead of the traditional open-loop gain. By degenerating the gain of the penultimate stage, the output impedance is shaped such that the stability of the audio amplifier is guaranteed for complex loudspeaker loads. The realized amplifier features a THD of 0.005% @ (1 kHz, 10 W), an SNR of 110 dB(A), and stable operation for any passive load up to 50 nF.

Index Terms—Audio amplifiers, frequency compensation, Miller compensation, multistage amplifiers, output impedance, power amplifiers.

I. INTRODUCTION

AUDIO signal processing is increasingly digital, which has led to ever larger dynamic range and tougher distortion requirements. The loudspeaker, however, is still driven by an analog voltage, either by a switching amplifier or a linear amplifier. From an efficiency point of view, switching amplifiers are preferred, but linear amplifiers are still superior in terms of frequency response, integration level and ease of application. But even though linear audio power amplifiers have been around for a long time, the frequency compensation design methodology is not well established.

The design of integrated audio amplifiers has many similarities to general OPAMP design. A low distortion is required, the amplifier needs to be stable for a wide load range, and external stabilization networks -as often used in discrete audio amplifiers- are not acceptable. Frequency compensation schemes for opamps range from traditional Miller compensation to a myriad of alternatives that promise a better trade-off between power consumption, bandwidth and stability [1]–[9]. There are two problems, though, that limit the applicability in the case of audio power amplifiers.

The first issue is that all techniques assume compensation capacitors that are larger than the circuit parasitics [1]–[9]. In audio power amplifiers, however, most of the chip area is occupied by the power transistors, so any compensation capacitor is in fact much smaller than the power transistor's gate-source capacitance. In most cases, the largest compensation capacitor that can be used, is the parasitic gate-drain capacitance (C_{ds})

of the power transistor, which is typically several times smaller than the gate-source capacitance (C_{gs}). Although pole splitting still occurs, the achievable bandwidth is reduced and distortion increases.

Another problem lies in the difficulty to assess the suitability of frequency compensation strategies for complex loads. Because the stability of an amplifier is usually derived from its open-loop frequency transfer, the analysis for all possible loads would become very complex. Consequently, the load is usually assumed to be fixed [1]–[4], partly variable [5]–[7] or capacitive in a certain range [8], [9]. This does not provide a full picture of the behavior of the amplifier for complex loads, which is a necessity for audio amplifiers due to the very wide impedance range of real-life loudspeakers [10].

To solve both these issues, we propose a modification to Nested Miller Compensation (NMC) such that we can use compensation capacitors that are smaller than the power transistor parasitics without sacrificing bandwidth or power. We arrive at this result by using an output impedance shaping technique. Adding to [11], we will present a more thorough discussion of this technique, including the mathematics behind it. We will show how this technique reduces mathematical complexity compared to open-loop analysis, and how it gives insight and information about stability for all complex loads. Also, the amplifier is discussed in more detail.

The outline of the paper is as follows: In Section II, the use of output impedance analysis is motivated. Subsequently, in Section III, the output impedance is shaped to design the frequency compensation topology of a MOS audio power amplifier. Sections IV and V discuss the realization and measurements, with conclusions in Section VI.

II. OUTPUT IMPEDANCE ANALYSIS

A. Stability

For power amplifiers with varying loads, traditional analysis of the open loop frequency transfer becomes extremely complicated. Alternatively, analysis of the closed-loop output impedance offers several advantages. Refer to Fig. 1 for a classic two-stage Miller compensated opamp in unity-gain feedback and its corresponding closed-loop output impedance. Also sketched in Fig. 1 is the impedance of a possible load capacitor C_{load} .

An intuitive approach to assess the stability is to notice that the output impedance is almost purely inductive at the frequency where the load is purely capacitive, forming a resonant tank that causes peaking in the frequency response. Also, neither very small, nor very large C_{load} will cause problems, as $1/sC_{load}$ will intersect a resistive Z_{out} in the far right and far left of the

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R. van der Zee is with the University of Twente, 7500 AE Enschede, The Netherlands (e-mail: ronan.vanderzee@utwente.nl).

