



A HiFi Power Amp



The recent acquisition of a pair of vintage Infinity RS-5b speakers prompted a search for an amplifier. According to the documentation that came with my speakers, an amplifier between 35W and 135W is recommended (not my 10W at 10% THD piece of garbage Sharp 3-in-1). Initially, I looked at commercial amplifiers (Yamaha, NAD, and Rotel), but was disappointed at their fairly pedestrian distortion figures. A hobby project was born :)

I came up with some nominal specifications:

- 1 Stereo. Whilst I'll use it for watching movies, my flat just doesn't have the room for a full set of sound speakers. My main motivation is listening to stereo music sources, so a stereo amplifier.
- 1 Low THD and IMD. In audio-speak, this translates either as clinical or as accurate. I'm an engineer, so I prefer terms like THD over "warm" or "cold" which could mean anything. Given that most commercial amps offer 0.02% THD or thereabouts, I figured a good target was 0.001%. This means that harmonics and intermodulation will be 100 dB below the fundamental, and will thus mean that the system performance will be dictated by the source (CD) material, rather than the amplifier.
- 1 Ample power. 100W seems a reasonable amount. I currently live in a two bedroom flat, and don't want to annoy my neighbours to kill me :) However seriously, I figured 100W gives me a clear 20dB headroom at a listening level of 1W.
- 1 Low noise. I usually listen to my music at reasonably low level, so it's important for the amplifier to have a low noise contribution.
- 1 Moderate cost. I'm happy to blow around \$1000, as long as I get plenty of enjoyable hobby hours.

listening hours) as payback.

- ┆ Good looks. This thing will (along with my preamp) live in my loungeroom. That means it needn't want something that looks like an escapee from my shed, so a significant part of the design is with building a nice case.
- ┆ Useability. This pertains more to the [preamp](#), but I wanted the whole thing to be completely reconfigurable and controllable.

Design Background

When I was a kid, one of my favorite monthly reads was ETI magazine. In January 1981, they published articles describing the ETI477 MOSFET power amplifier, designed by David Tilbrook. This monoblock was the basis of the "series 5000" HiFi amp. I desperately wanted to build one, but being all of 9 at the time, it never happened.

The neat thing about the series 5000 is that it was built around new (at the time) Hitachi lateral power MOSFETs. Most power MOSFETs (VMOS, trenchFETs, HexFETs etc) use a vertical structure, where the current flows vertically. This has the advantage of stunningly low R_{ds} and hence high efficiency, but does nothing to reduce gate capacitance. Lateral MOSFETs are a much simpler structure, where the gate oxide is formed on a flat surface and the current flows across the substrate. This results in well defined, controllable device parameters, good linearity and relatively low gate capacitance. However, the R_{ds} of lateral MOSFETs is nothing to write home about.

Most amplifiers at the time (and now as well) used bipolar output drivers. Bipolar transistors are cheap and plentiful. They have relatively high transconductance, and can operate reasonably fast. However they have drawbacks when used at high power. The main one is thermal runaway. The gain of a bipolar transistor increases as it gets hotter. That means that if there's any imbalance between output transistors, the hottest one will draw more current, getting hotter until it ultimately expires. MOSFETs don't have this problem. Their gain decreases as temperature increases, so they share the load well.

MOSFETs also have a high input impedance at low frequencies, and are capable (when driven by a source with a low output impedance) of extremely high slew-rates. Of course this very attribute makes them rather prone to HF oscillation, but with careful design they're capable of impressive intermodulation performance.

So having decided that now was the time to build a MOSFET amp, I wandered into the library at work to look up the old ETI series 5000 amp articles, and had a read. I subsequently found that the series 5000 wasn't the final word on MOSFET amps. In 1987 he revisited the topic for a new magazine, Australian Electronics. This time (with the AEM6000 amplifier) he went all-out, with a matched-JFET differential input stage, a complementary symmetrical voltage amplifier stage. A quest for super low distortion figures, with the maximum available gain. This looks like a good place to start.

Optimisation

There were a couple of drawbacks with the design. Firstly, it was based on obsolete TO-3 packaged MOSFETs, and secondly the PCBs (like many kitset boards) were a pretty poor design anyway. I set about redesigning the PCBs around modern flatpack equivalents (Hitachi 2SK1058 and 2SJ162). While I was at it, I swapped many of the remaining transistors for modern (faster) equivalents.

I made the following active device substitutions:

- ┆ JFET input diffamp: SST404 (SO-8). Was ECG461.
- ┆ Low power bipolars: MMBTA06/56 (SOT-23). Were BC547/557 and BC639/640.
- ┆ Medium power bipolars: MJE340/350 (TO-126). Unchanged.
- ┆ Power MOSFET drivers: 2SK1058/2SJ162 (TO-3P). Were 2SK176/2SJ56 (AEM6000) or 2SK237/2SJ276 (AEM6005 and ETI5000).

In order to dissipate 100W in an 8 Ohm speaker, one needs to put 28V RMS across the load. That's 40V peak (assuming a resistive load) the amplifier needs supply 5A. Doing the SOA sums (more later),

pairs of drivers are needed. Further, the V_{gs} for the MOSFET can be around 10V at high current. This supply must be at least 10V greater than the peak output voltage. A twin 40V transformer is appropriate peak secondary voltage of $\pm 56V$.

Now that I'd changed the transistors, I had to play with the values of most of the other components as to get reasonable performance while maintaining stability. Firstly, I decided that rather than the usual power input, I'd increase this a bit, to 1.8V RMS. This allows me to use more of the available dynamic preamp, and requires a gain of 16, or 24dB.

Transistors (and valves) are inherently non-linear devices. They must be linearised, or else they'll distort. There are three ways to achieve this goal:

- 1 Use the transistor over a very small operating range.
- 1 Use feedforward to cancel distortion (symmetry).
- 1 Use feedback to cancel distortion.

Pretty much all amplifiers use a combination of the three. Feedback has a bad name amongst "audiophiles". Poorly thought out feedback (especially across multiple stages) can result in oscillation (usually at very high frequency, which isn't audible in itself, but destroys the performance of the amplifier. Feedback needs to be global though. A robust scheme involves linearising each stage of an amplifier independently (for example, emitter degeneration), then using overall feedback (with appropriate compensation) to set the gain.

I used Linear-Tech's free [spice simulator](#) to redesign the circuit around the newer parts. My main change was to increase the emitter degeneration in each stage, to improve the linearity of each stage, at the expense of overall open-loop gain. This is an approach that makes for an easily stabilised amplifier.

A somewhat simplified schematic is shown below. Yes, it's a wonderfully complex beast of an amplifier with good symmetry, and plenty of stages, for ample open-loop gain. The schematic doesn't show the AEM6000 take on Tilbrook's design. The topology is the same, but the component values are different. For the AEM6000, you'll have to visit the library. Click on the schematic shown for a .pdf version of the real thing including power supply decoupling and gate protection zeners etc.:

