

makefile	
<pre> # makefile for OFDM software # using Microsoft Fortran v. 4.0 compiler # EQ.Casas COMP = fl /Od /c /4yd # disable optimize, compile only ADD = /4Yb /Ge # strict declarations LINK = link # /4Yb for runtime error trace, LINKOPT = /E ; # /Ge for stack check D = ..\bin # link command MOVE = (copy \$*.exe \$D ; \ # EXEPACK option del \$*.exe) # binary destination directory .SUFFIXES: .exe .obj .f .asm # install in binary directory .f.obj : \${COMP} \${ADD} /Tf \$< .asm.obj : masm \$< ; all : \$D\fdint.exe \$D\simrun.exe \ \$D\pint.exe \$D\bkp.exe \$D\ceval.c clean : del *.obj del *.fx # select code for for little- or big-endian CPUs simfde.obj : simfde.f sed -f sim2lbn.sed <simfde.f >simfde.fx \${COMP} \${ADD} /4Ns /Tf simfde.fx # IEEE DSP routines don't declare variables and subscripts are wrong siftt.obj : siftt.f \${COMP} \${ADD} /4Ns /4Nb /4Nd /Tf siftt.f pint.obj : pint.f simpdef.f simpget.f \$D\pint.exe : pint.obj simut.obj simsnr.obj \${LINK} \${MOVE} </pre>	<pre> \$D\fdint.exe : fdint.obj simsnr.obj simfde.obj \ simdiv.obj simut.obj simgen.obj \${LINK} fdint.obj simsnr.obj simfde.obj \ simdiv simut simgen \${LINKOPT} \${MOVE} simrun.obj : simrun.f simpdef.f simpget.f io.obj : \hw\pcio\io.asm masm \hw\pcio\io.asm ; \$D\simrun.exe : simrun.obj simmod.obj simfec.obj \ simfde.obj simgen.obj siftt.obj simsnr.obj simut.obj \ simhw.obj simdiv.obj io.obj \${LINK} simrun.obj simmod.obj simfec.obj \ simfde.obj simgen.obj siftt.obj simsnr.obj simut.obj \ simhw.obj simdiv.obj io.obj \${LINKOPT} \${MOVE} \$D\bkp.exe : bkp.c tc bkp.c \${MOVE} \$D\ceval.exe : ceval.c tc eval.c \${MOVE} </pre>
	<div style="border: 1px solid black; padding: 5px; display: inline-block;"> makefile (unix) </div>
<pre> FC = f77 FFLAGS = -f68881 -C -u -C LD = f77 -f68881 LDOPT = -o \$@ CVT = sed -f sim2unix.sed OBJ = fdint pint simrun bkp ceval BINDIR = ..\bin all : \${OBJ} install : mv \${OBJ} ../bin siftt.o : siftt.f \${FC} -f68881 -c siftt.f simfde.f : simfde.f \${CVT} simfde.f >simfde.f </pre>	

<pre> fdintX.f : fdint.f simpdef.f simpget.f \${CVT} fdint.f >fdintX.f fdint : fdintX.o simsnr.o simfdex.o \ simdiv.o simut.o singen.o \${LD} fdintX.o simsnr.o simfdex.o \ simdiv.o simut.o singen.o \${LDOPT} pintX.f : pint.f simpdef.f simpget.f \${CVT} pint.f >pintX.f pint : pintX.o simsnr.o simut.o \${LD} pintX.o simsnr.o simut.o \${LDOPT} simrunX.f : simrun.f simpdef.f simpget.f \${CVT} simrun.f >simrunX.f simrun : simrunX.o simmod.o simfec.o \ simfdex.o singen.o sifft.o simsnr.o simut.o \ simdum.o simdiv.o \${LD} simrunX.o simmod.o simfec.o \ simfdex.o singen.o sifft.o simsnr.o simut.o \ simdum.o simdiv.o \${LDOPT} pdf : pdf.o simut.o \${LD} pdf.o simut.o \${LDOPT} bkp : bkp.c gcc bkp.c -lm -o bkp ceval : ceval.c gcc ceval.c -lm -o ceval </pre>	<pre> simrun.f \$LARGE c ***** c simrun.f - OFDM Simulation c ***** c Ed.Casas c c link with the following files : c c sifft.f - FFTs (adapted from IEEE DSP library) c simdiv.f - switching diversity c simdum.f - dummy hardware interface (non-IBM PC) c simfde.f - fading envelope generation c simfec.f - FEC and bit/block error counting c singen.f - noise and data generators c simhw.f - hardware interface (IBM PC) c simmod.f - OFDM (de)modulation c simsnr.f - IF to AF signal/noise level conversion c simut.f - various utilities c c includes the following files: c c simpdef.f - declarations of simulation parameters c simpget.f - input of simulation parameters c c Major Revisions: c c 85- VAX/VMS FORTRAN and FPS AP-100 routines c 86- VAX/VMS FORTRAN c 87-8 FORTRAN 77 (MicroSoft v3.3 compiler subset) c program simrun c ----- \$INCLUDE: 'simpdef.f' c ----- c local variables: c c i - trial counter c j - simulation block counter c k - signal block size counter c l - snr counter c n - signal block size c m - index of OFDM block in simulation block </pre>
<pre> sim2ibm.sed /^cIBM/s/cIBM/ / </pre>	
<pre> sim2unix.sed /^\$LARGE/d /^\$INCLUDE: 'simpdef.f'/r simpdef.f /^\$INCLUDE: 'simpdef.f'/d /^\$INCLUDE: 'simpget.f'/r simpget.f /^\$INCLUDE: 'simpget.f'/d /^cSUN/s/cSUN/ / </pre>	

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c sp - signal power (calculated by modu)
c oldsnr - fading waveform average snr
c if1 - Index into array corresponding to value of f1
c if2 - Index into array corresponding to value of f2
c nf - if2-if1+1 = number of frequencies between if1 and if2
c n0, n1 - number of non-error and errors (for runs test)
c nr - number of runs (for runs test)
c r - normalized number of runs statistic (for runs test)

integer i, j, k, l, m, n
integer n0, n1, nr
integer if1, if2, nf
real r
real oldsnr

c statistics functions: mean, lower and upper .95 CI
real stmn, stl95, stu95

c convert power to dB
real dbp

c transmitted, received, and corrected bits and signal sample vectors:
logical txdata(ns), crdata(ns)
real txsig(ns), rxsig(ns), crsig(ns)

c the common block is a ns-real-long temporary work area
c used by interleaving and diversity routines
real wrk(4*ns)
common wrk

c noise and fading waveform sample vectors:
real noise(ns), fade(ns)

c pre-emphasis and de-emphasis/channel-inversion vectors
real prev(ns), dev(ns)

c error measurement variables:
c npass - count of dfb correction passes
c nbe1 - bit error count on previous pass
c nbe2 - bit error count on current pass
c nwe2 - word error count on current pass
c nbe, nwe - bit and word errors (in one trial)
c ber, bker - sums of BERS and BkERS

c runs - sums of number of runs statistics
c ber2, bker2 - sums of squares of BERS and BkERS
c runs2 - sums of squares of number of runs statistics
c nber, nbker - number of BER and BKER trials
c nruns - number of runs statistics
c nb, nw - number of bits/words per trial
c nbx, nwx - number of bits/words examined in a trial

integer nbe1, nbe2, nwe2, npass
integer nbx, nwx
integer nbe (mxblk, mxsnr), nwe (mxblk, mxsnr)
integer nb (mxblk, mxsnr), nw (mxblk, mxsnr)
real ber (mxblk, mxsnr), bker (mxblk, mxsnr)
real runs (mxblk, mxsnr)
real ber2 (mxblk, mxsnr), bker2 (mxblk, mxsnr)
real runs2 (mxblk, mxsnr)
integer nber (mxblk, mxsnr), nbker (mxblk, mxsnr)
integer nruns (mxblk, mxsnr)

c test dibits or bits for runs test
logical dibits
parameter (dibits=.false.)

c initialize BER, BKER, and runs statistics variables
data ber /mxblsn*0./, bker /mxblsn*0./, runs /mxblsn*0./
data ber2/mxblsn*0./, bker2/mxblsn*0./, runs2/mxblsn*0./
data nber/mxblsn*0 //, nbker/mxblsn*0 //, nruns/mxblsn*0 /

c *** program starts here ****
write(*,*)'% OFDM Simulation - 88-6-4.'
c -----
$INCLUDE: 'simpget.f'
c -----
if(fs.le.0)then
write(*,*) ' simrun: bad fs '
stop
endif
if((2*nbe+n)*2.gt.4*ns)then
write(*,*) ' simrun: ne too large '
stop
endif

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c -----
      if(hw)then
c measure channel transfer function and generate pre/de-emphasis filter
c initialize for block of maximum lenght (ns), with sample work array
c enough for 4 seconds (4*8000=32000). txsig is temporary work vector.
          call hwinit(txdata,prev,dev,txsig,neqbl,
+           wrk,ns,8*ns,fs,f1,f2,dbd,rms,peak,txemp,demp,
+           emp scl,nemp scl,emp fr,emp sc)
      else
c initialize snr-to-signal and snr-to-noise tables
          call s2init(b,w,rms,peak,fm,fd,rfm,agclim,sqlim,
+           nints,intsr,intss, nintn,intnr,intnn,
+           fading, noising )
      endif
c do ntr trials
          do 6 i=1,ntr
c reset bit, word, bit error and word error counts for this trial
              call vifill(nb ,mxblsn,0)
              call vifill(nw ,mxblsn,0)
              call vifill(nbe,mxblsn,0)
              call vifill(nwe,mxblsn,0)
c do "nblk" blocks per trial
              do 4 j=1,nblk
c generate data bits
                  if(np at.eq.0)then
                      call prbs(txdata,ns,sr)
                  else
                      call dwg(txdata,ns,pat,np at)
                  endif
                  if(.not.hw)then
c if noise enabled, generate a noise vector for proper noise density
c scaling is because measured noise powers only include frequencies

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c in the data range.
          if(noising)then
              if(impse)then
                  call vimp(noise,ns,prbimp,nseed)
              else
                  call vgrand(noise,ns,nseed)
              endif
              call vsmul(noise,ns,sqrt( (fs/2.0)/(f2-f1) ))
          else
              call vfill(noise,ns,0.)
          endif
c if fading enabled, generate snr values with 0 dB mean
          if(fading)then
              call genfdb(fd/fs,fseed,fade,ns,
+           ndbr,thr,ndw,ns w)
          +
          else
              call vfill(fade,ns,0.)
          endif
          oldsnr=0.
      endif
c do for all snrs
          do 3 l=1,nsnr
c set the fading waveform average snr
              if(.not.hw)then
                  call setsnr(fade,ns,oldsnr,snr(l))
              endif
c do for all data block sizes
              do 3 k=1,nn
c set the block size
                  n=na(k)
                  if(n.le.0.or.n.gt.ns)then
                      write(*,*) ' simrun: bad N '
                      stop
                  endif
c set lower/upper frequency limits and number of channels

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<pre> call setif(fs,n,f1,if1,f2,if2) nf=if2-if1+1 do 3 m=1,ns,n encode the data into complex data values call encode(txdata(m),txsig,if1,if2,n) do pre-emphasis if sending over HW channel if(hw)then call emp(txsig,prev,f1,f2,fs,n,ns) endif modulate the data into an ofdm signal (unit variance) call modu(txsig,n,nf,serial) send signal through noisy, fading channel call vquant(txsig,n,tmax,tquant) if(hw)then call hwch(txsig,crsig,wrk,n,n+2*ne,rms,peak) else call ch(txsig,crsig,noise(m),fade(m),n,rms,peak) endif call vquant(crsig,n,rmax,rquant) demodulate call demodu(crsig,n,nf,serial) do de-emphasis if sent over HW channel if(hw)then call emp(crsig,dev,f1,f2,fs,n,ns) endif call decode(crdata,crsig,if1,if2,n) do [fec &] error checking call fec(crdata,txdata(m),nf,intlv,ecn,ect, nbe2,nbx,nwe2,nwx) </pre>	<pre> c continue correction passes until exceed iteration limit or no errors if(mxdfbp.gt.0)then call vcopy(crsig,rxsig,n) call modu(rxsig,n,nf,serial) npass=0 nbel=ns+1 1 continue if((nbe2.eq.0) .or. (npass.gt.mxdfbp))goto 2 c remodulate received data call encode(crdata,crsig,if1,if2,n) call modu(crsig,n,nf,serial) c use original signal where received level > fade limit call vsel(crsig,rxsig,fade(m),fdlim(1),n) c demodulate the composite signal call demodu(crsig,n,nf,serial) call decode(crdata,crsig,if1,if2,n) npass=npass+1 nbel=nbe2 c do [fec &] error checking + call fec(crdata,txdata(m),nf,intlv,ecn,ect, + nbe2,nbx,nwe2,nwx) 2 goto 1 continue if(dndfbp)then write(*,'(1X,A2,I5,F4.0,I3,I5)') '\$D', n, snr(1), npass, nbe2 endif c if required, display signal and noise powers in this block if(dsn)then call ddsn(crsig(if1),txdata(m),nf, f1,fs,n,snr(1),ndsn,wrk(1),wrk(ns+1)) + </pre>
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endif
if(dsnv)then
  call davn(crsig(if1),txdata(m),nf,
    n,snr(1),wrk(1),wrk(ns+1))
endif
+
c if required, display transmitted and received signal values
if(odd)then
  call vddat(crsig(if1),txdata(m),nf,
    f1,f2,n,snr(1))
endif
+
c if required, display number of errors in each fec block
if(dnerr)then
  call dne(crdata,txdata(m),nf,intlv,ecn,n,snr(1))
endif
c update bit/word add bit/word error counts
nbe(k,1)=nbe(k,1)+nbe2
nb(k,1)=nb(k,1)+nbx
nwe(k,1)=nwe(k,1)+nwe2
nw(k,1)=nw(k,1)+nwz
c write(*,*) ' blk, BER = ', j, float(nbe2)/nbx
c do runs test if required, only consider block with minrn+ runs
if(rnstst)then
  if(dibits)then
    call vl2diff(crdata,txdata(m),nf,wrk,nf/2)
    call runcnt(wrk,nf/2,n0,n1,nr,r)
  else
    call vxor(crdata,txdata(m),wrk,nf)
    call runcnt(wrk,nf,n0,n1,nr,r)
  endif
  if(nr.ge.minrn)then
    call stat(r,runs(k,1),runs2(k,1),nruns(k,1))
  endif
endif
3 continue
4 continue
c update BER and BKER statistics
do 5 k=1,nn

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do 5 l=1,nsnr
  n=na(k)
  if(nb(k,1).ne.0)
    call stat(float(nbe(k,1))/nb(k,1),
      ber(k,1),ber2(k,1),nber(k,1))
  +
  if(nw(k,1).ne.0)
    call stat(float(nwe(k,1))/nw(k,1),
      bker(k,1),bker2(k,1),nbker(k,1))
  +
  continue
5 continue
6 continue
c display results: runs tests
if(rnstst)then
  do 7 k=1,nn
  do 7 l=1,nsnr
    if(nruns(k,1).ge.2)then
      write(*,'(4X,A,I6,F6.1,2(1PE11.2))')%,na(k),snr(1),
        stl95(runs(k,1),runs2(k,1),nruns(k,1)),
        stu95(runs(k,1),runs2(k,1),nruns(k,1))
    +
    else
      write(*,'(4X,A,I6,F6.1,A)')%,na(k),snr(1),
        ': too few runs.'
    +
    endif
  continue
7 endif
c display results: mean BERs and BkERs (with .95 CI)
write(*,*) '% bit error rates : '
do 8 k=1,nn
do 8 l=1,nsnr
  if(nber(k,1).ne.0)then
    write(*,'(1X,I6,F6.1,3(1PE11.2))')na(k),snr(1),
  +
  stmn(ber(k,1),ber2(k,1),nber(k,1)),
  +
  stl95(ber(k,1),ber2(k,1),nber(k,1)),
  +
  stu95(ber(k,1),ber2(k,1),nber(k,1))
  endif
8 continue
write(*,*) '% FEC block error rates : '
do 9 k=1,nn
do 9 l=1,nsnr
  if(nbker(k,1).ne.0)then
    write(*,'(1X,I6,F6.1,3(1PE11.2))')na(k),snr(1),
  +
  stmn(bker(k,1),bker2(k,1),nbker(k,1)),
  +
  stl95(bker(k,1),bker2(k,1),nbker(k,1)),
  +
  stu95(bker(k,1),bker2(k,1),nbker(k,1))
  endif
endif