F. Mostert is with NXP Semiconductors, Eindhoven, The Netherlands.

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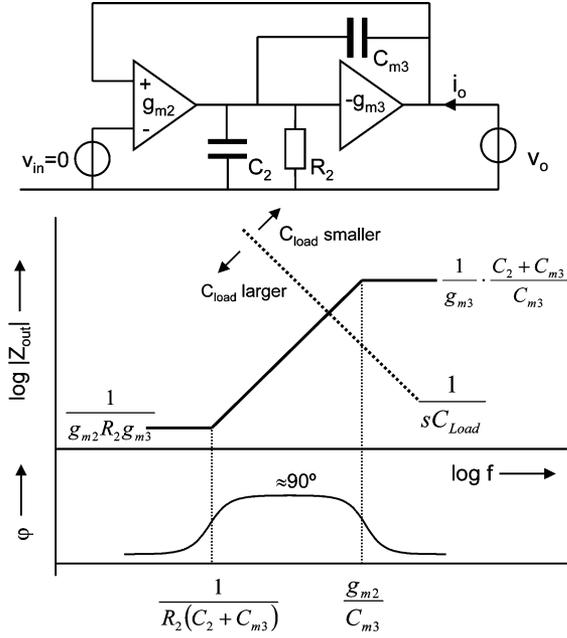


Fig. 1. Output impedance of a 2-stage Miller compensated OPAMP.

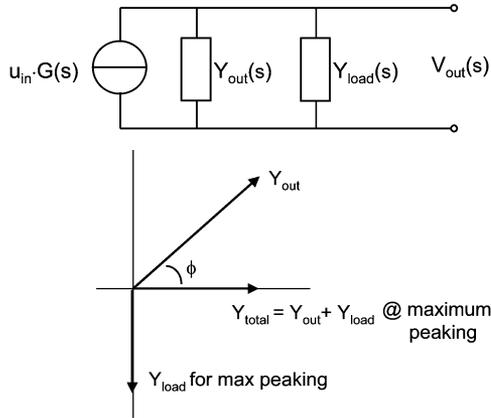


Fig. 2. Norton equivalent of amplifier with output admittance plot.

plot, respectively. This is in line with common knowledge about a Miller-compensated opamp [8], and serves as an illustration of how easy this can be evaluated from Z_{out} .

To address the issue more quantitatively, let us first ease the analysis by representing the amplifier by a current source with parallel admittances, as shown in Fig. 2. The unloaded amplifier frequency response is $G(s)/Y_{out}(s)$. By connecting a load, this changes to $G(s)/(Y_{out}(s) + Y_{load}(s))$. If $|Y_{out} + Y_{load}|$ is smaller than $|Y_{out}|$, peaking occurs. A special case is $Y_{out} + Y_{load} = 0$, where an input signal is no longer necessary to achieve an output signal, commonly referred to as oscillation. This is only possible if $\text{Re}(Y_{out}(s)) \leq 0$ for some frequency. In that case, there is always a passive load $Y_{load} = -Y_{out}$ that can cause the amplifier to oscillate.

For $\text{Re}(Y_{out}(s)) > 0$, the amplifier is stable for passive loads, but peaking in the frequency response is possible. Maximum peaking is obtained by minimizing the total admittance $|Y_{total}| = |Y_{out} + Y_{load}|$. Given the fact that $\text{Re}(Y_{load}) > 0$

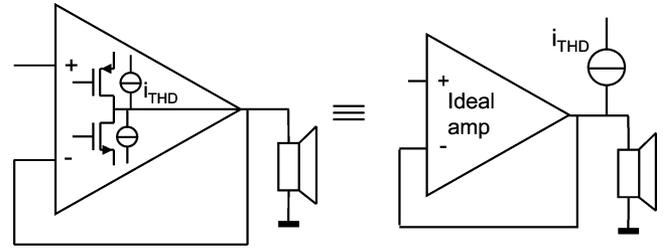


Fig. 3. Distortion modeling.

for any passive load, it quickly becomes clear from Fig. 2 that minimum $Y_{total} = Y_{min} = \text{Re}(Y_{out}) = |Y_{out}| \cos(\arg(Y_{out}))$. Since $\arg(Y_{out}) = -\arg(Z_{out})$ and $\cos(\phi) = \cos(-\phi)$, we can rewrite the maximum peaking compared to no load to

$$\text{Maximum peaking} = 20 \log \left(\frac{1}{\cos(\phi)} \right) \text{ [dB]},$$

$$\text{with } \phi = \arg(Z_{out}) \in \left[-\frac{1}{2}\pi, \frac{1}{2}\pi \right]. \quad (1)$$

B. Distortion

Another property that can directly be derived from Z_{out} is the distortion. Most of the distortion in audio power amplifiers is generated by the strong non-linearity of the common-source class-AB biased power transistors. In quiescent, or for small output currents, they operate in weak inversion. For larger output currents they operate in saturated strong inversion, and at the onset of clipping, when the output voltage is near the supply rails, they operate in the linear region. Since this non-linearity is mostly determined by the output voltage and load, we can model the distortion as a distortion current source parallel to the output transistors. As shown in Fig. 3, we can move this THD source outside the amplifier, and conclude that lower closed loop Z_{out} in the audio band means lower distortion.

