

El-Cheapo - A Really Simple Power Amplifier

Rod Elliott - ESP (Semi-Original Design)

"Semi-Original Design" - What is that supposed to mean? Well, many years ago, there was an amplifier circuit in a magazine (I don't remember which one, but I think it was a US edition). This amp was called "El-Cheapo", and used a single power supply and capacitor coupled speaker. I do not recall the exact circuit details well, but it was a very simple amp, and used quasi-complementary symmetry for the output stage. Note the really sneaky way the Class-A driver amp's collector load is bootstrapped !

For those younger than I who have no idea what I'm talking about, quasi-complementary symmetry was a scheme used in the days when PNP power transistors were expensive and useless. If you wanted any sort of voltage and current rating, you had to use NPN devices. The quasi-complementary output stage used a (discrete) darlington for the positive side, and a complementary pair for the negative (i.e. a PNP driver coupled to an NPN power transistor).

Figure 1 shows the circuit as I remember it (with component values re-calculated - I have no idea what they used to be, but those shown should be pretty close), and it **was** a cheap amp compared to most offerings of the day. It also managed to sound respectable - again by comparison - and I and many of my friends of the day built these amps with abandon - guitar amps, hi-fi, you name it, El-Cheapo was in there!

Note that the transistor types are "modern" equivalents - I cannot remember what the originals were, but they are almost certainly obsolete. It is also likely upon reflection that R6 was probably closer to 4k7, since the speaker provides the DC voltage return for the bootstrap circuit, and is only 6 Ohms or so DC resistance.

Another change is the speaker coupling capacitor - I do remember that it was 1000uF (for a -3dB of 20Hz and a 8 Ohm load). This is too small, and the 2200uF shown is actually marginal. 4700uF would be better, or even more - but that would defeat the purpose, since the amp would no longer be cheap. Besides that, it still has a single supply, and such amps are not well considered by anyone these days.

A useful increase in gain may be achieved by increasing the current through Q1, by reducing the value of R4. There is a problem with this however, since the voltage across R5 becomes excessive, raising the input DC voltage on the base of Q1. One can reduce the value of R5 (the feedback resistor) but then the required capacitance of C4 becomes too high to be sensible because R12 must be reduced for the same audio gain.

Further Improvements

Figure 3 shows all the additional improvements possible while still retaining the input stage, and a simulation indicates that the open-loop gain of this configuration is over 150dB (or 30 Million) open loop - this is likely to be somewhat optimistic, but is a good indicator of the available gain one can achieve without the current mirrors and other accoutrements generally found in typical input circuits. With a gain as high as this, there is enough feedback for anyone - without getting more complex.

The input capacitor has been changed to a polyester (or similar) and with 1uF has a lower -3dB frequency of 7Hz. This may be made larger if your speakers can go lower than that. One thing you cannot do with this input stage is direct couple from a preamp. The voltage on the base of Q1 will be about 1.3V for 0 Volts at the speaker output. If the input were grounded, then there will be -1.3V across the speakers - this is generally considered to be a bad idea. It is only 200mW for an 8 Ohm load, but it should be avoided.

Speaking of feedback - because the input stage creates an inherently stable amp, there is no reason to expect that TIM (Transient Intermodulation Distortion) will be a problem, since feedback is simply applied to the emitter of the input amp, and little or no frequency "compensation" is needed. This is an area where some experimentation is needed, and it might be necessary to connect a low value (47pF ?) capacitor between collector and base of Q2 - it was not needed in the original, but this configuration has vastly more gain.

Construction Hints

Please: bear in mind that these are all theoretical circuits - the designs are sound (pun intended) and have been simulated, but they have not been built at this stage. I have no reason to suspect that the designs as shown will not work perfectly - or as perfectly as they will work (que?) - but I would not be happy without providing this warning.

Construction of any of these variations is non-critical, within the normal bounds of amplifier building at least, and will not be discussed in any detail. I will, however, make the following observations.

I would recommend that Figure 1 be avoided. Use of an electrolytic capacitor in the speaker output is not a good thing, and measurements made by [Doug Self](#) (and others) show low frequency distortion is generated by electros (although the actual mechanism that creates the distortion is unclear).

Naturally, this circuit (*absolutely*) cannot be DC coupled - but I know for a fact that I cannot hear DC, my speakers will not reproduce it, DC will not be recorded and no musical instrument creates it - so why should I (or anyone else) bother?

The bias servo (Fig 2, Q3 or Fig 3, Q4) is designed to allow enough adjustment of the voltage between the bases of the driver transistors to allow accurate bias setting - this transistor should not be mounted on the heatsink, unless the drivers are also mounted there (which I do not recommend!). Quiescent (no-load) current should be about 100mA, measured across the 0.1 Ohm emitter resistors - this will give a reading of 10mV on a multimeter.

The trimpot VR2 is used to set the DC voltage at the output to 0 Volts (+/- 50mV). This should be set finally after the amp has had time to stabilise, which will require at least 30 minutes of operation.

Make sure that there is sufficient heatsinking for the power transistors to avoid excessive temperature rise. I tend to prefer a heatsink which is too large rather than the other way 'round, and anything better than about 1 degree C / Watt should be good - if a little on the large and expensive side. This will be the same for any amplifier you build, regardless of complexity for a given output power.

With this amp (or any amp of similar power) quiescent power is less than 10W (based on a current of 100mA, and given that the power supply voltage will be higher than the nominal 35V), and at worst case dissipation will reach a maximum of about 75 Watts. It is uncommon - but possible - for amps to run at their worst case dissipation during normal use, but it should be accounted for. With a heatsink of 1°C/W, this means that the transistors may reach a temperature of 100°C or more, which will reduce their life expectancy considerably. With heatsinks, size **does** matter.

Is It All Worth It?

The big question (which I cannot answer at the time of writing) is - does this input stage sound better, worse or the same as the more complex versions? The financial considerations are negligible, since we are only talking about a few 50 Cent transistors and some even cheaper resistors, but if the final outcome is that this configuration sounds the same (or even better), then there seems to be no point in making input stages more complex.

Further, I am yet to be convinced that a power amp with dual (or even triple) long tailed pairs, cascode mirror image Class-A drivers, abundant (rampant?) current mirrors, fully DC coupled and 37 compensation and bypass capacitors scattered throughout the circuit sound any better than the amp described in Project 03 in my project pages. I might be wrong (it happened once), but I do believe that a good simple design is just as capable, and may even sound better than one which has been over-designed to such an extent as to be (to me, anyway) completely over the top.

If it doesn't (sound better, that is), then one must ask if the improvement is worth all the extra effort (and cost). There has to be a limit somewhere, and many of us cannot justify 20 output transistors each in 4 monoblock 100W Class-A systems (bi-amped, naturally) when the speakers, room acoustics and recording techniques (plus the demands or restrictions imposed by s/he who must be obeyed) simply do not come even close to the standard of a passably decent power amplifier. Besides, who wants a 2kW heater in the listening room in the middle of summer anyway.

The point of all of this is that I do not believe that the perceived differences in amplifiers is as great as the imagination of the listener. There is a new article in progress which discusses these phenomena, and I hope to have it on-line soon - look out for it, because I might change some thinking on a topic or three, or maybe just start some more discussion.