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9	continue		
	end		
		simpdef.f	
c	*****		
c	simpdef.f - define simulation parameters		
c	*****		
c	-----		
c	array sizes:		
c	ns - maximum number of samples in each "simulation block"		
c	mxpat - maximum number of elements in npat		
c	mxblk - maximum number of elements in na		
c	mxsnr - maximum number of elements in snr		
c	mxblsn - mxblk * mxsnr		
c	srln - number of elements in sr (must be 23 for CCITT v.29)		
c	mxint - number of elements in intrn, intnr, intrs, intsn		
c	mxthr - maximum number of elements in thrsh		
	integer ns, mxblk, mxsnr, mxblsn, mxpat, srln, mxint, mxemp		
	integer mxthr		
	parameter (ns=4096 ,mxblk=10, mxsnr=10, mxblsn=100, mxpat=20)		
	parameter (srln=23, mxint=100, mxemp=50, mxthr=30)		
c	agclim - agc gain limit (SNR dB)		
c	agcvar - true to test effect of varying agc threshold (pint only)		
c	b - IF bandwidth (kHz)		
c	dbd - dB per decade of pre-emphasis		
c	demp - true to display equalization characteristics		
c	dndfbb - true to display number of correction passes for each block		
c	dnerrr - true to display number of errors per fec work		
c	dsn - true to display received data value signal and noise power		
c	dsnav - true to display average of received signal and noise powers		
c	ddat - true to display transmitted and received data values		
c	ecn - word size		
c	ect - correctable bit errors per word (0=no FEC)		
c	empfr - upper frequency limit to use for scaling values in empsc		
c	empsc - scaling values (in dB) for scaling pre-emphasis vectors		
c	empsc1 - true to do scaling of pre-emphasis vector		
c	f1 - lowest frequency to use on channel (<= 0 -> use minimum)		
c	f2 - highest frequency to use on channel (>= fs/2 -> use maximum)		
c	fading - true to apply fading		
c	fd - doppler rate (Hz)		
c	fdlim - "faded" decision levels for DFB correction (SNR dB), per SNR		
c	fm - true for FM, false for SSB		
c	fs - sample rate (bps)		
c	fseed - RNG seed to initialize fading generator (limits as above)		
c	hw - true to send samples over A/D/A board (through hardware f.s.)		
c	impuse - true to use IMPulse (cf. gaussian) Noise		
c	intlv - true to do interleaving before checking for word errors		
c	intnn - "n" part of points to use to change r-n table		
c	intrn - SNR part of points to use to change r-n table		
c	intrs - SNR part of points to use to change r-s table		
c	intss - "s" part of points to use to change r-s table		
c	minrms - minimum number of runs in block to use it for runs testing		
c	mxdfbb - maximum number of dfb correction passes (0=no DFB)		
c	na - array of OFDM block sizes, each a power of 2 in [8,ns]		
c	nblk - number of simulation blocks (ns samples) per trial		
c	ndbr - Number of Diversity BRanches		
c	ndsn - number of channels averaged when display signal/noise powers		
c	ndw - Number of samples to wait (DWell) after threshold crossed		
c	ne - number of guard samples for hardware channel		
c	nempsc - number of values in empsc and empfr		
c	negbl - number of blocks to average in measuring channel equalization		
c	nintn - number of elements in intrn and intnn		
c	nints - number of elements in intrs and intss		
c	nn - number of OFDM block sizes to be tested		
c	noisng - true to add noise		
c	npat - Number of elements in data PATTERN (0 for random data)		
c	nseed - RNG seed for noise generators (0 <= nseed < 67108864)		
c	nsnr - number of SNRs to test		
c	nsw - Number of samples to blank output while Switching branches		
c	nthrsh - number of squelch or agc thresholds to test (pint only)		
c	ntr - number of trials done each time subroutine called		
c	pat - description of data PATTERN		
c	pfile - name of simulation parameter file		
c	rms - rms voltage of modulating signal		
c	peak - peak voltage of modulating signal		
c	prbimp - probability of a sample having an impulsive noise "hit"		
c	rffm - true to add Random FM		
c	rmax - Receiver A/D MAXimum (clipping) value		
c	rnstst - true to compute runs tests		
c	rquant - bits of Receiver A/D QUANTization (0=no quantization)		
c	serial - true for "serial" (no OFDM) simulation		
c	snr - IF SNRs (dB)		
c	sglim - squelch limit (SNR dB)		
c	sgval - true to test effect of varying squelch threshold (pint only)		
c	sr - shift register logical values to generate PRBS data stream		
c	thr - switching threshold (dB rel. to mean)		
c	thrsh - agc or squelch thresholds to test		
c	tmax - Transmitter D/A MAXimum (clipping) value		
c	tquant - bits of Transmitter D/A QUANTization (0=no quantization)		
c	txemp - true to amp./phase correction at transmitter (not receiver)		
c	vname - variable name		
c	w - baseband bandwidth (kHz)		
	integer na(mxblk), nn, ntr, nblk		

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integer ecn, ect
integer ndbr, npat, pat(mxpat), nsnr
integer tquant, rquant
integer nsw, ndw
integer ne, mxdfbp, neqbl
integer nintr, nints, minrns
integer ndsn, nempsc, nthrsh
real fs, fd, snr(mxsnr), fdlim(mxsnr)
real agclim, sqlim, b, w, rms, peak
real prbimp
real f1, f2, tmax, rmax
real dbd
real intr(mxint), intrn(mxint), intr(mxint), intss(mxint)
real empsc(mxemp), empfr(mxemp)
real thrsh(mxthr)
double precision nseed, fseed
logical fm, fading, noising, rfm, intlv, impnse, hw, serial
logical dndfbb, dnerr, dsn, dsnav, demp, ddat, rnstst, txemp
logical emp scl, agcvar, sqvar
logical sr(sr len)
character*70 vname, pfile

c intrinsic functions
integer min

c -----
simpget.f
c *****
c simpget.f - read simulation parameters
c *****
c -----
write(*,*) '% simulation parameter file ? '
read(*, '(A70)') pfile
write(*,*) '% ', pfile

open(10, file=pfile)

read(10,*) vname
write(*,*) '% ', vname

read(10,*) vname, nn, (na(i), i=1, min(nn, mxblk))
if(nn.gt.mxblk) then
  write(*,*) '% simpget : too many blocks sizes '
endif

stop
write(*,*) '% na(i) = ', (na(i), i=1, nn)

read(10,*) vname, ntr, nblk
write(*,*) '% ntr, nblk = ', ntr, nblk

read(10,*) vname, fs, fd
write(*,*) '% fs, fd = ', fs, fd

read(10,*) vname, nsnr, (snr(i), i=1, min(nsnr, mxsnr))
if(nsnr.gt.mxsnr) then
  write(*,*) '% simpget : too many SNRs '
endif

stop
write(*,*) '% snr(i) = ', (snr(i), i=1, nsnr)

read(10,*) vname, serial
write(*,*) '% serial = ', serial

read(10,*) vname, fm, b, w, rms, peak
write(*,*) '% fm, b, w, rms, peak = ', fm, b, w, rms, peak

read(10,*) vname, agclim, sqlim
write(*,*) '% agc, sqlim = ', agclim, sqlim

read(10,*) vname, fading, noising
write(*,*) '% fading, noising = ', fading, noising

read(10,*) vname, fseed, nseed
write(*,*) '% fseed, nseed = ', fseed, nseed

read(10,*) vname, rnstst, minrns
write(*,*) '% rnstst, minrns = ', rnstst, minrns

read(10,*) vname, mxdfbb, (fdlim(i), i=1, nsnr)
write(*,*) '% mxdfbb, fdlim(i) = ', mxdfbb, (fdlim(i), i=1, nsnr)

read(10,*) vname, dndfbb, dnerr, demp, ddat
write(*,*) '% dndfbb, dnerr, demp, ddat = ',
+ dndfbb, dnerr, demp, ddat

read(10,*) vname, dsn, ndsn, dsnav
write(*,*) '% dsn, ndsn, dsnav = ', dsn, ndsn, dsnav

read(10,*) vname, intlv
write(*,*) '% intlv = ', intlv

read(10,*) vname, ecn, ect
write(*,*) '% ecn, ect = ', ecn, ect

```



```

read(10,*) vname, ndbr, thr, ndw, nsw
write(*,*) '% ndbr, thr, ndw, nsw =', ndbr, thr, ndw, nsw

read(10,*) vname, npat, (pat(i),i=1,min(npatt,mpat))
if(npatt.gt.mpat)then
  write(*,*) ' simpget : too many pattern elements '
  stop
endif
write(*,*) '% pat(i) =', (pat(i),i=1,npatt)

read(10,*) vname, (sr(i),i=1,srlen)
write(*,*) '% sr(i) =', (sr(i),i=1,srlen)

read(10,*) vname, rfm
write(*,*) '% rfm =', rfm

read(10,*) vname, impnse, prbimp
write(*,*) '% impnse, prbimp =', impnse, prbimp

read(10,*) vname, hw, ne, dbd, neqbl, txemp
write(*,*) '% hw, ne, dbd, neqbl, txemp =',
+ hw, ne, dbd, neqbl, txemp

read(10,*) vname, f1, f2
write(*,*) '% f1, f2 =', f1, f2
if(f2.le.f1)then
  write(*,*) ' simpget : f2 <= f1 '
  stop
endif

read(10,*) vname, tquant, tmax, rquant, rmax
write(*,*) '% tquant, tmax, rquant, rmax =',
+ tquant, tmax, rquant, rmax

read(10,*) vname, nints,
+ (intr(i), intss(i), i=1,min(nints,mxint))
if(nints.gt.mxint)then
  write(*,*) ' simpget : too many signal interpolation values '
  stop
endif
write(*,*(a,1000(/' ',2f10.3)))
+ '% intr(i), intss(i) =',
+ (intr(i), intss(i), i=1,nints)

read(10,*) vname, nintn,
+ (intr(i), intnn(i), i=1,min(nintn,mxint))
if(nintn.gt.mxint)then
  write(*,*) ' simpget : too many noise interpolation values '
  stop
endif

read(10,*) vname, agcvar, sqvar, nthrsh,
+ (thrsh(i), i=1,min(nthrsh,mxthr))
if(nthrsh.gt.mxthr)then
  write(*,*) ' simpget : too many agc/squelch test thresholds '
  stop
endif
write(*,*) '% agcvar, sqvar =', agcvar, sqvar
write(*,*(a,1000(/' ',f6.2)))
+ '% thrsh(i) =', (thrsh(i), i=1,nthrsh)

read(10,*) vname, emp scl
write(*,*) '% emp scl =', emp scl

read(10,*) vname, nempsc,
+ (empfr(i), emp sc(i), i=1,min(nempsc,mxemp))
if(nempsc.gt.mxemp)then
  write(*,*) ' simpget : too many pre-emphasis scaling values '
  stop
endif
write(*,*(a,1000(/' ',2f10.3)))
+ '% empfr(i), emp sc(i) =',
+ (empfr(i), emp sc(i), i=1,nempsc)

close(10)
c -----

```

simfde.f

```

c *****
c simfde.f - fading envelope generator (Jake's method)
c *****
c 87-7-9
c -----
c
c subroutine genfd(fdbyps,seed,x,ns,b)
c generate samples of the fading envelope
c method from W. C. Jakes, 1974, p. 65
c output vector (x) is in dB
c real fdbyps

```

<pre> integer ns real x(ns) double precision seed logical init integer b, m c constant to scale envelope to unity rms real k c max number of diversity branches integer nb parameter (nb=10) c storage for phase counters for diversity branches integer*4 phs(9,nb) c word and full-word length phase counter integer*4 pl, p2, p3, p4, p5, p6, p7, p8, p9 integer*2 pw(18) integer*2 il, i2, i3, i4, i5, i6, i7, i8, i9 equivalence (pf,pw) equivalence (pf(1), p1), (pf(2), p2), (pf(3), p3), + (pf(4), p4), (pf(5), p5), (pf(6), p6), + (pf(7), p7), (pf(8), p8), (pf(9), p9) c only one of the following two tables can be used c using the wrong table will produce out-of-range subscripts c this table for computers that store MS INTEGER*2 value first c e.g. MOTOROLA 68000 (SUN) cSUN equivalence (pw(1), i1), (pw(3), i2), (pw(5), i3), cSUN + (pw(7), i4), (pw(9), i5), (pw(11), i6), cSUN + (pw(13), i7), (pw(15), i8), (pw(17), i9) c this table for computers that store LS INTEGER*2 value first c e.g. INTEL 8088 (IBM PC), DEC VAX-11 cIBM equivalence (pw(2), i1), (pw(4), i2), (pw(6), i3), cIBM + (pw(8), i4), (pw(10), i5), (pw(12), i6), cIBM + (pw(14), i7), (pw(16), i8), (pw(18), i9) c itl is the table length for each sine/cosine lookup table c *** values of itl must match *** </pre>	<pre> integer itl parameter (itl=2048) c scaled sine and cosine lookup tables integer*2 c1(itl), c2(itl), c3(itl), c4(itl), c5(itl), + c6(itl), c7(itl), c8(itl), c9(itl) integer*2 s1(itl), s2(itl), s3(itl), s4(itl), s5(itl), + s6(itl), s7(itl), s8(itl), s9(itl) c I and Q sums integer*2 sumi, sumq c phase increments per sample integer*4 dp1, dp2, dp3, dp4, dp5, dp6, dp7, dp8, dp9 common /fdtab1/ c1, c2, c3, c4, c5, c6, c7, c8, c9, + s1, s2, s3, s4, s5, s6, s7, s8, s9 common /fdtab2/ dp1, dp2, dp3, dp4, dp5, dp6, dp7, dp8, dp9 real tpi, oldfds integer i, j real uni real float data oldfds/0., init/.false./ data tpi/6.28318/ c compute look-up tables if have not been initialized if(.not.init)then call initos(k) init=.true. endif c compute phase increments and initialize oscillator phases c if doppler or sampling rate have changed if(fdbyps.ne.oldfds)then call initph(fdbyps) oldfds=fdbyps </pre>
---	---

```

c initialize phases using uniform RNG
      m=1
      do 1 i=1,nb
        do 1 j=1,9
          phs(j,i)=itl*65536*uni(seed)
1
        continue
      endif
c range-check branch number
      if((b.lt.1).or.(b.gt.nb))then
        write(*,*)'genfd:diversity branch out of range.'
        stop
      endif
c copy branch phases to phase counters
      do 2 i=1,9
        pf(i)=phs(i,b)
2
      continue
c generate ns samples
      do 3 i=1,ns
c increment oscillator phases
        p1=p1+dp1
        p2=p2+dp2
        p3=p3+dp3
        p4=p4+dp4
        p5=p5+dp5
        p6=p6+dp6
        p7=p7+dp7
        p8=p8+dp8
        p9=p9+dp9
c keep phase modulo two pi
        if(i1.gt.itl)i1=i1-itl
        if(i2.gt.itl)i2=i2-itl
        if(i3.gt.itl)i3=i3-itl
        if(i4.gt.itl)i4=i4-itl
        if(i5.gt.itl)i5=i5-itl
        if(i6.gt.itl)i6=i6-itl
        if(i7.gt.itl)i7=i7-itl
        if(i8.gt.itl)i8=i8-itl
      do 4 i=1,9
        phs(i,b)=pf(i)
4
      continue
      return
      end
c ---
      subroutine initos(k)
c initialize oscillator lookup tables
      real pi, tpi, sqrt2
      real N, NO
      real stk, ctk, k
      integer i, j
      real float, cos, sin
c itl is the table length for each sine/cosine lookup table
c *** values of itl must match ***
      integer itl
      parameter (itl=2048)
c scaled sine and cosine lookup tables
      integer*2 cw(itl,9), sw(itl,9)
      real rc(itl)
      if(i9.gt.itl)i9=i9-itl
c add up two sets of scaled oscillator outputs and find mag. squared
      sumi=c1(i1)+c2(i2)+c3(i3)+c4(i4)+
1      c5(i5)+c6(i6)+c7(i7)+c8(i8)+c9(i9)
      sumg=s1(i1)+s2(i2)+s3(i3)+s4(i4)+
1      s5(i5)+s6(i6)+s7(i7)+s8(i8)+s9(i9)
      x(i) = k * ( float(sumi)**2 + float(sumg)**2 )
3
      continue
c convert to dB (power) (x is magnitude squared of envelope)
      call vdbp(x,ns)
c copy phase counters back
      do 4 i=1,9
        phs(i,b)=pf(i)
4
      continue
      return
      end
c ---
      subroutine initos(k)
c initialize oscillator lookup tables
      real pi, tpi, sqrt2
      real N, NO
      real stk, ctk, k
      integer i, j
      real float, cos, sin
c itl is the table length for each sine/cosine lookup table
c *** values of itl must match ***
      integer itl
      parameter (itl=2048)
c scaled sine and cosine lookup tables
      integer*2 cw(itl,9), sw(itl,9)
      real rc(itl)

```