III. OUTPUT IMPEDANCE SHAPING

We will now actively use Z_{out} in our design procedure to deal with stability and distortion in relation to the limited compensation capacitors in a MOS power amplifier.

A. Small C_m

To clarify the issue with a limited Miller capacitance C_m , refer to Fig. 1 again. Suppose we decrease C_{m3} compared to Fig. 1. This means that the high frequency output impedance $1/g_{m3} * (1 + C_2/C_{m3})$ increases, as well as the pole g_{m2}/C_{m3} of Z_{out} . Consequently, the inductive behavior of Z_{out} extends to a much higher frequency, meaning that instability is reached for smaller load capacitances than originally. To avoid this unwanted effect, we have two options. The first is to increase g_{m3} , but this comes at the expense of power consumption. The second option is to keep g_{m2}/C_{m3} at the same frequency. This means the whole Z_{out} curve shifts up, which leads to a higher distortion. We would like to find a better solution to this trade-off between C_m , power, distortion and stability.

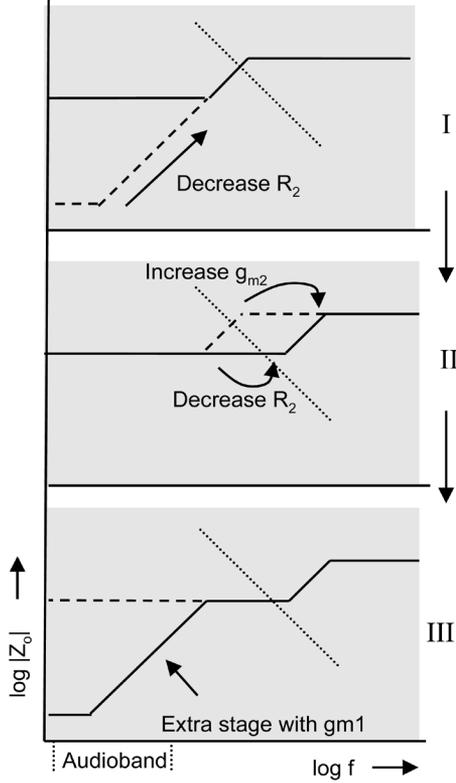


Fig. 4. Shaping the output impedance.

B. Gain Degeneration

Our solution starts by referring to Fig. 1 again. We decrease the value of R_2 such that the zero in the Z_{out} plot moves higher, and LF Z_{out} increases. When the zero and the pole of Z_{out} are close enough together, the phase change in the frequency range in between is limited, also limiting the maximum peaking. However, the decrease of R_2 has also resulted in a limited LF gain and thus high distortion. This step is shown in step I of Fig. 4. The low LF gain is overcome in [8] by using a very large C_m . We propose to still use our small C_m , and decrease R_2 even further while at the same time increasing g_{m2} (step II in Fig. 4). Finally, we add an extra stage g_{m1} in front (step III in Fig. 4). Fig. 5 displays the resulting topology. We will discuss the implications of these steps after the following mathematical analysis.

To calculate Z_{out} in Fig. 5, we neglect the direct contribution of C_{m3} and C_{m2} to the output impedance. Since they are small, this is reasonable to assume. Further below we will show that this assumption does not lead to large deviations between calculations and measurements. This yields (2), shown at the bottom of the page.

To get a simple expression for the poles, we assume that $g_{m1}R_1 \gg 1$ and $C_{m2} \gg C_1$. The first assumption says that the DC-gain of the first stage is considerably larger than 1, which

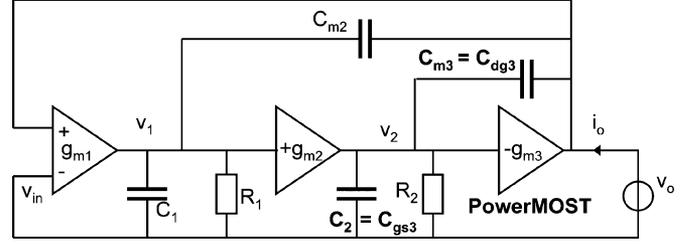


Fig. 5. NMC with low R_2 .