```

common /fdtab1/ cw, sw
data pi/3.14159/, tpi/6.28318/, sqrt2/1.414213/
N0=8.
N=4.*N0+2.
c compute un-scaled oscillator (cosine) look-up table
do 1 j=1,itl
  rc(j)=cos(float(j-1)/itl*tpi)
  continue
1
c generate scaled oscillator tables
k=0.
do 3 i=1,8
  ctk=1750.*2.*cos(pi*float(i)/N0)
  stk=1750.*2.*sin(pi*float(i)/N0)
  k=ctk**2+stk**2
  do 2 j=1,itl
    cw(j,i)=rc(j)*ctk
    sw(j,i)=rc(j)*stk
  continue
2
3
c last oscillator has different amplitude
ctk=1750.*1./sqrt2*2.*cos(pi/4.)
stk=1750.*1./sqrt2*2.*sin(pi/4.)
k=ctk**2+stk**2
do 4 j=1,itl
  cw(j,9)=rc(j)*ctk
  sw(j,9)=rc(j)*stk
4
c compute normalization constant
k=2./k
write(*,*)'% oscillator tables (re)initialized '
return
end
subroutine initph(fdbyfs)
-----
c

```

```

c initialize phase increment table
real fdbyfs
real tpi
real N, N0
integer i
real cos, float
c itl is the table length for each sine/cosine lookup table
c *** values of itl must match ***
integer itl
parameter (itl=2048)
integer*4 dpf(9)
common /fdtab2/ dpf
data tpi/6.28318/
if((fdbyfs.lt.0.).or.(fdbyfs.ge.0.5))then
  write(*,*)'genfd:doppler too high or negative.'
  stop
endif
c compute phase increments
N0=8.
N=4.*N0+2.
do 1 i=1,8
  dpf(i)=cos(tpi*float(i)/N)*fdbyfs*itl*65536.
  continue
  dpf(9)= fdbyfs*itl*65536.
write(*,*)'% phase increments (re)initialized '
write(*,*)'% fdbyfs = ',fdbyfs
return
end
-----
subroutine setsnr(fade,n,oldsnr,newsnr)
integer n
real fade(n), oldsnr, newsnr
real delta
-----

```

<pre> delta-newsnr-olddsnr call vadd(fade,n,delta) olddsnr=newsnr return end </pre>	<pre> c m - number of samples to generate in one pass c thr - switching THRESHOLD (fraction of mean=1) c maxbuf - size of local buffer integer i, j, k, nleft, m integer bufsiz, maxbuf parameter (maxbuf=1000) c buffer array (in common work area) real buf(maxbuf) common buf c intrinsic functions integer min, int c range-check number of diversity branches if(nb.lt.1)then write(*,*) ' genfdb: nb < 0 ' stop endif </pre>
<pre> ----- simdiv.f ----- c ***** c simdiv.f - diversity c ***** c 87-11-29 c ----- c subroutine genfdb(fdbyps,seed,x,ns, + nb,thr,ndw,ns) c fading envelope generator (in db) c parameters: c fdbyps - Doppler rate / Sampling rate c seed - RNG seed for initializing fading generator c x - output vector containing fading sequence (dB) c ns - Number of values in x c nb - Number of diversity Branches c nb>1 means diversity c thr - switching diversity THRESHOLD (dB relative to mean) c ndw - minimum Number of samples to DWell on branch c before switching c ns - Number of samples lost while SWITCHING branches real fdbyps double precision seed integer ns, nb real x(ns) real thr integer ndw, ns c local variables: c i - index into output vector x c j - branch counter c k - temporary variable for indexing c nleft - Number of samples LEFT to generate </pre>	<pre> c m - number of samples to generate in one pass c thr - switching THRESHOLD (fraction of mean=1) c maxbuf - size of local buffer integer i, j, k, nleft, m integer bufsiz, maxbuf parameter (maxbuf=1000) c buffer array (in common work area) real buf(maxbuf) common buf c intrinsic functions integer min, int c range-check number of diversity branches if(nb.lt.1)then write(*,*) ' genfdb: nb < 0 ' stop endif c initialize for piece-wise generation of ns samples i=1 nleft=ns bufsiz=int(maxbuf/nb) c generate (and combine) "nb" sections of "bufsize" samples per pass 1 continue m=min(nleft,bufsiz) c generate branch signals do 2 j=1,nb k=(j-1)*m+1 call genfd(fdbyps,seed,buf(k),m,j) 2 continue c combine branches if(nb.eq.1)then call vcopy(buf,x(i),m) else call divsw(buf,nb,m,x(i),nsw,ndw,thr) endif </pre>

<pre> i=i+m nleft=nleft-m c repeat until done if(nleft.gt.0)goto 1 return end c ----- subroutine divsw(buf,nb,m,x,nsw,ndw,thr) c switching diversity routine c NOTE: the switching and dwell counts are static so this c routine cannot be used to generate independent c diversity-switched signals. c parameters: c buf - input buffer (m by nb) with m samples from each of c nb branches c nb - number of diversity branches c m - number of samples per branch c x - output vector containing the resulting fading sequence c ndw - minimum Number of samples to Dwell on one branch (>=1) c (after switching is complete) c nsw - Number of samples blanked out while Switching branches c thr - switching THRESHOLD (dB relative to mean) integer nb, m, ndw, nsw real buf(m,nb), x(m), thr c local (static!) variables: c br - current BRANCH c idw - count for timer that indicates that must Dwell on c current branch c isw - count for timer for Switching time c zval - VALUE value (dB) to indicate samples that should be c Zero'ed. (anything > 1E6 can be used). used by "ch" c routine. c maxdw - maximum dwell count (to avoid overflows) integer br, idw, isw, i real zval parameter(zval=2.0e6) </pre>	<pre> data br/1/, idw/0/, isw/0/ c write(*,*) ' thr = ',thr c range checks if(nsw.lt.0)then write(*,*) ' divsw: nsw < 0 ' stop endif if(br.gt.nb)then write(*,*) ' divsw: branch > number of branches ' stop endif if(ndw.lt.0)then write(*,*) ' divsw: ndw < 0 ' stop endif c do for all samples ... do 1 i=1,m c if below threshold not switching and and dwell timer expired, c go to next antenna and reset timers if(buf(i,br).lt.thr .and. isw.eq.0 .and. idw.eq.0)then br=br+1 isw=nsw idw=ndw+nsw endif c if switching (blanking) timer has expired, select current branch's c level, else output is blanked if(isw.eq.0)then x(i)=buf(i,br) else x(i)=zval endif c if still on, decrement switching timer if(isw.gt.0)then isw=isw-1 endif </pre>
---	---

```

c if still on, decrement channel dwell timer
   if(idw.gt.0)then
     idw=idw-1
   endif
c   write(*,*) (buf(i,1),i=1,nb), '=> ', br, x(i)
1  continue
   return
   end

```

simgen.f

```

c *****
c simgen.f - data and noise generators
c *****
   subroutine prbs(x,n,sr)
c PRBS generator
c The generator's shift register is implemented as a logical array.
c For efficiency, instead of shifting the array to the right (up)
c pointers to the exclusive-or gate connections and the LS bit are
c shifted left (down the array).
c The tap connection (18,23) are taken from the data scrabler
c described in Appendix 2 of the CCITT standard (V.29) for a
c 9600 bps modem. The pattern period is 2**23-1 = 8 388 607.
   integer len
   parameter (len=23)
c i - counts output values
c i0, i1, i2 - point to least significant bit and XOR connections
c   in shift register array
c n - number of values to put in x
c x - output logical array
c len - length of shift register
c sr - a logical array of len elements corresponding to the
c   elements of a prbs generator shift register.
   integer i, i0, i1, i2, n
   logical sr(len)
   logical x(n)

```

```

   i0=1
   i1=18
   i2=23
   do 1 i=1,n
     i0=i0-1
     if(i0.eq.0)i0=len
     if(sr(i1))then
       if(sr(i2))then
         sr(i0)=.false.
         x(i)=.false.
       else
         sr(i0)=.true.
         x(i)=.true.
       endif
     else
       if(sr(i2))then
         sr(i0)=.true.
         x(i)=.true.
       else
         sr(i0)=.false.
         x(i)=.false.
       endif
     endif
     i1=i1-1
     if(i1.eq.0)i1=len
     i2=i2-1
     if(i2.eq.0)i2=len
     1 continue
   call rotldwn(sr,len,i0-1)
   return
   end
   subroutine rotldwn(sr,n,i)
c rotate (circular shift) the first "n" elements of logical array
c "sr" down the array by "i" places.
   integer n, i
   logical sr(n)
   integer j, k

```

<pre> logical t do 2 j=1,i t=sr(1) do 1 k=1,n-1 sr(k)=sr(k+1) continue sr(n)=t continue return end subroutine vimp(x,n,p,seed) c generates impulse noise vector c samples are uncorrelated with equal probability of an impulse = p c zero mean, unit average power c calls uniform RNG "uni" integer i, n real x(n), p, u real sqrt, uni double precision seed if(p.lt.0.or.p.gt.1)then write(*,*)' vimp: p out of range. ' stop endif u=sqrt(1./p) do 1 i=1,n if(uni(seed).lt.p)then if(uni(seed).gt.0.5)then x(i)=u else x(i)=-u endif else x(i)=0 endif endif continue return end subroutine dwg(x,n,y,m) c digital waveform generator </pre>	<pre> c x - output logical vector c n - number of values in x c y - pattern definition vector (number false, number true)... c m - number of elements in m (>=1) c if pattern counts in y are insufficient to fill x, the pattern is c repeated starting at y(3) c i - index into x c j - index into y c k - counts up to y(j) integer i, j, k, n, m logical x(n), tf integer y(m) if(m.lt.3)then write(*,*)'dwg:pattern definition < 3 elements.' stop endif do 1 i=1,y(1) x(i)=.false. continue do 2 i=y(1)+1,y(1)+y(2) x(i)=.true. continue tf=.false. j=3 k=0 do 3 i=y(1)+y(2)+1,n if(k.ge.y(j))then tf=.not.tf j=j+1 k=0 if(j.gt.m)then j=3 endif endif x(i)=tf k=k+1 continue return end </pre>
---	--

<pre> subroutine vgrand(xo,n,seed) integer n real xo(n) integer i real r, rx, ry double precision seed, dmod real sqrt,alog c function that returns a Gaussian distributed random number c of zero mean, unity variance. c adapted from IEEE DSP program library (in MXFFT.FOR) do 1 i=1,n,2 10 continue seed=dmod(67081293.0d0*seed+14181771.0d0,67108864.0d0) RX=(seed/67108864.0d0*2.0) - 1.0 seed=dmod(67081293.0d0*seed+14181771.0d0,67108864.0d0) RY=(seed/67108864.0d0*2.0) - 1.0 C----- C SUBROUTINE: NORMAL C GENERATES AN INDEPENDENT PAIR OF RANDOM NORMAL DEVIATES C METHOD DUE TO G. MARSAGLIA AND T.A. BRAY, C SIAM REVIEW, VOL. 6, NO. 3, JULY 1964. 260-264 C----- C C OUTPUT: X,Y = INDEPENDENT PAIR OF RANDOM NORMAL DEVIATES C FUNCTION UNI GENERATES PSEUDO-RANDOM NUMBER BETWEEN 0.0 AND 1.0 C R = RX**2 + RY**2 IF (R.GE.1.0) GO TO 10 R = SQRT(-2.0*ALOG(R)/R) xo(i)=RX*R if(i+1.le.n)then xo(i+1)=RY*R endif 1 continue return end real function uni(seed) </pre>	<pre> c modulo congruential uniform RNG on [0,1) c parameters are from FPS AP library routine VRAND double precision seed double precision dmod seed=dmod(67081293.0d0*seed+14181771.0d0,67108864.0d0) uni=seed/67108864. return end double precision function iuni(seed) c integer version of uni: RNG on [0,67108864) c to check integer operations using d.p. f.p. numbers c values taken from FPS AP library routine VRAND double precision seed double precision dmod seed=dmod(67081293.0d0*seed+14181771.0d0,67108864.0d0) iuni=seed return end c ***** c simmod.f - OFDM [de]modulation c ***** subroutine encode(d,sig,if1,if2,n) c encode logical data into data values c d - logical data c sig - generated data values, unit variance c if1 - index of lowest frequency c if2 - index of highest frequency c n - total number of values integer if1, if2, i, j, n logical d(n) real sig(n) c zero out low frequency terms </pre>
--	--

simmod.f

```

do 1 i=1,if1-1
  sig(i)=0.
  continue
1
c encode data into QAM (complex) format
  j=1
do 2 i=if1,if2
  if(d(j))then
    sig(i)=1.
  else
    sig(i)=-1.
  endif
  j=j+1
2  continue

c zero out high-frequency terms
do 3 i=if2+1,n
  sig(i)=0.
  continue
3  return
   end

subroutine decode(d,sig,if1,if2,n)
c recover logical data from signal values
c data not in [if1,if2] is unchanged
integer n, if1, if2, i, j
logical d(n)
real sig(n)
j=1
do 1 i=if1,if2
  if(sig(i).gt.0)then
    d(j)=.true.
  else
    d(j)=.false.
  endif
  j=j+1
1  continue
   return
   end

subroutine modu(sig,n,nf,serial)
c OFDM modulation
c sig - generated signal samples, unit variance
c n - OFDM block size = number of samples generated
c nf - number of data values = +/- 1 (rest assumed equal 0)
c serial - true for serial modulation (no FFTs)
integer n, nf
logical serial
real sig(n)
real k
real sqrt, float
real rms
c if OFDM (not serial) modulate the data values into an OFDM signal
c with unity power (variance = std. dev. = 1 V**2)
if(.not.serial)then
  call ffsn(sig,n)
  k=sqrt(float(n)/nf) * sqrt(float(n)/2.)
  call vsmul(sig,n,k)
  write(*,*) ' modu: output rms signal level = ',rmsv(sig,n)
endif
return
end