will usually be the case. The second assumption is also easy to satisfy, since C_1 , being the input capacitance of the second stage, is small compared to the much larger capacitances of the final stage. The resulting pole and zero locations of the output impedance are then as follows:

$$\begin{aligned} z_1 &= -\frac{1}{R_1(C_1 + C_{m2})} & z_2 &= -\frac{1}{R_2(C_2 + C_{m3})} \\ p_1 &= -\frac{g_{m2}}{2C_{m3}} \left(1 + \sqrt{1 - 4\frac{g_{m1}C_{m3}}{g_{m2}C_{m2}}} \right) \\ p_2 &= -\frac{g_{m2}}{2C_{m3}} \left(1 - \sqrt{1 - 4\frac{g_{m1}C_{m3}}{g_{m2}C_{m2}}} \right). \end{aligned} \quad (3)$$

The crucial aspect of our solution, as discussed above, was to decrease R_2 while increasing g_{m2} . By increasing g_{m2} such that $g_{m2}/C_{m3} \gg g_{m1}/C_{m2}$, in other words, making the unity-gain frequency (UGF) of the inner loop much larger than the UGF of the outer loop, the pole locations can be approximated with a Taylor expansion and become

$$p_1 = -\frac{g_{m2}}{C_{m3}} \quad p_2 = -\frac{g_{m1}}{C_{m2}}.$$

The resulting output impedance is shown in Fig. 6. As a reference, the output impedance of the original two-stage OPAMP with only g_{m2} and g_{m3} is plotted as a dashed line.

To assess the result, let us first consider a very small capacitive load C_{L1} in Fig. 6. It intersects an almost real output impedance in region I, so for a small load capacitance the amplifier is stable. When C_L is increased, it will cross the output impedance in an inductive part (II). This constitutes a resonant circuit at that frequency, leading to peaking in the frequency response. The amount of peaking, according to (1), is determined by the maximum phase of Z_{out} in region II, which is determined by the ratio between the zero and pole that form the borders of region II. This distance is equal to a factor $g_{m2}R_2(C_2 + C_{m3})/C_{m3}$, and since $C_{m3} = C_{dg3} = 20$ pF and $C_2 = C_{gs3} = 80$ pF are determined by the power transistor, we have $g_{m2}R_2$ as design freedom. We chose $g_{m2}R_2 = 1$, since this is easy to realize as a source follower in the final schematic. It leads to a pole/zero

$$Z_{out} = \frac{v_o}{i_o} = \frac{(1 + sR_1(C_1 + C_{m2}))(1 + sR_2(C_2 + C_{m3}))}{g_{m2}g_{m3}R_1R_2(g_{m1} + sC_{m2}) + g_{m3}R_2sC_{m3}(1 + sR_1(C_1 + C_{m2}))} \quad (2)$$

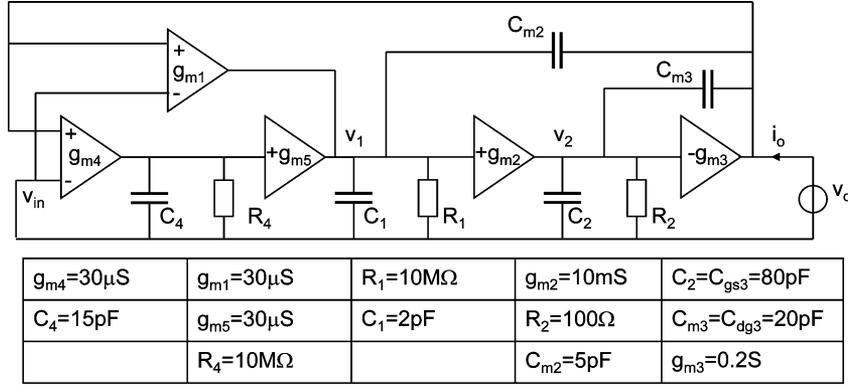
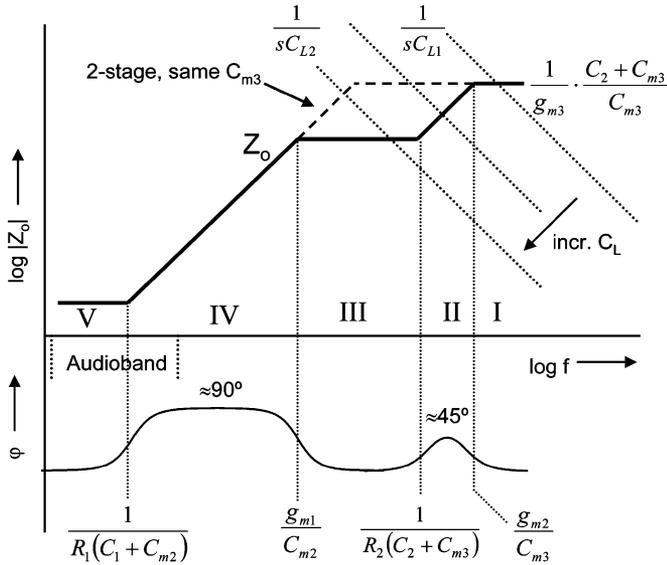

 Fig. 7. Frequency compensation model after adding $g_{m4,5}$.


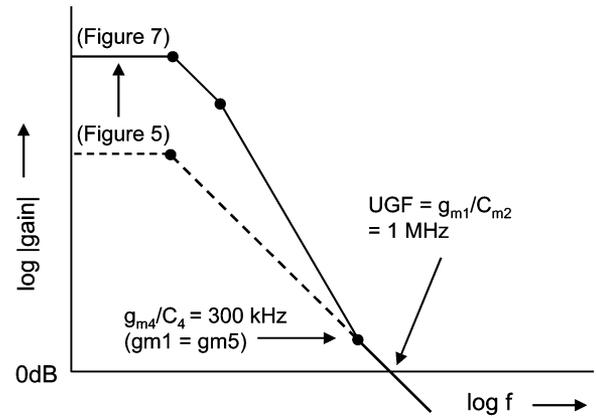
Fig. 6. Output impedance of Fig. 5.

ratio of 5, which gives a maximum phase of Z_o of 45° , corresponding to a maximum peaking of 3 dB according to (1).

Increasing C_L further, the system is more stable again (region III), and only for larger C_L (region IV) stability is compromised. Note that the stability of a two-stage Miller compensated opamp would already be compromised for C_L larger than C_{L2} in Fig. 6, which is the same factor of 5 lower than in our new design. Thus, we see that the load limitations imposed by a limited Miller capacitance have been overcome.

Two remarks are in place at this point. First of all, we have used a capacitive load in the reasoning above, while real loudspeakers are RLC loads. We have done this because the maximum amount of peaking for any load is described by (1), but it does not specify for which load it occurs. As illustrated in Fig. 6, the output impedance of our circuit has a positive phase, meaning that only loads with a capacitive component will cause problems, so we used C_L as worst case.

A second remark relates to R_2 . Because R_2 is small, one might be tempted to look at this structure as simply driving the gates of the power transistors with a low-impedance source, a kind of resistive broadbanding. It is not that simple, however, because a smaller C_{m3} would then be favorable for stability,


 Fig. 8. Increase of open loop gain by adding $g_{m4,5}$.

as it limits the capacitive load seen by g_{m2} . Our analysis, however, shows that a smaller C_{m3} will actually decrease the phase margin for capacitive loads in region II in Fig. 6.

C. Extra Gain Stage

Although one could accept the topology of Fig. 5, or even degenerate the first stage gain too, to achieve stability for all capacitive loads, this is not an option here since the amplifier still has too much distortion for our purpose. An extra gain path is added in parallel to g_{m1} , consisting of g_{m4} and g_{m5} as shown in Fig. 7. This extra path is dimensioned such that it adds gain (and phase shift) only below the UGF of the open loop transfer, as shown in Fig. 8. The extra gain lowers Z_{out} in the audio band further (Fig. 9), reducing distortion. Fig. 10 shows the more classical open loop frequency response for various loads. In this case, of course, we can only analyse a limited number of loads. For loads of 4Ω or $10\text{ k}\Omega$ in parallel to 10 pF or 10 nF , the phase margin stays above 65° , and for values in between the picture was similar.