subroutine demodu(sig,n,nf,serial)
c OFDM demodulation
c sig - generated signal samples, unit variance
c n - OFDM block size = number of samples generated
c nf - number of data values = +/- 1 (rest assumed equal 0)
c serial - true for serial modulation (no FFTs)
integer n, nf
logical serial
real sig(n)
real k
real sqrt, float
real rms
c if OFDM, un-scale values and demodulate the ofdm signal samples
c back into data values.
if(.not.serial)then
  write(*,*) ' demodu: input rms signal level = ',rmsv(sig,n)
  k=sqrt(float(n)/nf) * sqrt(float(n)/2.)
  call vsmul(sig,n,1./k)

```

<pre> call ffan(sig,n) endif return end </pre>	<pre> c s2min is "npdb" times the minimum SNR value in the table c s2max is "npdb" times the maximum SNR value in the table integer mxint parameter (mxint=100) real b, w logical fm, rfm real rms, peak, pk2rms real agclim, sqlim integer nintn, nints real intrn(mxint), intnn(mxint), intrs(mxint), intss(mxint) integer i, j, i1, i2 real pi, n, a, r, s2asum, e2nr, nk, sk, c, nrfm, fd integer npdb, s2min, s2max real snr2s(1101), snr2n(1101) integer ix logical fading, noise real alog, float, exp, sqrt real dbtor, x common /s2com/ npdb, s2min, s2max, snr2s, snr2n data pi /3.14159/ c function to address 1-base arrays with negative indexes ix(i)=i-s2min+1 c function to convert db to linear (voltage) units dbtor(x)=10.**(x/20.) npdb=10 s2min=-500 s2max=600 write(*,*)' FM, IF & AF bandwidths = ', fm, b, w write(*,*)' rms, peak deviation = ', rms, peak write(*,*)' random FM, doppler rate = ', rfm, fd write(*,*)' AGC & squelch limits = ', agclim, sqlim write(*,*)' fading, noise = ', fading, noise </pre>
<pre> c ***** c simsnr.f - IF/AF SNR conversion c ***** c modified 89-3-9 to do agc and squelch after corrections. c c subroutine s2init(b,w,rms,peak,fm,fd,rfm,agclim,sqlim, + nints,intsr,intss, nintn,intnr,intnn, + fading, noise) c initialize IF SNR -to- AF SNR conversion tables c input parameters: c fm true for an FM channel false for SSB c b IF bandwidth (Hz or kHz) c w baseband bandwidth (same units as "b") c rms rms level of the modulating signal c peak peak level of the modulating signal c peak-to-average ratio = (10 for voice, about 2-3 for data) c pk2rms peak/rms c fd doppler rate (same units as "b" and "w") c rfm true to add random fm noise c agclim SNR above which the AGC operates (dB) c sqlim SNR below which the receiver output is squelched (dB) c (applied after interpolation given by intXX) c nintn - number of elements in intrn and intnn c nints - number of elements in intrs and intss c intrn - SNR part of points to use to change r-n table c intnn - "n" part of points to use to change r-n table c intrs - SNR part of points to use to change r-s table c intss - "s" part of points to use to change r-s table c fading - false to disable fading (constant signal output) c noise - false to disable noise (no noise added) c mxint - number of elements in intrn, intnn, intrs, intsn (not req'd) c c the arrays snr2s and snr2n contain the SNR-to-signal level c and SNR-to-noise level lookup tables. the look-up function c is: s2min+1 + ifix(SNR)*npdb c c npdb is number of table entries per dB of SNR change </pre>	<pre> c check number points in interpolation tables if(nints.lt.0).or. (nints.ge.1) .or. + (mod(nints,2).ne.0)) then </pre>

simsnr.f

<pre> write(*,*) ' % s2init - warning : bad nints ' endif if((nintn.lt.0) .or. (nintn.eq.1) .or. + (mod(nintn,2).ne.0)) then write(*,*) ' % s2init - warning : bad nintn ' endif c check for random FM being in valid approximation region if(rfm.and.(fd.gt.0.1*w))then write(*,*) ' s2init : doppler > .1 AF B/W ' stop endif if(rms.le.0.)then write(*,*) ' rms <= 0. ' stop endif pk2rms = peak/rms if((pk2rms).lt.1.)then write(*,*) ' s2init : (pk/rms) < 1 ' stop endif if(fm.and.(w.ge.b/2.0))then write(*,*) ' % s2init : warning : AF B/W >= 1/2 IF B/W ' endif c modified to give snr2s=1 maximum (for FM) if(fm)then a=s2asum(b,w) nk=8.*pi*b*w sk=pi**2/(pk2rms**2)*(b-2.*w)**2 if(rfm)then nrfm=2.*pi**2*log(10.)*fd**2 else nrfm=0. endif do 1 i=s2min,s2max r=10.0**(float(i)/(npdb*10)) e2mr=exp(-r) n=a*(1.0-e2mr)**2/r + nk*e2mr/sqrt(2*(r+2.35)) + nrfm c (sk factor scales snr2n to make snr2s independent of b & w) </pre>	<pre> snr2n(ix(i))=sqrt(n/sk) snr2s(ix(i))=(1.-e2mr) 1 continue else do 2 i=s2min,s2max snr2s(ix(i))=10.**float(i)/(npdb*20)) snr2n(ix(i))=1. 2 continue endif c apply corrections to conversion tables using linear interpolation c algorithm - c for each pair of points : c find indices in snr table for this pair c for all table entries between these points : c if the point is in the table : c interpolate and substitute new table value c corrections to snr2s do 9 i=1,nints-1,2 write (*, '(a,2f10.3,a,2f10.3)') ' % snr2s corr'n : ', c + intrs(i),intss(i),' to ',intrs(i+1),intss(i+1) i1=intsr(i)*npdb i2=intsr(i+1)*npdb do 8 j=i1,i2 if((j.ge.s2min) .and. (j.le.s2max))then snr2s(ix(j))=dbtor(+ interp(intrs(i),intrs(i+1),intss(i),intss(i+1), + float(j)/npdb)) endif 8 continue 9 continue c corrections to snr2n do 11 i=1,nintn-1,2 write (*, '(a,2f10.3,a,2f10.3)') ' % snr2n corr'n : ', c + intrn(i),intnn(i),' to ',intrn(i+1),intnn(i+1) i1=intnr(i)*npdb i2=intnr(i+1)*npdb do 10 j=i1,i2 if((j.ge.s2min) .and. (j.le.s2max))then snr2n(ix(j))=dbtor(+ interp(intrn(i),intrn(i+1),intnn(i),intnn(i+1), + float(j)/npdb)) endif 10 continue </pre>
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```

11 continue
c apply AGC for snrs from agclim up
  j=agclim*npdb
  if(j.lt.s2min)j=s2min
  if(j.gt.s2max)j=s2max
  c=1./snr2s(ix(j))
  do 3 i=j,s2max
    snr2n(ix(i))=snr2n(ix(i))/snr2s(ix(i))
    snr2s(ix(i))=1.0
  3 continue
c scale transfer curves below agc limit to make them continuous
  do 4 i=s2min,j-1
    snr2s(ix(i))=snr2s(ix(i))*c
    snr2n(ix(i))=snr2n(ix(i))*c
  4 continue
c squeelch for snrs below sqlim
  j=sqlim*npdb
  if(j.gt.s2max)j=s2max
  do 5 i=s2min,j
    snr2n(ix(i))=0.
    snr2s(ix(i))=0.
  5 continue
c if no fading
  if(.not.fading)then
    call vfill(snr2s, s2max-s2min+1, 1.)
  endif
c if no noise
  if(.not.noise)then
    call vfill(snr2n, s2max-s2min+1, 0.)
  endif
  return
  end
  real function s2asum(b,w)
c function to calculate a sum for calculating noise power
c for FM, see Jakes, Ch. 4.
  real w, b, s, pi, t, t1
  parameter (pi=3.14159)
integer n, nmax
parameter (nmax=12)
t1=(-pi)*(w/b)**2
t=1.
s=0.
do 1 n=0,nmax
  s=s+t/(n+n+3)
  t=t*t1/(n+1)
  1 continue
s2asum=s*4.*pi*pi*w*w/w/b
return
end
subroutine ch(sin,sout,noisev,fadev,ns,rms,peak)
c channel simulation
c sin/sout - input/output signal samples
c noisev - additive noise samples, unit variance
c fadev - signal snr levels (in dB snr)
c - any value > 1E6 (dB) indicates a blanked sample
c rms rms level of the modulating signal
c peak peak level of the modulating signal
integer ns
real sin(ns), sout(ns), noisev(ns), fadev(ns), rms, peak
integer i, j
integer ifix
integer npdb, s2min, s2max
real snr2s(1101), snr2n(1101)
real s, smax, smin
real rmsv
common /s2com/ npdb, s2min, s2max, snr2s, snr2n
c function to address 1-base arrays with negative indexes
integer ix
ix(i)=i-s2min+1
c write(*,*) ' ch:sin:rms: ', rmsv(sin,ns)
c write(*,*) ' ch:noisev:rms: ', rmsv(noisev,ns)

```

<pre> c compute maximum and minimum signal levels smax = peak/rms smin = -peak/rms c multiply by fading and add noise do 1 i=1,ns c check for blanking if(fadev(i).ge.1.0e6)then sout(i)=0.0 else j=ifix(fadev(i)*npdb) if(j.lt.s2min)j=s2min if(j.gt.s2max)j=s2max j=ix(j) c clip signal (at transmitter - before fading) s = sin(i) if(s.gt.smax) s = smax if(s.lt.smin) s = smin sout(i) = s*snr2s(j) + noisev(i)*snr2n(j) endif 1 continue return end subroutine r2sns(r,s,n,ns,snr) c convert signal envelope input vector to s and n vectors c used for numerical integration routine c r - received signal envelope level (dB) (0 dB mean) c s - corresponding signal scale value (linear) c n - corresponding noise scale value (linear) c ns - number of samples in r, s, n c snr - average snr integer ns real r(ns), s(ns), n(ns) real snr </pre>	<pre> integer i, j integer ifix integer npdb, s2min, s2max real snr2s(1101), snr2n(1101) common /s2com/ npdb, s2min, s2max, snr2s, snr2n c function to index 1-base arrays with negative indices integer ix ix(i)=i-s2min+1 do 1 i=1,ns j=ifix((r(i)+snr)*npdb) if(j.lt.s2min)j=s2min if(j.gt.s2max)j=s2max j=ix(j) s(i)=snr2s(j) n(i)=snr2n(j) c write(*,*) ' r2j: ', j 1 continue return end real function interp(x1,x2,y1,y2,x) c linear interpolation for value x using line between x1,y1 and x2,y2 c for x1=x2, returns y2 real x1, x2, y1, y2, x if(x2.ne.x1)then interp = y1 + (y2-y1)/(x2-x1) * (x-x1) else interp = y2 endif return end </pre>
---	--

simfec.f

```

c *****
c simfec.f - test and/or correct a block of data
c *****

      subroutine fec(data,okdata,n,intlv,ecn,ect,ne,nbx,nwe,nwx)

c number of remaining bit errors is returned in ne
c bits that don't fit into FEC blocks are ignored for BKER counts
c and are not processed for FEC correction
c no bits corrected if number of errors in a block > ect
c all bits corrected if number of errors in block <= ect
c (this is an idealized FEC code)
c for efficiency make ect as large as possible

c data - input/output data
c okdata - correct (transmitted) data
c n - number of values in data, okdata
c intlv - true to do interleaving
c ect - FEC block size
c ne - maximum number of correctable errors (0 for no FEC)
c nbx - number of bit errors remaining
c nbe - number of bits examined
c nwe - number of word errors remaining
c nwx - number of words examined

      integer n
      logical data(n), okdata(n)
      integer ne, nwe
      integer nbx, nwx
      logical intlv
      integer ect, nect

      integer i, k
      integer nbe
      integer mod

      if(ect.le.0)then
        write(*,*) ' % fec : ect <= 0 '
        stop
      endif

      if(ect.lt.0.or.ect.gt.ect)then
        write(*,*) ' fec : correctable errors (ect) out of range '
        stop
      endif

c interleave the received data and the correct (transmitted) data

```

```

      if(intlv)then
        call scr(data ,n,.true.)
        call scr(okdata,n,.true.)
      endif

c initialize number of bit errors and bits examined
      ne=0
      nbx=0

c count (and optionally correct) word errors
      nwe=0
      nwx=0

      i=1
      continue
      if(i+ect-1.gt.n)goto 2

c find number of errors in this word
      k=nbe(data(i),okdata(i),ect)
      ne=ne+k
      nbx=nbx+ect

c if any errors
c if they are correctable, correct them
c else increment word error count
      if(k.eq.0)then
        elseif(k.le.ect)then
          call vcopy(okdata(i),data(i),ect)
        else
          nwe=nwe+1
        endif
        nwx=nwx+1

      i=i+ect
      goto 1
    2 continue

c count remaining bit errors
      if(i.le.n)then
        ne=ne+nbe(data(i),okdata(i),n-i+1)
        nbx=nbx + n-i+1
      endif

```

```

c un-interleave data and correct data
  if(intlv)then
    call scr(data, n,.false.)
    call scr(okdata,n,.false.)
  endif
  return
end

integer function nbe(data,okdata,n)

c return number of differences between data and okdata

c data - input/output data
c okdata - correct data
c n - number of values in data, okdata (FEC block size)
c ne - local error counter

integer i, n, ne
logical data(n), okdata(n)

ne=0

do 1 i=1,n
  if(data(i).neqv.okdata(i))ne=ne+1
  continue
nbe=ne
return
end

subroutine scr(in,n,fwd)

c interleave (scramble) n in(put) data bits
c interleaving factor is sqrt(n)
c in direction fwd

integer i, j, k, l, n
logical in(n)
logical fwd

integer int
real sqrt, float

c common work vector
logical out(4096)
common out

k=int(sqrt(float(n))+0.5)
l=1
do 1 i=1,k
  do 1 j=i,n,k
    if(fwd)then
      out(l)=in(j)
    else
      out(j)=in(l)
    endif
    l=l+1
  continue
1

do 2 i=1,n
  in(i)=out(i)
2
return
end

subroutine dne(data,okdata,n,intlv,ecn,nofdm,snr)

c display number of remaining bit errors

c data - input/output data
c okdata - correct (transmitted) data
c n - number of values in data, okdata
c intlv - true to do interleaving
c ecn - FEC block size
c nofdm - OFDM block size (for printing only)
c snr - RF SNR (for printing only)

integer n
logical data(n), okdata(n)
logical intlv
integer ecn
integer nofdm
real snr

integer i, j, k
integer nwoerr
integer nbe
integer mod

if(ecn.le.0)then
  write(*,*) '% fec : ecn <= 0 '
  stop
endif

c interleave the received data and the correct (transmitted) data

```


<pre> if(intlv)then call scr(data ,n,.true.) call scr(okdata,n,.true.) endif i=1 continue if(i+ecn-1.gt.n)goto 2 1 c find number of errors in this word k=nbe(data(i),okdata(i),ecn) c print a flag, block size, snr, number of errors in word write(*,'(1X,A2,I5,F4.0,I5)' '%N', nofcm, snr, k i=i+ecn goto 1 continue 2 c compute and print error-free run lengths for the block nwoerr=0 do 3 i=1,n if(data(i).negv.okdata(i))then call dnew(nwoerr) nwoerr=0 else nwoerr=nwoerr+1 endif continue call dnew(nwoerr) 3 c terminate the block with a -1 call dnew(-1) c un-interleave data and correct data if(intlv)then call scr(data, n,.false.) call scr(okdata,n,.false.) endif return end subroutine dnew(nwoerr) integer nwoerr </pre>	<pre> write(*,'(1X,A2,I5)' '%R', nwoerr) return end </pre> <div style="border: 1px solid black; padding: 5px; text-align: center; margin: 10px 0;"> simut.f </div> <pre> c ***** c simut.f - Simulation Utility Routines c ***** real function vsum(x,n) integer n real x(n) integer i real sum sum=0. do 1 i=1,n sum=sum+x(i) continue 1 vsum=sum return end subroutine vmul(a,b,c,n) integer n real a(n), b(n), c(n) integer i do 1 i=1,n c(i)=a(i)*b(i) continue 1 return end subroutine vlcopy(in,out,n) c copy in into out integer i, n logical in(n), out(n) </pre>
--	--

<pre> do 1 i=1,n out(i)=in(i) continue return end subroutine vxor(in1,in2,out,n) c exclusive-or of two logical vectors integer i, n logical in1(n), in2(n), out(n) do 1 i=1,n out(i)=in1(i).negv.in2(i) continue return end subroutine vdbp(x,n) c convert vector x to dB (power) integer i, n real x(n) real alog10 do 1 i=1,n if(x(i).le.0.)then write(*,*) ' vdbp: argument <= 0 result set to -100 ' x(i)=-100. else x(i)=10.*alog10(x(i)) endif continue return end real function rmsv(x,n) c root mean square of a vector integer n real x(n) real sqrt, ssq if(n.le.0)then write(*,*) ' rms : n <= 0 ' </pre>	<pre> stop endif rmsv=sqrt(ssq(x,n)/n) return end real function ssg(x,n) c sum of squares of elements of a vector integer n real x(n), p integer i p=0. do 1 i=1,n p=p+x(i)**2 continue ssq=p return end subroutine vsq(x,y,n) c vector square y=x**2 for all n elements integer i, n real x(n), y(n) do 1 i=1,n y(i)=x(i)**2 continue return end subroutine vquant(x,n,max,k) c quantize all n elements of x to k bits. c if k = 0 no quantization is done c values are assumed to lie between +/- max c the range +/- max is divided into 2**k equal regions c all values within a region are converted to the mean of the region integer n, k real x(n), max real c1, c2 integer i integer nint if(k.ne.0)then </pre>
--	--

```

if(k.lt.0)then
  write(*,*) ' vquant : number of bits < 0 '
  stop
endif
if(max.le.0)then
  write(*,*) ' vquant : max < or = 0 '
  stop
endif
c1=2**(k-1)/max
c2=1./c1
do 1 i=1,n
  if(x(i).gt.max)then
    x(i)=max
  elseif(x(i).lt.-max)then
    x(i)=-max
  else
    x(i) = c2 * ( nint( x(i)*c1 +0.5 ) - 0.5)
  endif
  continue
1
endif
return
end
subroutine vifill(k,n,v)
c fill all n elements of integer vector k with value v
integer i, n, k(n), v
do 1 i=1,n
  k(i)=v
return
end
subroutine vfill(x,n,v)
c fill all n elements of real vector k with value v
integer i, n
real x(n), v
do 1 i=1,n
  x(i)=v
return
end
subroutine vcopy(x1,x2,ns)
integer ns
real x1(ns),x2(ns)
integer i
do 1 i=1,ns
  x2(i)=x1(i)
continue
return
end
subroutine vadd(x,n,a)
c add a to each element of x
integer n
real x(n), a
integer i
do 1 i=1,n
  x(i)=x(i)+a
continue
return
end
subroutine vsmul(x,n,a)
c multiply each element of x by a
integer n
real x(n), a
integer i
do 1 i=1,n
  x(i)=x(i)*a
continue
return
end
c statistics routines:
c initialized statistics variables
subroutine stinit(x,x2,n)
c x - sum of a's