D. Driving the Load

Up to this point, we have assumed a constant g_{m3} , whereas in reality g_{m3} varies strongly depending on the load current and output voltage. The calculations are done for the class AB quiescence current, where g_{m3} is very low. When the power transistors carry a larger current, expression (2) shows that the Z_{out} curve in Fig. 9 simply shifts down, only improving the stability

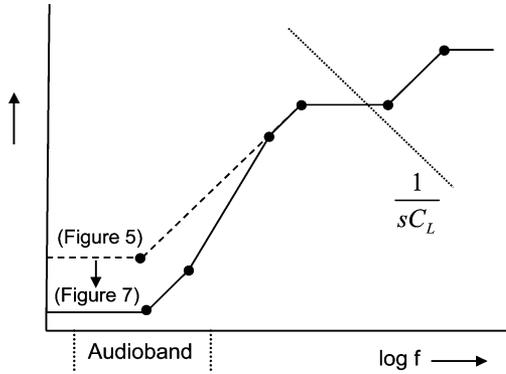
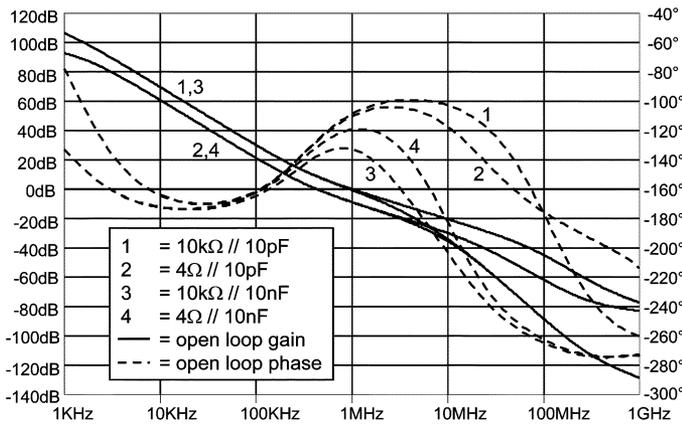
Fig. 9. Decrease of Z_{out} after adding $g_{m4,5}$.

Fig. 10. Open loop response of Fig. 7 for various loads.

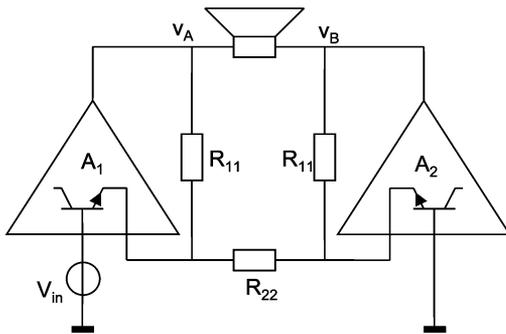


Fig. 11. Current feedback bridge configuration.

of the amplifier. Note that the analysis holds for all possible load impedances. We didn't need to assume $g_{m3}R_L \gg 1$, as is usually the case.

IV. REALIZATION

The amplifier was realized in the NXP ABCD2 process, an SOI Bipolar-CMOS-DMOS process with $1 \mu\text{m}$ feature size. The chip is targeted as a quad channel audio amplifier for automotive applications. The four channels drive the loudspeakers in BTL mode, as shown in Fig. 11, and each bridge half has the frequency compensation setup as described above. If the bridge halves would have been equipped with standard differential pair inputs, the common-mode feedback factor would have been unity, considerably larger than the differential feedback

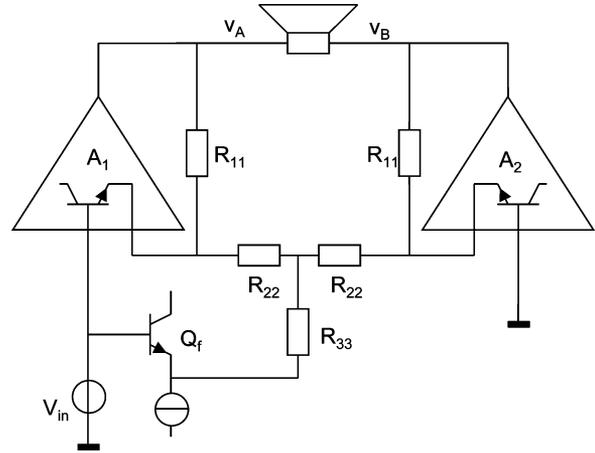


Fig. 12. Topology of one channel.

factor it is designed for, giving rise to stability problems. Using current-mode inputs ensures that the common-mode stability is the same as the differential-mode stability. As a consequence, g_{m1} and g_{m4} in Fig. 7 are set by the feedback resistors R_{11} . Unfortunately, the configuration of Fig. 11 suffers from unequal gains to the two bridge halves: $v_A/v_{in} = 1 + R_{11}/R_{22}$ and $v_B/v_{in} = R_{11}/R_{22}$ respectively, which causes asymmetrical clipping and consequently reduced low distortion output power. Simply changing one of the R_{11} 's is not an option because it would change stability. The solution is shown in Fig. 12, where we apply an extra signal to the "middle" of R_{22} . The gains of the two bridge halves now become

$$\begin{aligned} \frac{v_A}{v_{in}} &= 1 + \frac{R_{11}}{R_{22}} \left(\frac{R_{33}}{R_{22} + 2R_{33}} \right) \\ \text{and } \frac{v_B}{v_{in}} &= -\frac{R_{11}}{R_{22}} \left(\frac{R_{22} + R_{33}}{R_{22} + 2R_{33}} \right). \end{aligned} \quad (4)$$

For symmetrical clipping, the absolute value of the gains should be equal. From (4) it follows how R_{33} must be chosen to achieve this: $R_{33} = (R_{11} - R_{22})/2$. It can also be shown (although this is easy to see because of the symmetry) that the extra path has the same effect on v_A and v_B . Therefore, any distortion caused by Q_f , which works open-loop and has to drive R_{33} , only results in a common-mode term, not affecting the differential output voltage.

Fig. 13 shows a simplified circuit schematic of one bridge half, where component numbering corresponds to Fig. 7. As mentioned above, the gain $g_{m2}R_2$ is chosen equal to 1 and realized by a source follower which behaves like $g_{m2}R_2 \approx 1$ in the frequency range of interest. To achieve a high value of g_{m2} , I_2 must be large, but we need a large I_2 anyway because of the high charge- and discharge currents of the gate of M_3 during crossover and clipping. Class AB control is similar to [12]. The addition of M_2 to drive M_3 , however, would reduce the minimum supply voltage to $3V_{GS} + V_{DS,sat}$. We included M_{AB2} to reduce this value to $2V_{GS} + 2V_{DS,sat}$. Integration of M_{AB2} in the cross-coupled bias circuit instead of realizing it as a separate level shift reduces quiescent current spread. The quiescent current I_Q is set by the translinear loop of M_3 , M_2 , M_{AB2} , M_{AB1} and two stacked V_{gs} (not shown) that generate V_{bias2} . I_Q of M_3

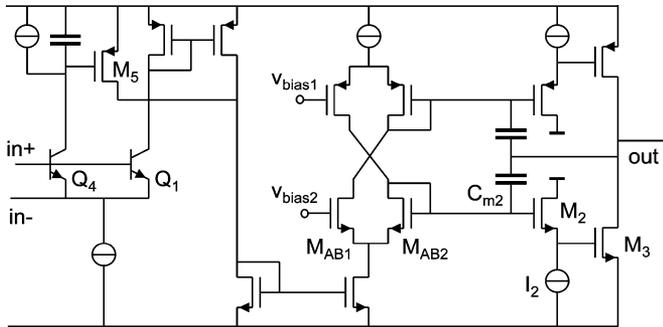
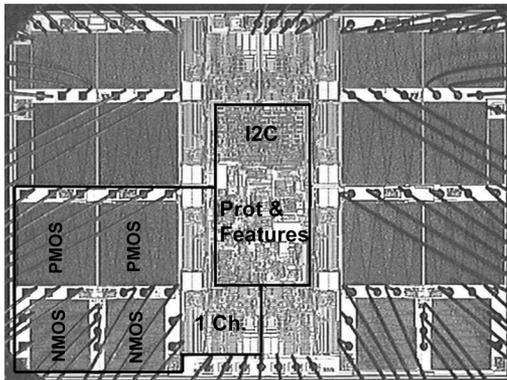
Fig. 13. Simplified schematic of one bridge half (A_1 in Fig. 12).

Fig. 14. Chip photo.

is 20 mA. This is close to weak inversion for this power transistor, and it brings the ratio between quiescent and maximum current close to 200.

The chip photo is shown in Fig. 14. A major selling point for automotive amplifiers is the output power of a saturated square wave output signal. The resistance of the leadframe, bondwires, on-chip metal and transistor R_{on} all deteriorate this value. Large power transistors, three bondwires per pin and a slew-rate much larger than needed for audio help to reach the measured value of 46 W at 14.4 V supply and a 4 Ω load. Other features include a low-gain line driver mode, no-plop startup, 10 μ A standby current, soft mute, load detection, overtemperature protection that gradually decreases the gain with rising temperature, and several other protection features, all accessible by an I2C interface.