```

```

c x2 - sum of a**2's
c n  - number of observations

real x, x2
integer n

x=0.
x2=0.
n=0

return
end

c update statistics variables

subroutine stat(a,x,x2,n)

c a  - observed value
c x  - sum of a's
c x2 - sum of a**2's
c n  - number of observations

real a, x, x2
integer n

x=x+a
x2=x2+a**2
n=n+1

return
end

c mean

real function stmn(x,x2,n)

real x, x2
integer n

if(n.le.0)then
  write(*,*) ' stmn : n <= 0 '
  stop
endif

c add 0 * x2 to avoid compiler warnings

stmn=x/n + 0.0*x2

return
end

c sample variance

real function stvr(x,x2,n)

real x, x2
integer n

if(n.le.1)then
  stvr=0.
else
  stvr=x2/(n-1)-x**2/(n*(n-1))
endif

return
end

c lower .95 CI

real function stl95(x,x2,n)

real stmn, stvr, t95, sqrt
real x, x2, t
integer n

if(n.le.0)then
  write(*,*) ' stl95 : n <= 0 '
  stop
endif

t=stvr(x,x2,n)/n
if(t.lt.0)then
  write(*,*) ' stl95: negative variance, set to 0.'
  t=0.
endif

stl95=stmn(x,x2,n)-t95(n)*sqrt(t)

return
end

c upper .95 CI

real function stur95(x,x2,n)

real stmn, stvr, t95, sqrt
real x, x2, t
integer n

if(n.le.0)then
  write(*,*) ' stur95 : n <= 0 '

```

<pre> stop endif t=stvr(x,x2,n)/n if(t.lt.0)then write(*,*)' stu95: negative variance, set to 0.' t=0. endif stu95=stmm(x,x2,n)+t95(n)*sqrt(t) return end real function t95(n) c t-table for 0.95 confidence interval c n is number of trials (degrees of freedom plus 1) c alpha = 0.025 c (rounded to 3 sig. digits) integer n real ttab(30) data ttab/ 0., 12.70, 4.30, 3.18, 2.78, 1 2.57, 2.45, 2.37, 2.31, 2.26, 2 2.23, 2.20, 2.18, 2.16, 2.15, 3 2.13, 2.12, 2.11, 2.10, 2.09, 4 2.08, 2.07, 2.07, 2.06, 2.06, 5 2.06, 2.05, 2.05, 2.05, 2.04 / if(n.lt.1)then write(*,*)'t95: too few trials : ',n elseif(n.le.30)then t95=ttab(n) elseif(n.le.40)then t95=2.03 elseif(n.le.60)then t95=2.01 else t95=2.0 endif return end c single precision complementary error function c Ed Casas - UBC Electrical Engineering c Feb. 21, 1986 c ref.: W. J. Cody, "Rational Chebyshev Approximation for the c Error Function," Mathematics of Computation, 23(107), pp. 631-638, </pre>	<pre> c 1969. c these functions should be accurate to the limit of single c precision operations. real function erfc(x0) logical neg real x0, x, x2, x3, x4, y real abs, erf, exp if(x0.lt.0.)then neg=.true. else neg=.false. endif x=abs(x0) if(x.le.0.5)then evaluate indirectly y=1.0 - erf(x) elseif(x.ge.4.)then approximation 3 x2=1.0/(x*x) x4=x2*x2 y = exp(-x*x)/x * (0.5641896 + x2 * 1 (-4.257996e-2 -1.960690e-1*x2 -5.168823e-2*x4) / 2 (1.509421e-1 + 9.214524e-1*x2 + 1.000000e00*x4)) else approximation 2 x2=x *x x3=x2*x x4=x3*x y = exp(-x2) * 1 (7.373888e00 + 6.865018e00*x + 3.031799e00*x2 2 + 5.631696e-1*x3 + 4.318779e-5*x4) / 3 (7.373961e00 + 1.518491e01*x + 1.279553e01*x2 4 + 5.354217e00*x3 + 1.000000e00*x4) endif if(neg)then erfc=2.0-y </pre>
--	--

<pre> else erfc=y endif return end c single precision error function real function erf(x0) logical neg real x0, x, x2, x4, y real abs, erfc if(x0.lt.0.)then neg=.true. else neg=.false. endif x=abs(x0) if(x.le.0.5)then approximation 1 x2=x*x x4=x2*x2 y = x * (2.138533e1 + 1.722276e0*x2 + 3.166529e-1*x4) / 1 (1.895226e1 + 7.843746e0*x2 + 1.000000e0*x4) else evaluate indirectly y=1.0-erfc(x) endif if(neg)then erf=-y else erf=y endif return end subroutine vsel(a,b,x,z,n) c substitute b into a for x > z </pre>	<pre> integer n real a(n), b(n), x(n), z integer i do 1 i=1,n if(x(i).gt.z)then a(i)=b(i) endif continue 1 return end real function berblk(a,b,c) c theoretical BER of an OFDM block real a, b, c, d real erfc, sqrt d=2.0*(b-a**2+c) if(d.lt.-0.0001)then write(*,*)'berblk: b-a**2+c < 0 : ',d/2.0 stop endif c the case where a and d are close to zero is unstable if(d.le.0.)then if (a.gt.0) then berblk=0.0 else berblk=0.5 endif else berblk=0.5*erfc(a/sqrt(d)) endif return end double precision function dray(x,n) c Rayleigh CPDF : prob. that signal is x dB below mean c with n ideal selection diversity branches integer n real x </pre>
---	--

<pre> double precision dexp dray = (1.0d0 - dexp(-1.0d0*10.0**(x/10.0)))**n return end subroutine setif(fs,n,f1,if1,f2,if2) c (two real values per frequency, first pair has DC and fs components) integer n real fs, f1, f2 integer if1, if2 integer mod real float if(fs.le.0.)then write(*,*) ' setif : fs <= 0 ' stop endif c find array index corresponding to frequency f1 if1=2.*f1/fs*float(n)+1. c round if1 up to an odd number so that it points to the c real element of the first complex number c e.g. 1->1, 2->3 if1=if1-mod(if1,2)+1 c find array index corresponding to frequency f2 if2=2.*f2/fs*float(n)+1. c round if2 down to an even number so that it points to the c imaginary element of the last complex number c e.g. 2048->2048, 2047->2046 if2=if2-mod(if2,2) if(if1.le.0)then if1=1 endif if(if2.ge.n)then if2=n endif if(if1.gt.n)then </pre>	<pre> write(*,*) ' setif: f1 > N ' stop endif if(if2.lt.0)then write(*,*) ' setif: f2 < 0 ' stop endif if(if1.gt.if2)then write(*,*) ' setif: f1 > f2 ' stop endif return end c count number of 0's, 1's and runs subroutine runcnt(data,n,n0,n1,nr,r) integer n, n0, n1, nr integer i real tn0n1, avgr, varr, r logical data(n), prev c test for n > 1 if(n.le.1)then write(*,*) ' runcnt: n < 2 . ' stop endif c initialize if(data(1))then n0=0 n1=1 else n0=1 n1=0 endif nr=1 prev=data(1) c go through data and count up true, false, and changes do 1 i=2,n if(data(i))then n1=n1+1 </pre>
---	---

```

if(.not.prev)then
  nr=nr+1
  prev=.true.
endif
else
  n0=n0+1
  if(prev)then
    nr=nr+1
    prev=.false.
  endif
endif
1 continue

c compute normalized r.v. (hopefully distributed n(0,1))
if(n0.eq.0 .or. n1.eq.0)then
  r=0.
else
  tn0n1 = 2.0 * n0 * n1
  avgr = 1.0 + tn0n1/float(n)
  varr = tn0n1*(tn0n1-float(n))/(float(n)*float(n)*float(n-1))
  r= (float(nr) - avgr) / sqrt(varr)
endif
return
end

real function norct(data,n)

c generate normalized run count (easier to call than runcnt)

integer n, i1, i2, i3
logical data
real r

call runcnt(data,n,i1,i2,i3,r)
norct=r

return
end

subroutine vl2diff(in1,in2,min,out,nout)

c compare dibits (2-bit sequences) and generate a logical vector
c map of dibit differences (.true.=difference)

integer nin, nout
logical in1(nin), in2(nin), out(nout)
integer i, j

```

```

if(2*nout.ne.nin)then
  write(*,*) ' vl2diff : nin <> 2*nout. '
  stop
endif
j=1
do 1 i=1,nin,2
  out(j) = ( in1(i) .eqv. in2(i) )
  + .and. ( in1(i+1) .eqv. in2(i+1) )
  j=j+1
1 continue
return
end

real function dbp(x)

c convert a power to dB

real x
if(x.gt.0.)then
  dbp=10.*alog10(x)
else
  write(*,*) ' dbp : db(x) for x<0 set to -99 '
  dbp=-99.
endif
return
end

subroutine vddat(crsig,txdata,nf,f1,f2,n,snr)

integer nf, n
real crsig(nf)
logical txdata(nf)
real f1, f2, snr
integer i
real f, txmag, rxmag, txan, rxan
real atan2
integer mod
real a, b

if(mod(nf,2).ne.0)then
  write(*,*) ' vddat : nf not even. '
  stop
endif
do 1 i=1,nf,2

```



```

txmag=1.4142
if(txdata(i))then
  if(txdata(i+1))then
    txan=45.
  else
    txan=-45.
  endif
else
  if(txdata(i+1))then
    txan=135.
  else
    txan=-135.
  endif
endif
rxmag=sqrt(crsig(i)**2+crsig(i+1)**2)
rxan=atan2(crsig(i+1),crsig(i))*57.3
f=f1+i*(f2-f1)/float(nf)
if (txdata(i)) then
  a=1.0
else
  a=-1.0
endif
if (txdata(i+1)) then
  b=1.0
else
  b=-1.0
endif
write*,'(1X,A2,2F10.4)') '%D', a, crsig(i)
write*,'(1X,A2,2F10.4)') '%D', b, crsig(i+1)
+
write*,'(1X,A2,I5,F4.0,F6.0,2(F4.1,F6.0))')
'%A', n, snr, f, txmag, txan, rxmag, rxan
1 continue
return
end
subroutine dvsn(x,data,nx,fl,fs,n,snr,nav,svec,nvec)
c x - vector of received data values
c data - transmitted logical data values
c nx - number of elements in x and data
c nav - number of snr measurements (of x) to average
c n - OFDM block size (to print)
1 continue
return
end
subroutine dvsn(x,data,nx,n,snr,svec,nvec)
c same as dvsn but displays the average signal and noise power
c of several signal vectors over all frequencies.
c x - vector of received data values
c data - transmitted logical data values
c nx - number of elements in x and data
c n - OFDM block size (to print)
c snr - RF SNR (to print)
c nav - number of snr measurements (of x) averaged so far
c n2av - number of snr measurements to average
integer n, nx
logical data(nx)
real x(nx), svec(nx), nvec(nx)
real snr
integer nsn, n2av
real sums, sums2, sumn, sumn2
integer nsums, nsumn
real dbp, stmn
logical first
c snr - RF SNR (to print)
integer n, nav
logical data(nx)
real x(nx), svec(nx), nvec(nx)
real snr
integer nsn, n2av
real sums, sums2, sumn, sumn2
integer nsums, nsumn
real dbp, stmn
logical first

```

<pre> data first /.true./ if(first)then first=.false. call stinit(sums,sums2,nsums) call stinit(summ,summ2,nsumm) write(*,*) ' how many blocks to average ? ' read(*,*) n2av endif call vsnv(x,data,nx,svec,nvec,nx,nsn) call stat(svec(1),sums,sums2,nsums) call stat(nvec(1),summ,summ2,nsumm) n2av=n2av-1 if(n2av.le.0)then write(*, '(1X,A2,I5,F4.0,3F6.1)') + '%X', n, snr, + dbp(stmn(sums,sums2,nsums)), + dbp(stmn(summ,summ2,nsumm)), + dbp(stmn(sums,sums2,nsums)) - + dbp(stmn(summ,summ2,nsumm)) call stinit(sums,sums2,nsums) call stinit(summ,summ2,nsumm) write(*,*) ' how many blocks to average ? ', char(7) read(*,*) n2av endif return end subroutine vsnv(x,data,nx,s,n,nav,nsn) c computes vectors of mean square and variance of received data c values averaged over several values vector x, (to get signal c and noise powers as a function of frequency). c x - vector of received data values c data - transmitted logical data values c nx - number of elements in tx and rx c s - square of mean of received data values c n - variance of received data values c nav - (maximum) number of elements of x to average c nsn - number of elements in s and n integer nx, nav, nsn logical data(nx) </pre>	<pre> real x(nx) integer mxnsn parameter (mxnsn=100) real s(mxnsn), n(mxnsn) real sp, np integer i, k, nleft if(nx.le.0)then write(*,*) ' vsn : nx <= 0 .' stop endif i=1 nsn=1 nleft=nx continue k=min(nleft,nav) call vsn(x(i),data(i),k,sp,np) s(nsn)=sp n(nsn)=np i=i+nav nsn=nsn+1 nleft=nleft-nav if(nleft.gt.0)goto 1 nnsn=nsn-1 return end subroutine vsn(x,data,nx,s,n) c computes mean square and variance of received data values in c a vector x, (to get signal and noise powers) by using negatives c of values in x whose corresponding element in data are 'false'. c x - vector of received data values c data - transmitted logical data values c nx - number of elements in tx and rx c s - square of mean of received data values c n - variance of received data values integer nx logical data(nx) real x(nx), s, n real ts, fs, ts2, fs2, tsp, tnp, fsp, fnp integer i, nt, nf </pre>
---	--

```

if(nx.le.0)then
  write(*,*) ' vsn : nx <= 0 .'
  stop
endif
ts = 0.
ts2=0.
fs = 0.
fs2=0.
nt=0
nf=0
do 1 i=1,nx
  if(data(i))then
    ts =ts + x(i)
    ts2=ts2 + x(i)**2
    nt=nt+1
  else
    fs =fs + x(i)
    fs2=fs2 + x(i)**2
    nf=nf+1
  endif
endif
continue
1
tsp = ( ts / nt )**2
fsp = ( fs / nf )**2
tnp = ts2 / nt - tsp
fnp = fs2 / nf - fsp
s = ( tsp + fsp ) / 2.
n = ( tnp + fnp ) / 2.
return
end

$LARGE
c *****
c simhw.f - hardware channel routines
c *****
c 87-11-19
      subroutine hwinit(data,prev,dev,tmp,neqbl,
+         ia,nmax,ns,fs,f1,f2,dbd,rms,peak,txemp,demp,
+         emp scl,nempsc,empfr,empsc)
c generate pre- and de-emphasis and channel correction vectors

```

simhw.f

```

c by measuring channel response
c data - temporary logical vector, length ns
c prev - generated pre-emphasis vector, length ns
c dev - generated de-emphasis vector, length ns
c tmp - temporary real vector, length ns
c neqbl - number of blocks to average in generating equalization
c vector
c ia - integer*2 sample vector, length nmax
c nmax - (OFDM) block size of channel probe signal and size of
c prev/dev
c ns - duration of channel probe signal (samples)
c fs - sampling rate
c f1 - lower frequency limit of channel to use
c f2 - upper frequency limit of channel to use
c dbd - dB/decade of pre-emphasis to use
c rms - rms voltage of output signal
c peak - peak voltage of output signal
c txemp - true to do phase/amplitude correction at transmitter
c devrms - rms value of de-emphasis vector (dev)
c demp - true to display equalization
c emp scl -
c nempsc -
c empfr -
c empsc -
integer ns, nmax, nempsc
logical data(nmax), txemp, demp, emp scl
real prev(nmax), dev(nmax), tmp(nmax)
real empfr(nempsc), empsc(nempsc)
integer*2 ia(ns)
integer i, neqbl, if1, if2, nf
real fs, f1, f2, dbd, rms, peak
real dbp, atan2
integer j
logical sr(23)
real devrms
real rmsv
data sr/
1 .false., .true., .true., .false., .false.,
1 .false., .true., .true., .true., .true.,
1 .false., .true., .false., .true., .false.,
1 .true., .false., .true., .false., .true.,
1 .false., .true., .false. /
c write(*,*) ' fs,nmax,f1,f2 = ', fs,nmax,f1,f2
c find indices for frequency limits

```

<pre> call setif(fs,nmax,f1,if1,f2,if2) nf=if2-if1+1 call vfill(tmp,nmax,0.) c prompt operator to turn off noise and fading pause ' turn noise and fading * OFF * ' c run prbs a few times to get rid of possible transients call prbs(data,nmax,sr) call prbs(data,nmax,sr) c compute and average "negbl" equalization vectors do 1 j=1,negbl c generate vector for pre-emphasis ("dbd" dB/decade) call empgen(prev(if1),nf,f2/f1,dbd) c generate random +/- 1 values in frequency skipping DC and fs/2 terms call prbs(data,nmax,sr) call encode(data,dev,if1,if2,nmax) call emp(dev,prev,f1,f2,fs,nmax,nmax) call vcopy(dev,prev,nmax) c modulate to time-domain call modu(prev,nmax,nf,.false.) c send it through the channel call hwch(prev,dev,ia,nmax,ns,rms,peak) c recover the (frequency-domain) channel output call demodu(dev,nmax,nf,.false.) c regenerate the (frequency-domain) input (before pre-emphasis) call encode(data,prev,if1,if2,nmax) c complex divide input by output (in frequency domain) to generate c correction (de-emphasis) vector </pre>	<pre> call cdiv(prev(if1),dev(if1),dev(if1),nf) c accumulate this equalization vector call vvadd(dev,tmp,tmp,nmax) 1 continue c prompt operator to turn noise and fading back on pause ' turn noise and fading * ON * ' c scale by number of equalization vectors averaged call vsmul(tmp,nmax,1./float(negbl)) c re-generate pre-emphasis vector (no dynamic storage, *%!&%%) call empgen(prev(if1),nf,f2/f1,dbd) c if desired, print initial de-emphasis vector c *** debug *** c write(*,*) ' if1, if2 = ', if1, if2 if(demp)then do 2 i=if1,if2,50 write(*,'(1X,A3,F6.0,F5.1,F8.0)') '%E ', + float(i)/nmax*(fs/2), + dbp(dev(i)**2+dev(i+1)**2), + atan2(dev(i),dev(i+1))*57.3 2 continue endif c if phase/magnitude correction is done at transmitter, c swap emphasis vectors if(txemp)then c normalize the old de-emphasis vector to (complex) magnitude of 1 and c copy to pre-emphasis vector devrms=rmsv(dev(if1),nf)*sqrt(2.0) call vsmul(dev(if1),nf,1.0/devrms) call vcopy(dev,prev,nmax) c generate a de-emphasis vector with appropriate magnitude call empgen(dev(if1),nf,f2/f1,dbd) </pre>
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```

        call vsmul(dev,nmax,devrms)
    endif
    if(empsc1)then
        call empscf(prev(if1),nf,f1,(f2-f1)/float(nf/2),
        + empsc,empfr,nempsc)
    endif
    return
end

subroutine hwch(x,y,ia,n,ns,rms,peak)

c do modem i/o (with guard samples)
c x - samples to be sent (variance = 1)
c y - received samples (nominal variance = 1)
c ia - integer*2 sample work vector
c n - number of elements in x and in y
c ns - total number of samples to generate
c rms - rms voltage of output signal
c peak - peak voltage of output signal
c peak must be less than 2.5 (Volts) (maximum DAC output)
c
integer i, j, k, n, nov, over, istart, iend
real peak, rms
real x(n), y(n), z, k1, k2
integer*2 ia(ns)
integer int
integer io
real float

c voltage-to-DAC and ADC-to-voltage conversion factors.
c values are for a 10-bit ADC and 12-bit DAC

real dacsc1, adcscl
parameter ( dacsc1 = 4096./5. )
parameter ( adcscl = 5./1024. )

c gain required to compensate for any (measured) filter loss

real gfilt
parameter ( gfilt = 1.11 )

c ( skip i/o for debugging )
c call vcopy(x,y(2),n-1)
c y(1)=x(n)
c return

c test peak value

```

```

    if(peak.le.0.)then
        write(*,*) ' hwch : peak <= 0. '
        stop
    endif
    if(peak.gt.(2.5/gfilt))then
        write(*,*) ' hwch : peak level too large. '
        stop
    endif

c test block size and number of samples

    if(n.lt.0)then
        write(*,*) ' hwch: n < 0 '
        stop
    endif
    if(ns.lt.n)then
        write(*,*) ' hwch: ns < n '
        stop
    endif

c factor to give unit-rms samples the rms value
    k1=rms

c center the n OFDM samples in the ns-sample output
    istart=ns/2-n/2+1
    iend=ns/2+n/2

c combine dacsc1 and gfilt
    k2=dacsc1*gfilt

c convert the f.p. samples to DAC levels
c limit peak level (limits peak deviation)
c zmin= 1.e30
c zmax=-1.e30
c zrms= 0.
    nov=0

    j=istart
    do 1 i=1,n
        z=x(i)*k1
        if(z.gt.peak)then
            z=peak

```

<pre> nov=nov+1 else if(z.lt.-peak)then z=-peak nov=nov+1 endif ia(j)=int(z * k2 + 0.5) c if(z.lt.zmin)zmin=z c if(z.gt.zmax)zmax=z c zrms=zrms+z**2 j=j+1 continue 1 write(*,*) ' % TX signal rms = ',sqrt(zrms/n) write(*,*) ' % TX signal min = ',zmin write(*,*) ' % TX signal max = ',zmax c if(nov.ne.0)then write(*,*) ' % ', 100.*float(nov)/n, ' % overflow. ' endif c add guard band before data j=i-1 i=iend i=iend if(j.lt.1)goto 3 ia(j)=ia(i) i=i-1 j=j-1 goto 2 continue 3 c add guard band after data j=iend+1 i=i-1 i=iend continue if(j.gt.ns)goto 5 ia(j)=ia(i) j=j+1 i=i+1 goto 4 continue 5 c do io and stop if overrun over=io(ia,ns) </pre>	<pre> if (over.ne.0) then write(*,*) ' hwch: A/D or D/A overrun ' stop endif c scale A/D samples back to FP j=i-1 do 6 i=1,n y(i)=float(ia(j)) * adcscl j=j+1 6 continue return end subroutine emp(x,y,f1,f2,fs,n,ns) c multiply vector x by vector y over the indices if1 to if2 to do c pre-emphasis or de-emphasis and correction for channel gain/phase c transfer function c x - vector to be corrected c y - correction vector c if1 - first element of x to correct c if2 - last element of x to correct c n - number of values in x c ns - number of values in y (multiple of n) integer i, j, j2, n, ns, skip integer if1, if2 real f1, f2, fs real a, b, c, d real x(n) real y(ns) integer mod if(mod(ns,n).ne.0)then write(*,*) ' emp: ns not multiple of n ' stop endif call setif(fs,n,f1,if1,f2,if2) call setif(fs,ns,f1,j,f2,j2) skip=2*ns/n do 1 i=if1,if2,2 a=y(j) </pre>
--	--

<pre> b=y(j+1) c=x(i) d=x(i+1) x(i) = a*c - b*d x(i+1) = a*d + b*c j=j+skip 1 continue c (debugging) c write(*,*) ' emphasis results: ' c write(*,*) ' n, ns = ', n, ns c write(*,*) ' if1, if2 = ', if1, if2 c write(*,*) ' j, skip = ', j, skip c write(*,*) ' results: i j a,b c,d result ' c900 write(*,900) i, j, a, b, c, d, x(i), x(i+1) format(1x,2i5,6f7.3) return end subroutine cdiv(x,y,z,n) divide two complex vectors (z=x/y) integer i, n real x(n), y(n), z(n) real a, b, c, d, r integer mod if(mod(n,2).ne.0)then write(*,*) ' cdiv: n not even ' stop endif do 1 i=1,n,2 c find mag. squared of y c=y(i) d=y(i+1) r=c*c+d*d if(r.ne.0.)then c set c,d = 1/y c= c/r d=-d/r multiply by x=a,b a=x(i) b=x(i+1) z(i) =a*c-b*d z(i+1)=a*d+b*c else write(*,*) ' % cdiv: complex divide by zero at i = ', i </pre>	<pre> z(i) =0. z(i+1)=0. endif continue return end subroutine empgen(x,n,f2byf1,dbd) generate (complex) pre-emphasis vector c x - emphasis vector to be generated c n - number of elements (even) c f2byf1 - ratio of highest to lowest frequency c dbd - number of dB per decade emphasis integer n, i real x(n), dbd, f2byf1 real decs, xn, k, Y, ss real alog10, sqrt, float integer mod if(mod(n,2).ne.0)then write(*,*) ' empgen : n not even ' stop endif c number of decades between frequency limits decs = alog10(f2byf1) c total increase (linear factor) xn = 10. ** (dbd * decs / 20.) c constant factor to obtain required increase k = 10. ** (alog10(xn) / (n/2-1)) c generate scaled vector and find total power y=1. ss=0. do 1 i=1,n,2 x(i) =y x(i+1)=0. ss=ss+y**2 y=y*k 1 continue </pre>
---	---

<pre> ss=sqrt(ss/float(n/2)) c scale to unity power call vsmul(x,n,1./ss) c (debugging) c write(*,*) ' f2byfl, n, dbd, k = ', f2byfl, n, dbd, k c write(*,*) ' ss = ', ss c write(*,*) ' empgen power = ',rmsv(x,n)*sqrt(2.) return end subroutine waitfor(prompt) character*(*) prompt character c close(0) write(0,*)prompt read(0,'(A1)')c return end subroutine vvadd(a,b,c,n) c c(i)=a(i)+b(i) for i=1 to n integer i, n real a(n), b(n), c(n) do 1 i=1,n c(i)=a(i)+b(i) 1 continue return end subroutine empscf(x,nx,fl,df,sc,fr,nsc) c subroutine to scale a pre- or de-emphasis vector using a c measured channel power transfer function or baseband SNR c characteristics. c x - emphasis vector to be scaled c nx - number of real elements in x (assumed as real/imag. pairs) c fl - starting frequency of values in x c df - frequency increment between values in x c sc - the response of the channel to be used to scale x c fr - the upper frequency limits for each scaling value in sc </pre>	<pre> c nsc - number of values in sc and fr integer nx, nsc real x(nx), sc(nsc), fr(nsc) real fl, df integer i, j real k, f, ss if(nsc.lt.1)then write(*,*) ' empscf : nsc < 1 .' stop endif c initialize j=1 f=fl ss=0.0 k=10.0**(-1.0*sc(1)/20.0) c scale the real part of every emphasis vector pair do 3 i=1,nx,2 c go on to next scaling value if necessary for current frequency 1 continue if((f.le.fr(j)) .or. (j.ge.nsc))goto 2 j=j+1 k=10.0**(-1.0*sc(j)/20.0) goto 1 2 continue c scale and sum squares x(i)=x(i)*k ss=ss+x(i)**2 f=f+df 3 continue ss=sqrt(ss/float(nx/2)) call vsmul(x,nx,1.0/ss) return end </pre>
--	---

simdum.f

```

c *****
c simdum.f - dummy hardware channel routines for non-PC systems
c *****
      subroutine hwinit(data,prev,dev,tmp,neqbl,
+       ia,ni,ns,fs,f1,f2,dbd,rms,peak,ttemp,demp)
      integer neqbl, ni, ns
      logical data(ni), ttemp, demp
      real prev(ni), dev(ni), tmp(ni)
      integer*2 ia(ns)
      real fs, f1, f2, dbd, rms, peak
      write(*,*) ' hwinit - dummy routine called '
      stop
      end
      subroutine hwch(x,y,ia,n,ns,rms,peak)
      integer n, ns
      real x(n), y(n), rms, peak
      integer*2 ia(n)
      write(*,*) ' hwch - dummy routine called '
      stop
      end
      subroutine emp(x,y,f1,f2,fs,n,ns)
      integer n, ns
      real f1, f2, fs
      real x(n), y(ns)
      write(*,*) ' emp - dummy routine called '
      stop
      end

```

siftt.dif (diffs from FAST.FOR)

```

0a1,5
> c warning !!!! : these routines have been modified to work
> c on arrays of size N instead of N+2
> c modified 87-8-14 to send all output to default output
> c
> c
1a7,9
> c SUBROUTINE: FFA
> c FAST FOURIER ANALYSIS SUBROUTINE
> C-----
3c11
<      SUBROUTINE FFA(B, NFFT)
<
<
<      SUBROUTINE FFA(B, NFFT)
37c45
<      WRITE (IW,9999)
<
<
<      WRITE (*,9999)
72,75c80,83
<      T = B(2)
<      B(2) = 0.
<      B(NFFT+1) = T
<      B(NFFT+2) = 0.
<
<
<      T = B(2)
> c      B(2) = 0.
> c      B(NFFT+1) = T
> c      B(NFFT+2) = 0.
88c96
<      SUBROUTINE FFS(B, NFFT)
<
<
<      SUBROUTINE FFSn(B, NFFT)
119c127
<      WRITE (IW,9999)
<
<
<      WRITE (*,9999)
123c131
<      B(2) = B(NFFT+1)
<
<
> c      B(2) = B(NFFT+1)
710c718
<      WRITE (IW,9999)
<
<
<      WRITE (*,9999)
941a950,1062
>
> c

```

```

> C -----
> C FUNCTION: ILMACH
> C THIS ROUTINE IS FROM THE PORT MATHEMATICAL SUBROUTINE LIBRARY
> C IT IS DESCRIBED IN THE BELL LABORATORIES COMPUTING SCIENCE
> C TECHNICAL REPORT #47 BY P.A. FOX, A.D. HALL AND N.J. SCHRYSER
> C -----
> C
> C INTEGER FUNCTION ILMACH(I)
> C
> C I/O UNIT NUMBERS.
> C
> C ILMACH( 1) = THE STANDARD INPUT UNIT.
> C
> C ILMACH( 2) = THE STANDARD OUTPUT UNIT.
> C
> C ILMACH( 3) = THE STANDARD PUNCH UNIT.
> C
> C ILMACH( 4) = THE STANDARD ERROR MESSAGE UNIT.
> C
> C WORDS.
> C
> C ILMACH( 5) = THE NUMBER OF BITS PER INTEGER STORAGE UNIT.
> C
> C ILMACH( 6) = THE NUMBER OF CHARACTERS PER INTEGER STORAGE UNIT.
> C
> C INTEGERS.
> C
> C ASSUME INTEGERS ARE REPRESENTED IN THE S-DIGIT, BASE-A FORM
> C
> C SIGN ( X(S-1)*A**(S-1) + ... + X(1)*A + X(0) )
> C WHERE 0 .LE. X(I) .LT. A FOR I=0,...,S-1.
> C
> C ILMACH( 7) = A, THE BASE.
> C
> C ILMACH( 8) = S, THE NUMBER OF BASE-A DIGITS.
> C
> C ILMACH( 9) = A**S - 1, THE LARGEST MAGNITUDE.
> C
> C FLOATING-POINT NUMBERS.
> C
> C ASSUME FLOATING-POINT NUMBERS ARE REPRESENTED IN THE T-DIGIT,
> C BASE-B FORM
> C
> C SIGN (B**E)*( X(1)/B + ... + (X(T)/B**T) )
> C WHERE 0 .LE. X(I) .LT. B FOR I=1,...,T,
> C 0 .LT. X(1), AND EMIN .LE. E .LE. EMAX.
> C
> C ILMACH(10) = B, THE BASE.
> C
> C SINGLE-PRECISION
> C
> C ILMACH(11) = T, THE NUMBER OF BASE-B DIGITS.
> C
> C ILMACH(12) = EMIN, THE SMALLEST EXPONENT E.
> C
> C ILMACH(13) = EMAX, THE LARGEST EXPONENT E.
> C
> C DOUBLE-PRECISION
> C
> C ILMACH(14) = T, THE NUMBER OF BASE-B DIGITS.
> C
> C ILMACH(15) = EMIN, THE SMALLEST EXPONENT E.
> C
> C ILMACH(16) = EMAX, THE LARGEST EXPONENT E.
> C
> C TO ALTER THIS FUNCTION FOR A PARTICULAR ENVIRONMENT,
> C THE DESIRED SET OF DATA STATEMENTS SHOULD BE ACTIVATED BY
> C REMOVING THE C FROM COLUMN 1. ALSO, THE VALUES OF
> C ILMACH(1) - ILMACH(4) SHOULD BE CHECKED FOR CONSISTENCY
> C WITH THE LOCAL OPERATING SYSTEM.
> C
> C INTEGER IMACH(16), OUTPUT
> C
> C EQUIVALENCE (IMACH(4),OUTPUT)
> C
> C MACHINE CONSTANTS FOR THE VAX-11 WITH
> C FORTRAN IV-PLUS COMPILER
> C
> C DATA IMACH( 1) / 5 /
> C DATA IMACH( 2) / 6 /
> C DATA IMACH( 3) / 5 /
> C DATA IMACH( 4) / 6 /
> C DATA IMACH( 5) / 32 /
> C DATA IMACH( 6) / 4 /
> C DATA IMACH( 7) / 2 /
> C DATA IMACH( 8) / 31 /
> C DATA IMACH( 9) / 2147483647 /
> C imach 9 not used by fftsn
> C data imach( 9) / 0 /
> C DATA IMACH(10) / 2 /
> C DATA IMACH(11) / 24 /
> C DATA IMACH(12) / -127 /
> C DATA IMACH(13) / 127 /
> C DATA IMACH(14) / 56 /
> C DATA IMACH(15) / -127 /
> C DATA IMACH(16) / 127 /
> C
> C IF ( I .LT. 1 .OR. I .GT. 16) GO TO 10

```