V. MEASUREMENTS

Fig. 15 shows the measured Z_{out} of the packaged product together with the simulated Z_{out} of the small signal schematic in Fig. 7. There is a good match, and the stability of the amplifier can be assessed from this figure. First of all, the phase of Z_{out} does not go below 0° phase shift, meaning that inductive loads can not cause any peaking. Capacitive loads are more dangerous. The phase of Z_{out} stays below 45° above 1 MHz. At 1 MHz, $|Z_{out}| \approx 3 \Omega$, so the maximum capacitive load for 3 dB peaking is 50 nF. For larger C_{load} , stability quickly deteriorates. Indeed, in extensive measurements with a large number of different complex loads, the amplifier remained stable for any passive load with a capacitive component less than 50 nF, without the use of any external stabilizing network.

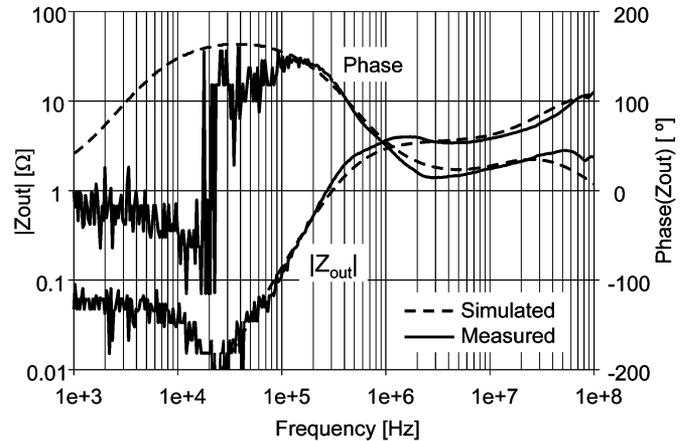


Fig. 15. Measured and simulated (Fig. 7) output impedance.

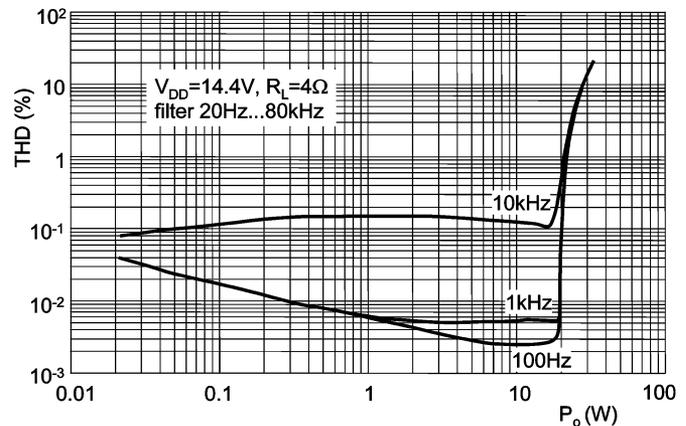


Fig. 16. Measured distortion.

The sole exception is a very small ($< 100 \text{ m}\Omega$) resistive load, because the phase of Z_{out} approaches 180° for low frequencies, as it would in any amplifier with a second-order open-loop gain. This also causes the notch in Z_{out} at 20 kHz. The lead frame plus bondwire resistance is positive, and Z_o is almost purely negative at this frequency. In practice, this means that the amplifier exhibits less stable behavior when it is in a near-short-circuit situation. Any actual instability, however, immediately leads to a large output current because of the low ohmic load, moving the Z_{out} curve down, increasing stability. In practice, we did not experience any problems with this phenomenon.

Fig. 16 shows the distortion performance. Typical THD + N is 0.005% @ 1 kHz (10 W, filter 20 Hz–80 kHz), SNR is 110 dB (A).

VI. CONCLUSION

Output impedance shaping is a technique that allows easy understanding and manipulation of the stability of amplifiers for complex loads. By degenerating the gain of the penultimate stage in a power amplifier, a frequency compensation technique is created that does not need compensation capacitors that are larger than the power transistor parasitics. This is demonstrated by the area-efficient realization of a MOS audio power amplifier that has low THD and good stability over a wide load range.

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Ronan van der Zee (M'07) was born in Hengelo, The Netherlands, in 1970. In 1994, he received the M.Sc. degree (*cum laude*) in electrical engineering from the University of Twente, Enschede, The Netherlands. In 1999, he received the Ph.D. degree from the same university on the subject of high-efficiency audio amplifiers.

In 1999, he joined Philips Semiconductors, where he worked on class AB and class D amplifiers. Since 2003, he has been an Assistant Professor in the IC Design group at the University of Twente.



Fred Mostert was born in 1959. He joined Philips Semiconductors in 1982 after receiving the Bachelor degree in electronics. As a System Architect at NXP Semiconductors, he is responsible for audio power amplifiers in car radios.