```

> C
> ILMACH=IMACH(I)
> RETURN
> C
> 10 WRITE(*,9000)
> 9000 FORMAT(39H1ERROR 1 IN ILMACH - I OUT OF BOUNDS)
> C
> STOP
> C
> END
>

```

fdint.f

```

$LARGE
c *****
c fdint.f - MC Integration
c *****
c Ed. Casas 86-2-19

c Monte-Carlo integration program to obtain OFDM BER.
c revised for variable r-sn characteristics 87-7-3
c added block error rate calculations 88-11-11

c -----
$INCLUDE: 'simpdef.f'
c -----

c local variables:
c counters
integer i, j, k, l

c workspace to generate fading waveforms
real fade(ns)

c vectors to store BER in one trial, sums of BERs, squares of BERs
c and number of trials
double precision berout, bkrount
real bert (mxblk,mxsnr), ber (mxblk,mxsnr), bers (mxblk,mxsnr)
real bkert (mxblk,mxsnr), bker (mxblk,mxsnr), bkern (mxblk,mxsnr)
real temp
integer ntrial(mxblk,mxsnr), nbktr(mxblk,mxsnr)

c vectors for received signal and noise values and their squares
real rs(ns), rn(ns), rs2(ns), rn2(ns)

```

```

c external utilities
real stmn, stu95, stl95

c initialize variance and BER sums (and sums of squares) to zero
data ber /mxblsn*0.0/, bers /mxblsn*0.0/

c -----
$INCLUDE: 'simpget.f'
c -----

c initialize snr tables
call s2init(b,w,rms,peak,fm,fd,rfm,agclim,sqlim,
+ nints,intsr,intss, ninthn,intnr,intcm,
+ fading, noising )

c do for each of ntr trials
do 50 l=1,ntr

c clear sum of BERs for this trial
do 60 j=1,nn
do 60 k=1,nsnr
bert (j,k)=0.d0
bkert(j,k)=0.d0
continue
60

c do appropriate number of passes
do 30 i=1,nblk

c generate a fading envelope with 0 dB mean
if(fading)then
call genfdb(fd/fs,fseed,fade,ns,
+ ndrbr,thr,ndw,nsw)
else
call vfill(fade,ns,0.)
endif
c do for each snr
do 30 k=1,nsnr

c convert received envelope level to r and s values and squares

```

<pre> call r2sns(fade,rs,rs2,rs,ns,snr(k)) call vsq(rs,rs2,ns) call vsq(rs,rs2,ns) c do for each block size do 30 j=1,nn c update variance and BER sums + call sum2(rs,rs2,rs2,rs2,ns,na(j),ecn,berout,bkrou, + snr(k),dnerr) + bert(j,k)=bert(j,k)+berout + bkert(j,k)=bkert(j,k)+bkrou c write(*,*) ' blk, BKER = ',i, bkrou 30 continue c update statistics counts at end of a trial do 70 j=1,nn do 70 k=1,nsnr temp=bert(j,k)/nblk call stat(temp,ber(j,k),bers(j,k),ntrial(j,k)) temp=bkert(j,k)/nblk call stat(temp,bker(j,k),bkers(j,k),nbktr(j,k)) c write(*,*) ' j,k,ber,ntrial=',j,k,ber(j,k),ntrial(j,k) 70 continue 50 continue c print results write(*,*) '% BER, -/+ .95 conf limits ' do 100 j=1,nn do 100 k=1,nsnr write(*, '(1X,I6,F6.1,3(1PE11.2))') na(j), snr(k), + stmn(ber(j,k),bers(j,k),ntrial(j,k)), + sti95(ber(j,k),bers(j,k),ntrial(j,k)), + stu95(ber(j,k),bers(j,k),ntrial(j,k)) 100 continue write(*,*) '% BKER, -/+ .95 conf limits ' do 200 j=1,nn do 200 k=1,nsnr </pre>	<pre> write(*, '(1X,I6,F6.1,3(1PE11.2))') na(j), snr(k), + stmn(bker(j,k),bkers(j,k),nbktr(j,k)), + sti95(bker(j,k),bkers(j,k),nbktr(j,k)), + stu95(bker(j,k),bkers(j,k),nbktr(j,k)) 200 continue end c----- Sum for one snr, len ----- subroutine sum2(rs,rs2,rs2,rs2,ns,len,ecn,berout,bkrou,snr,dnerr) c subroutine to calculate fade and snr statistics of a block and c corresponding BER c variables: c counters + integer i, j, k c length of received signal and noise vectors + integer ns c vectors for received signal and noise values and squares + real rs(ns), rs2(ns), rn2(ns) c OFDM block length, FEC block (word) length + integer len, ech c added to display block snr: snr and true to display block BER + real snr + logical dnerr c inverse of block length + real invlen c number of blocks tested + integer ntst, n c sums of BERS + double precision ber, berout, bker, bkrou, dtemp </pre>
--	--

<pre> c sums of s, s^2, n^2 real a, b, c, d c intrinsics integer mod c external utilities real berblk c find number of OFDM blocks included in the vector fade if(len.eq.0 .or. mod(ns,len).ne.0) then write(*,*)'sum2: bad OFDM block size.' stop endif n=ns/len c find inverse of block length invlen=1./len c start at the first sample of the input block i=1 ntst=0 ber= 0.0d0 bker=0.0d0 c do for each block do 10 j=1,n a=0. b=0. c=0. c get statistics for this block do 20 k=1,len a=a+rs(i) b=b+rs2(i) c=c+rn2(i) i=i+1 continue 20 a=a*invlen b=b*invlen </pre>	<pre> c=c*invlen write(*,*) ' a, b, b-a**2, c = ', a, b, b-a**2, c c add resulting block BER to the appropriate BER averaging sum c and compute Bker assuming independent errors dtemp=berblk(a,b,c) ber =ber +dtemp if (dnerr) write(*, '(IX,A2,I5,F4.0,E16.7)') '%B ', + len, snr, dtemp bker=bker+(1.0d0-(1.0d0-dtemp)**ecn) ntst=ntst+1 c write(*,*) ' BER = ', dtemp c write(*,*) ' Bker = ', (1.0d0-(1.0d0-dtemp)**ecn) 10 continue berout=ber /ntst bkrout=bker/ntst return end </pre>
<p>pint.f</p>	
<pre> c ***** c Integration for large/small N c ***** c pint.f c ed.casas 87-9-16 c ----- \$INCLUDE: 'simpdef.f' c ----- c local variables real r(2000), p(2000), s(2000), s2(2000), n(2000), wrk(2000) integer i, j, k real al, be, ga real sber, lber </pre>	<pre> c ***** c Integration for large/small N c ***** c pint.f c ed.casas 87-9-16 c ----- \$INCLUDE: 'simpdef.f' c ----- c local variables real r(2000), p(2000), s(2000), s2(2000), n(2000), wrk(2000) integer i, j, k real al, be, ga real sber, lber </pre>

<pre> real vsum, berblk real snrmin, snrinc integer nst parameter (nst=1100) snrmin=-50. snrinc=0.1 c ----- \$INCLUDE: 'simpget.f' c ----- c can only vary one threshold (agc OR squelch) at a time if (agcvar .and. sqvar) then write(*,*) ' pint : agcvar and sqvar true. ' stop endif c do at least one loop if (nthrsh .le. 0) then nthrsh=1 thrsh(1)=0. endif do 1 k=1,nthrsh if (.not. (agcvar .or. sqvar)) call + s2init(b,w,rms,peak,fd,rfm,agclim,sqlim, + nints,intsr,intss, nintn,intnr,intnn, + fading, noising) do 1 j=1,nsnr c initialize snr-to-signal and snr-to-noise tables c if testing effect of varying the agc limit if (agcvar) call + s2init(b,w,rms,peak,fd,rfm,snr(j)+thrsh(k),sqlim, + nints,intsr,intss, nintn,intnr,intnn, + fading, noising) c if testing effect of varying the squelch limit if (sqvar) call + s2init(b,w,rms,peak,fd,rfm,agclim,snr(j)+thrsh(k), </pre>	<pre> + nints,intsr,intss, nintn,intnr,intnn, + fading, noising) c initialize probability (Rayleigh) table call pgen(snrmin,snrinc,snr(j),ndbr,p,r,nst) c find signal and noise scaling signals from snr signal call r2sns(r,s,n,nst,0.) c compute s**2 and n**2 call vmul(s,s,s2,nst) call vmul(n,n,n,nst) c integrate to find averages of al(pha), be(ta), and ga(mma) call vmul(p,s,wrk,nst) al=vsum(wrk,nst) call vmul(p,s2,wrk,nst) be=vsum(wrk,nst) call vmul(p,n,wrk,nst) ga=vsum(wrk,nst) c use block BER equation to estimate long-block BER lber=berblk(al,be,ga) c find single-sample-block BERS and scale by the sample probability do 2 i=1,nst wrk(i)=berblk(s(i),s2(i),n(i))*p(i) write(*, '(IX,I10,F10.2,5E15.3) ',i,r(i), c & berblk(s(i),s2(i),n(i)),p(i),wrk(i) 2 continue c add up (average) to find short-block BER sber=vsum(wrk,nst) if (agcvar .or. sqvar) then write(*,*) snr(j), thrsh(k), sber, lber else write(*,*) snr(j), sber, lber endif </pre>
--	---

<pre> 1 continue end subroutine pgen(snrmin,snrinc,snravg,ndbr,p,r,n) generate tables of snrs and probabilities c input: c snrmin - minimum snr c snrinc - snr increment per step c snravg - average snr c ndbr - number of diversity branches c r - signal levels (snr) c p - probability of a given step c n - number of values in p integer n, ndbr real snrmin, snrinc, snravg real p(n), r(n) c local variables: c i - counter into p and r c sump - sum of probabilities (should add to 1) integer i real sump real snrk1, snrk2, snr c rayleigh CPDF double precision dray calculate signal level points and probabilities sump=0. snr=snrmin snrk1=snrinc/2.0+snravg snrk2=snrinc/2.0+snravg do 1 i=1,n c save the snr and probability between the two signal levels r(i) = snr p(i) = dray(snr+snrk1,ndbr) - dray(snr-snrk2,ndbr) c sum probabilities to check sump=sump+p(i) </pre>	<pre> c write(*,'(IX,I10,F10.2,2E15.3)') i, r(i), p(i), sump snr=snr+snrinc 1 continue if(abs(sump-1.0).gt.0.001)then write(*,*) 'pgen:total probability <1; = ',sump stop endif return end </pre>
<p>io.asm</p>	
<pre> COMMENT \$ This Microsoft FORTRAN-callable function reads/writes a block of samples from/to the analog interface board. Ed. Casas 87-10-19. The FORTRAN use is: NR=IO(IA,N) where: IA - INTEGER*2 (16-bit) array containing the D/A samples on entry and containing the A/D samples on return. If "convert" is not zero the samples are left justified in binary (unsigned) format. In this case the samples should be pre-/post- converted to two's-complement. N - INTEGER*4 number of values to be input and output. NR - INTEGER*4 number of samples *NOT* read/written. If this number is not zero, an over-run occurred. \$ IO_TEXT segment byte public 'CODE' assume cs:IO_TEXT ; timing constants MHZ equ 2 ; 8253 clock input frequency (MHz) PERIOD equ 125 ; sampling period, (us) (125 minimum) LEN equ 4 ; S/H sampling time (us) (4 maximum) </pre>	

```

; non-zero to convert between offset-binary and 2's complement
convert equ 1 ; set to 0 if get overrun errors on a slow PC
; hardware
IBMIO equ 300H ; I/O base address for IBM prototyping card
PIA equ IBMIO ; 8255 par. port: MS bit=overrun, LS=sampling
PIA0 equ PIA+0 ; PIA port A
PIA1 equ PIA+1 ; PIA port B
PIA2 equ PIA+2 ; PIA port C
PIA3 equ PIA+3 ; PIA control port
ADC equ PIA+4 ; NEC uPD7004 A/D converter
ADC0 equ ADC+0 ; channel select & LS 2 bits
ADC1 equ ADC+1 ; "initialize" & MS 8 bits
DAC equ PIA+8 ; National DAC1208 D/A converter
DAC0 equ DAC+0 ; LS 4 bits (load second)
DAC1 equ DAC+1 ; MS 8 *AND* LS 4 bits (load first)
CLK equ PIA+12 ; 8253 timer/counter
CLK0 equ CLK+0 ; counter 0
CLK1 equ CLK+1 ; counter 1
CLK2 equ CLK+2 ; counter 2
CLK3 equ CLK+3 ; mode register
IO proc far
public IO
; entry
push bp ; save bp
mov bp,sp ; save si
push si ; save flags (& interrupt status)
pushf
; disable interrupts
cli
; set up sample count
les bx, dword ptr [bp+6]
mov si, es:[bx] ; SI has LS word of sample count
mov di, es:[bx+2] ; DI has MS word of sample count
; negate sample count so can count up to zero
not si ; complement di:si
not di ; add 1
add si, 1
adc di, 0
; es:bx points into sample array
les bx, dword ptr [bp+10]
; dh retains high byte of I/O board address
mov dh, high PIA
; hardware initialization
; set up PIA
PIAr record moderset:1=1, Amod:2,Adir:1,Aptc:1, Bmod:1,Bdir:1,Bptc:1
mov dl, low PIA3
mov al, PIAr<1, 01,1,1, 0,1,1,>
out dx, al
; set timers first to stop them and prevent overrun error on
; first (unused) conversion
CLKr record counter:2, readload:2=3, clkmode:3, bcd:1=0
; start by setting timer 0 mode (and so stopping it)
mov dl, low CLK3
mov al, CLKr<0,,2,> ; 0 = MODE 2 (rate generator)
out dx, al
; set up timer 1
mov dl, low CLK3
mov al, CLKr<1,,1,> ; 1 = MODE 1 (one-shot)
out dx, al
mov dl, low CLK1 ; set S/H sample time
mov al, low (MHZ*LEN)
out dx, al
mov al, high (MHZ*LEN)
out dx, al
; set up ADC, start first (unused) conversion, and clear "sampling"
; and "overrun" latches

```



```

ADCr0 record channel:3=0          ; CH 0 input
ADCr1 record twoscomp:1=0, divider:2=1 ; binary, divide clock by 2

mov dl, low ADC1
mov al, ADCr1<>
out dx, al

mov dl, low ADC0
mov al, ADCr0<>
out dx, al

; finish setting up timer 0 :

; set sampling rate and start the timer, first S/H pulse is PERIOD us
; later. the first (initialization) conversion will have completed by
; then.

mov dl, low CLK0          ; set sampling rate
mov al, low (MHZ*PERIOD)
out dx, al
mov al, high (MHZ*PERIOD)
out dx, al

; end of hardware initialization

; *** critical timing within loop: do not change code ***

loop:
; load sample into DAC (to be transferred by next S/H pulse)

mov ax, es:[bx]
if convert                ; if converting
rept 4                    ; make sample left-aligned
shl ax, 1
endm
xor ax, 8000h             ; convert to offset-binary
endif

xchg al, ah              ; write MS byte, then LS
mov dl, low DAC1         ; write DAC1 and wrap around to DAC0
out dx, ax

; set up ADC control word in cl, and PIA address in dx

mov cl, ADCr0<>
mov dl, low PIA0

; wait for S/H to start sampling (implies conversion complete)

```

```

ll:      in  al, dx
        or  al, al          ; test for sampling or overrun
        jz  ll

; get A/D sample and start next conversion (sampling should be complete)

mov dl, low ADC0
in ax, dx                ; input A/D result
xchg ax, cx              ; sample to CX, get A/D control to AL
out dx, al                ; start A/D conversion

; exit if MS bit of PIA0 was set (overrun)

jl done

; replace output sample with input sample

if convert                ; if converting in loop
mov ax, cx
xor ax, 8000h            ; convert to 2's complement
rept 6
sar ax, 1                ; make right-justified
endm
mov es:[bx], ax

else
; if not converting, save sample as read

mov es:[bx], cx
endif

; point to next element in (possibly $LARGE) array

add bx, 2                ; increment pointer to next sample
jnc l2                   ; test for offset < 64k
mov ax, es                ; move segment up if not
add ax, 1000h
mov es, ax

l2:
; increment (negated) sample count

inc si
jnz l3
inc di

l3:
; loop 'til done

```

<pre> jnz loop ; return to caller done: not si not di add si, 1 adc di, 0 mov ax, si mov dx, di ; return popf pop si mov sp, bp pop bp ret 08h io endp IO_TEXT ends end </pre>	<pre> sum macro lodsw add stosw lodsw adc stosw and mov bx, ax opI cx, cosI[bx] opQ dx, cosQ[bx] endm rom segment assume org 8000h init: cli cld mov ax, 0 ds, ax es, ax ss, ax mov cx, 9*2 mov di, offset phases rep stosw mov sp, 8000h al, 10011011b out switch+3, al mov dx, 0 mov cx, m0dB test al, 10000000b jz l1 mov cx, m20dB jmp outpt l1: in al, switch loop: jz and xchg bx, ax mov si, ptrs[bx] mov di, offset phases sum mov, cos3, mov, cos1 sum add, cos2, add, cos2 sum add, cos1, add, cos3 sum < / >, , add, cos4 sum sub, cos1, add, cos3 sum sub, cos2, add, cos2 sum sub, cos3, add, cos1 sum sub, cos4, < / >, </pre>	<pre> ; get LS word of phase increment ; add it to the LS word of phase ; store the LS word of phase ; repeat for MS word ; (plus carry) ; si and di now point to next ones ; MS word of phase mod Ncos *2 ; is now cosine table offset in bx ; add/subtract cosine value to I sum ; and to Q sum cs:rom, ds:rom, es:rom, ss:rom ; EPROM starts at 32k ; interrupts off ; set direction flag = up ; set segment registers = 0 ; clear phase counters ; set stack (not used) ; set all 8255 ports as unattached ; input and start in off mode ... ; set Q for minimum o/p ; and I for for 0dB ; -20 dB level switch on ? ; if not, skip ahead ; else set I for -20 dB ; and set D/As ; run switch on ? ; if not, go set a fixed level ; middle 6 bits of switch is pointer ; to pointer to phase increments ; si --> first of 9 phase increments ; di --> first of 9 phase variables ; increment phases and sum ; cosine table values for ; I (in cx) and Q (in dx) </pre>
<pre> ; Fading simulator controller. Ed Casas and Ron Jeffery. July 24, 1987. idac equ 00H qdac equ 20H xfer equ 40H switch equ 60H Ncos equ 2048 ioffset equ 2066 shl 4 qoffset equ 2046 shl 4 m0dB equ 7215 m20dB equ m0dB/10 mem segment at 0 org 0400h phases dd 9 dup (?) org 8000h start label far mem ends </pre>	<pre> ; address of I D/A ; address of Q D/A ; address of common D/A output strobe ; address of switch port (8255 port A) ; number of entries in cosine tables ; measured I and Q DAC values for minimum ; RF output. 12 bits left justified ; DAC output for 0 dB (MS 12 bits used) ; DAC output for -20 dB ; absolute memory references. (DEBUG ; creates file for EPROM programmer). ; phase counters ; code start address (init:) </pre>	<pre> ; Fading simulator controller. Ed Casas and Ron Jeffery. July 24, 1987. idac equ 00H qdac equ 20H xfer equ 40H switch equ 60H Ncos equ 2048 ioffset equ 2066 shl 4 qoffset equ 2046 shl 4 m0dB equ 7215 m20dB equ m0dB/10 mem segment at 0 org 0400h phases dd 9 dup (?) org 8000h start label far mem ends </pre>
sim.asm		

<pre> sum add, cos5, add, cos5 ; / outpt: add cx, ioffset ; convert I sum to offset-binary mov al, ch ; move to ax and swap bytes to mov ah, cl ; output MS byte first out idac+1, ax ; idac+0 is also at idac+2 add dx, goffset ; \ mov al, dh ; mov ah, dl ; repeat for Q sum out qdac+1, ax ; / out xfer, al ; change both outputs at same time jmp loop ; repeat forever pntrs label word ; pointers into phase increments x = offset incrmnts ; table. 4 byte * 9 increments rept 128 ; per frequency = 36 bytes/entry dw x x = x+36 endm include tables ; cosl to cos5 and incrmnts org 0fff0h ; reset vector jmp start ends rom end </pre>	<p style="text-align: center;">ptabgen.for</p> <p>C Print phase increment tables in 8088 assembler format.</p> <pre> integer*4 i, j, M, tmp(9), No, N real fs, tpi, cos, float data No/8/, N/2048/, tpi/6.28318/, fs/2958./ M=4*No+2 write*,'(, , incrmnts label word, ,)' do 2 i=0,254,2 do 1 j=1,8 tmp(j)=cos(tpi*float(j)/M)*i/fs*N*65536*2 continue tmp(9)=i/fs*N*65536*2 write*,'(, , dd ',4(i11,',',',),i11,)' (tmp(j), j=1,5) write*,'(, , dd ',3(i11,',',',),i11,)' (tmp(j), j=6,9) continue end </pre>
<p style="text-align: center;">ctabgen.for</p> <p>C Print scaled cosine tables in 8088 assembler format.</p> <pre> integer i, j, k, N, No, tmp(8), tmsusd(5) real A, pwr, pi, sqrt, sin, cos, float data N/2048/, No/8/, pi/3.14159/, pwr/0./, tmsusd/2,2,2,1,1,1/ do 3 i=1,5 if(i.eq.5)then A=1750*sqrt(2.)*sin(pi/4.) else A=1750*2.*sin(pi*float(i)/No) endif pwr=pwr+tmsusd(i)*(A**2)/2. write*,'(, , cos'',i1, , label word, ,)'i do 2 j=1,N,8 do 1 k=1,8 tmp(k)=A*cos(float(j+k-2)*2.*pi/N) continue write*,'(, , dw ',7(i5,',',',),i5,)'(tmp(k),k=1,8) continue continue write*,'(, , ; dB at ',f10.1,')sqrt(2.*pwr) end </pre>	<p style="text-align: center;">ceval.c</p> <pre> /* ceval.c - evaluate BCH code performance from a run-length file */ #include <stdio.h> #include <assert.h> #include <math.h> #define BUFFSIZE 4096 main(int argc, char **argv) { int i, /* arg counter */ buf[BUFFSIZE], /* bit buffer */ *p, /* pointer to start of an FEC block in buf */ n=0, /* number of errors in a block */ argn = 0, /* symbols per block */ m, argm = 0, /* bits per symbol */ t, argt = 0, /* correctable symbols per block */ arge = 0, /* display errors in each block */ argi = 0, /* do interleaving */ argh = 0, /* produce histogram of errors/block */ argc = 0, /* display cumulative pdf */ argN = 0, /* normalize pdf */ </pre>

```

args = 0, /* BER, BKER summary */
nm, /* n*m */
bits; /* bits left to test in buf */

long
sume, sumb, /* sum of errors and bits tested */
h_sum, /* sum of histogram values */
*phist, /* pointer to hist */
hist [BUFFSIZE+1]; /* histogram */

FILE
*infile=NULL; /* input file */

for (i=1; i<argc; i++) {
    if ( !strcmp(argv[i], "-m") ) sscanf(argv[++i], "%d", &argm);
    if ( !strcmp(argv[i], "-n") ) sscanf(argv[++i], "%d", &argn);
    if ( !strcmp(argv[i], "-t") ) sscanf(argv[++i], "%d", &argt);
    if ( !strcmp(argv[i], "-h") ) argh=1;
    if ( !strcmp(argv[i], "-e") ) arge=1;
    if ( !strcmp(argv[i], "-i") ) argi=1;
    if ( !strcmp(argv[i], "-C") ) argC=1;
    if ( !strcmp(argv[i], "-N") ) argN=1;
    if ( !strcmp(argv[i], "-s") ) args=1;
    if ( !strcmp(argv[i], "-f") )
        if ( (infile=fopen(argv[++i], "r")) == NULL )
            perror (argv[i]);
}

/* initialize histogram and bit/error counters */

for ( i=0; i<BUFFSIZE+1; i++) hist[i] = 0;
sume = sumb = 0;

/* ensure an input file */

if ( infile == NULL ) infile=stdin;

/* do for all FEC blocks in one OFDM block :
interleave if necessary
if fewer than t symbol errors, bit errors = 0
else count bit errors
update histogram and bit/error counts
maybe print number of bit errors */

while ( (bits-get_buf(infile, buf)) > 0 ) {
    if ( n <= 0 ) { /* initialize n,m,t,nm */
        if ( argn ) n=argn; else n=bits;
        if ( argm ) m=argm; else m=1;
        if ( argt ) t=argt; else t=0;
    }
    nm=n*m;
    printf("%% BCH (n=%d, m=%d, t=%d)\n", n, m, t);
    if ( n <= 0 || m <= 0 || t < 0 ) err("bad parameter");
    if ( arg_e ) printf("%% errors per block :\n");
};

p=buf;
while ( bits >= nm ) {
    if ( arg_i ) intlv (p, nm, 1);
    if ( !nerr(p,m,n) > t ) e=nerr(p,nm); else e=0;
    (hist[e])++;
    sume += e;
    sumb += nm;
    if ( arg_e ) printf("%d\n", e);
    bits-=nm;
    p+=nm;
};

/* compute sum and maybe make cumulative */

h_sum = 0;
for ( i=0; i<nm; i++ ) {
    h_sum+=hist[i];
    if ( argC ) hist[i] = h_sum;
};

/* error checks */

if ( sumb <= 0 ) err("no input");
assert( h_sum*n*m == sumb );
assert( h_sum > 0 );
assert( sumb > 0 );

/* maybe display [C] pdf, maybe normalized */

if ( arg_h ) {
    printf("%% [C]PDF : \n");
    for ( i=0; i<= n; i++ )
        if ( argN ) printf ("%d %g\n",
            i, ((float) (hist[i])) / h_sum );
        else printf("%d %ld\n", i, hist[i]);
};

/* maybe display summary */

if ( args ) {
    printf("%% BER = %g\n",
        (float) (sume) / (float) (sumb) );
    printf("%% BKER = %g\n",

```

```

    } ;
    (float) (h_sum - hist[0]) / (float) h_sum ) ;
} ;
err (char *msg) {
    fputs(msg,stderr) ;
    fputs("\n",stderr) ;
    exit(1) ;
} ;

int nserr(int *p, int m, int n) {
/* count number of m-bit symbols with errors in a block of n symbols */
    int e=0 ;
    while ( n-- ) {
        if ( nerr(p,m) ) e++ ;
        p+=m ;
    } ;
    return ( e ) ;
} ;

int nerr(int *p, int n) {
/* count number of bits in error in a block of n bits */
    int e=0 ;
    while ( n-- ) if ( *p++ ) e++ ;
    return ( e ) ;
} ;

int get_buf(FILE *file, int *buf) {
/* Reads error-free run lengths from a file and unpacks the run
lengths into a bit-error pattern (0=no error, 1=error). Last
(error-free) run length in the block should be followed by a -1.
Returns the number of bits generated. */
    int i=0, k ;
    while ( fscanf(file,"%d",&k) == 1 && k >= 0 ) {
        if ( i+k+1 >= BUFFSIZE ) k = BUFFSIZE-i-2 ;
        while ( k-- ) { buf[i++] = 0 ; } ;
        buf[i++] = 1 ;
    } ;
    return ( i ? i-1 : 0 ) ;
} ;

int put_buf(FILE *file, int *p, int n) {
#define put_cnt(x) fprintf(file,"%d\n",x)
/* Packs a bit-error pattern (0=no error, 1=error) into
corresponding error-free run lengths and writes it to a file.
The last (error-free) run length in the block is followed by a
-1. */
    int k ;
    k=0;
    while ( n-- )
        if ( *p++ ) { put_cnt(k) ; k=0 ; }
        else k++ ;
    put_cnt(k) ;
    put_cnt(-1) ;
} ;

int intlv(int *in, int n, int dir) { /* block interleaver */
    int
        out [ BUFFSIZE ],
        i, j, k, l ;
/* compute interleaving step size (round up to make sure
interleave all) */
    k = (int) floor ( 1.0 + sqrt ( (float) n ) ) ;
    l=0 ;
    for ( i=0 ; i<k ; i++ ) { /* step through offsets within blocks */
        for ( j=i ; j<n ; j+=k ) { /* step through blocks */
            if ( dir ) out [l] = in [j] ;
            else out [j] = in [l] ;
            l++ ;
        } ;
    } ;
    for ( i=0 ; i<n ; i++ ) in[i] = out[i] ;
} ;

```

bkp.c

```

/* bkp.c

computes average distribution of the number of bit errors in blocks
of size N assuming independent errors within each block but different
BERS for each block. The standard input contains the block BERS
and the program takes one argument, the block size. At input EOF
the distribution is written to standard output. The binomial
distribution is computed using the 'bico' routine from _Numerical
Recipes_.

Ed.Casas 89-3-7 */

double lnbico(), factln(), gammln();
double ipow();
void nrerror ();

#include <stdio.h>
#include <math.h>

#define NMAX 4096

main(int argc, char **argv) {
    int i, n, nblk;
    double ber, nber;
    double prob [ NMAX+1 ];
    double sump;

    /* check arguments */
    if ( argc < 2 ) {
        fprintf(stderr, "Usage %s <bits/block>\n", argv[0]);
        exit(1);
    };
    if ( sscanf(argv[1], "%d", &n) != 1 || n > NMAX ) {
        fprintf(stderr, "N (%s) bad or too large.\n", argv[1]);
        exit(1);
    };
}

/* test for log factorial
{ int x1, x2;
  printf ("enter x1 and x2 ");
  scanf ("%d %d", &x1, &x2);
  printf ("ln of factorial: = %f %f\n", factln(x1), factln(x2));
  printf ("x1!/x2! ?= %lg\n",

```

```

        exp( (double) (factln(x1)-factln(x2))) );
    };
}

/*
/* initialize distribution */
for ( i=0; i<n; i++) prob[i]=0.0;
nblk = 0;

/* loop through input BERS, compute, and sum distributions */
while ( scanf("%lg", &ber) == 1 ) {
    for ( i=0; i<n; i++) prob[i] += exp ( lnbico(n,i)
        + log ( 1.0-ber ) * ( n-i )
        + log ( ber ) * i );
    nblk ++;
};

/* display cumulative results */
sump=0.;
if ( nblk <= 0 ) {
    fprintf(stderr, "No input.\n");
    exit(1);
}
else for ( i=0; i<n; i++)
    printf("%d %lg\n", i, sump += (prob[i]/nblk) );
};

/* the following routines are adapted from _Numerical_Recipes_in_C_ */
double lnbico(n,k) /* modified to return ln */
int n,k;
{
    double factln();
    return factln(n)-factln(k)-factln(n-k);
}

#define MAXN 2048
double factln(n)
int n;
{
    static double a[MAXN+1]; /* cache blocksizes up to MAXN bits */
    double gammln();
    void nrerror();
    if ( n < 0 ) nrerror("Negative factorial in routine FACTLN");
    if ( n <= 1 ) return 0.0;

```

<pre> if (n <= MAXN) return a[n] ? a[n] : (a[n]=gammln(n+1.0)); else return gammln(n+1.0); } double gammln(xx) double xx; { double x,tmp,ser; static double cof[6]={76.18009173,-86.50532033,24.01409822, -1.231739516,0.120858003e-2,-0.536382e-5}; int j; x=xx-1.0; tmp=x+5.5; tmp -= (x+0.5)*log(tmp); ser=1.0; for (j=0;j<=5;j++) { x += 1.0; ser += cof[j]/x; } return -tmp+log(2.50662827465*ser); } void nerror(error_text) char error_text[]; { fprintf(stderr,"Numerical Recipes run-time error...\n"); fprintf(stderr,"%s\n",error_text); fprintf(stderr,"...now exiting to system...\n"); exit(1); } </pre>	<p style="text-align: center;">out2bers.awk</p> <pre> # bers.awk - extracts the block BERs for one set of N, SNR values # ed.casas 89-3-17 # /\ %B/ { if (\$2 == n && \$3 == snr) print \$4 } </pre> <p style="text-align: center;">out2runs.awk</p> <pre> # runs.awk - extracts the error-free run lengths for one set of # N, SNR values # ed.casas 89-3-2 # /\ %N/ { if (\$2 == n && \$3 == snr) on=1 ; else on=0 } # un-comment next line for testing # /\ %N/ { if (on) print \$0 } # /\ %R/ { if (on) print \$2 } </pre>
<p style="text-align: center;">out2bers.csh</p> <pre> #!/bin/csh # # extracts block BERs from fdint output (out.bers) # Ed.Casas 89-3-17 # foreach s (10 15 20 25) foreach n (256 1024 4096) awk -f out2bers.awk n=\$n snr=\$s. out.bers >bers.\$n.\$s end end </pre>	